# Programming on Parallel Machines <br> Norm Matloff <br> University of California, Davis 

GPU, Multicore, Clusters and More

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## Author's Biographical Sketch

Dr. Norm Matloff is a professor of computer science at the University of California at Davis, and was formerly a professor of statistics at that university. He is a former database software developer in Silicon Valley, and has been a statistical consultant for firms such as the Kaiser Permanente Health Plan.

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Dr. Matloff is the author of two published textbooks, and of a number of widely-used Web tutorials on computer topics, such as the Linux operating system and the Python programming language. He and Dr. Peter Salzman are authors of The Art of Debugging with GDB, DDD, and Eclipse. Prof. Matloff's book on the R programming language, The Art of $R$ Programming, was published in 2011. His book, Parallel Computation for Data Science, will come out in 2014. He is also the author of several open-source textbooks, including From Algorithms to Z-Scores: Probabilistic and Statistical Modeling in Computer Science (http://heather.cs.ucdavis.edu/probstatbook), and Programming on Parallel Machines (http://heather.cs.ucdavis.edu/~matloff/ParProcBook. pdf).


#### Abstract

About This Book

Why is this book different from all other parallel programming books? It is aimed more on the


 practical end of things, in that:- There is very little theoretical content, such as O() analysis, maximum theoretical speedup, PRAMs, directed acyclic graphs (DAGs) and so on.
- Real code is featured throughout.
- We use the main parallel platforms-OpenMP, CUDA and MPI—rather than languages that at this stage are largely experimental or arcane.
- The running performance themes-communications latency, memory/network contention, load balancing and so on-are interleaved throughout the book, discussed in the context of specific platforms or applications.
- Considerable attention is paid to techniques for debugging.

The main programming language used is $\mathrm{C} / \mathrm{C}++$, but some of the code is in R , which has become the pre-eminent language for data analysis. As a scripting language, $R$ can be used for rapid prototyping. In our case here, it enables me to write examples which are much less less cluttered than they would be in $\mathrm{C} / \mathrm{C}++$, thus easier for students to discern the fundamental parallelixation principles involved. For the same reason, it makes it easier for students to write their own parallel code, focusing on those principles. And R has a rich set of parallel libraries.

It is assumed that the student is reasonably adept in programming, and has math background through linear algebra. An appendix reviews the parts of the latter needed for this book. Another appendix presents an overview of various systems issues that arise, such as process scheduling and virtual memory.

It should be note that most of the code examples in the book are NOT optimized. The primary emphasis is on simplicity and clarity of the techniques and languages used. However, there is plenty of discussion on factors that affect speed, such cache coherency issues, network delays, GPU memory structures and so on.

Here's how to get the code files you'll see in this book: The book is set in LaTeX, and the raw .tex files are available in http://heather.cs.ucdavis.edu/~matloff/158/PLN. Simply download the relevant file (the file names should be clear), then use a text editor to trim to the program code of interest.

In order to illustrate for students the fact that research and teaching (should) enhance each other, I occasionally will make brief references here to some of my research work.

Like all my open source textbooks, this one is constantly evolving. I continue to add new topics, new examples and so on, and of course fix bugs and improve the exposition. For that reason, it is better to link to the latest version, which will always be at http://heather.cs.ucdavis.edu/ $\sim_{\text {matloff/ }}$ /158/PLN/ParProcBook.pdf, rather than to copy it.

For that reason, feedback is highly appreciated. I wish to thank Stuart Ambler, Stuart Hansen, Bill Hsu, Sameer Khan, Mikel McDaniel, Richard Minner, Lars Seeman and Johan Wikström for their comments. I'm also very grateful to Professor Hsu for his making available to me advanced GPU-equipped machines.

You may also be interested in my open source textbook on probability and statistics, at http: //heather.cs.ucdavis.edu/probstatbook.

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## Chapter 1

## Introduction to Parallel Processing

Parallel machines provide a wonderful opportunity for applications with large computational requirements. Effective use of these machines, though, requires a keen understanding of how they work. This chapter provides an overview of both the software and hardware.

### 1.1 Why Use Parallel Systems?

### 1.1.1 Execution Speed

There is an ever-increasing appetite among some types of computer users for faster and faster machines. This was epitomized in a statement by the late Steve Jobs, founder/CEO of Apple and Pixar. He noted that when he was at Apple in the 1980s, he was always worried that some other company would come out with a faster machine than his. But later at Pixar, whose graphics work requires extremely fast computers, he was always hoping someone would produce faster machines, so that he could use them!

A major source of speedup is the parallelizing of operations. Parallel operations can be either within-processor, such as with pipelining or having several ALUs within a processor, or betweenprocessor, in which many processor work on different parts of a problem in parallel. Our focus here is on between-processor operations.

For example, the Registrar's Office at UC Davis uses shared-memory multiprocessors for processing its on-line registration work. Online registration involves an enormous amount of database computation. In order to handle this computation reasonably quickly, the program partitions the work to be done, assigning different portions of the database to different processors. The database field has contributed greatly to the commercial success of large shared-memory machines.

As the Pixar example shows, highly computation-intensive applications like computer graphics also have a need for these fast parallel computers. No one wants to wait hours just to generate a single image, and the use of parallel processing machines can speed things up considerably. For example, consider ray tracing operations. Here our code follows the path of a ray of light in a scene, accounting for reflection and absorbtion of the light by various objects. Suppose the image is to consist of 1,000 rows of pixels, with 1,000 pixels per row. In order to attack this problem in a parallel processing manner with, say, 25 processors, we could divide the image into 25 squares of size 200x200, and have each processor do the computations for its square.

Note, though, that it may be much more challenging than this implies. First of all, the computation will need some communication between the processors, which hinders performance if it is not done carefully. Second, if one really wants good speedup, one may need to take into account the fact that some squares require more computation work than others. More on this below.

We are now in the era of Big Data, which requires Big Computation, thus again generating a major need for parallel processing.

### 1.1.2 Memory

Yes, execution speed is the reason that comes to most people's minds when the subject of parallel processing comes up. But in many applications, an equally important consideration is memory capacity. Parallel processing application often tend to use huge amounts of memory, and in many cases the amount of memory needed is more than can fit on one machine. If we have many machines working together, especially in the message-passing settings described below, we can accommodate the large memory needs.

### 1.1.3 Distributed Processing

In the above two subsections we've hit the two famous issues in computer science - time (speed) and space (memory capacity). But there is a third reason to do parallel processing, which actually has its own name, distributed processing. In a distributed database, for instance, parts of the database may be physically located in widely dispersed sites. If most transactions at a particular site arise locally, then we would make more efficient use of the network, and so on.

### 1.1.4 Our Focus Here

In this book, the primary emphasis is on processing speed.

### 1.2 Parallel Processing Hardware

This is a common scenario: Someone acquires a fancy new parallel machine, and excitedly writes a program to run on it - only to find that the parallel code is actually slower than the original serial version! This is due to lack of understanding of how the hardware works, at least at a high level.

This is not a hardware book, but since the goal of using parallel hardware is speed, the efficiency of our code is a major issue. That in turn means that we need a good understanding of the underlying hardware that we are programming. In this section, we give an overview of parallel hardware.

### 1.2.1 Shared-Memory Systems

### 1.2.1.1 Basic Architecture

Here many CPUs share the same physical memory. This kind of architecture is sometimes called MIMD, standing for Multiple Instruction (different CPUs are working independently, and thus typically are executing different instructions at any given instant), Multiple Data (different CPUs are generally accessing different memory locations at any given time).

Until recently, shared-memory systems cost hundreds of thousands of dollars and were affordable only by large companies, such as in the insurance and banking industries. The high-end machines are indeed still quite expensive, but now multicore machines, in which two or more CPUs share a common memory ${ }^{1}$ are commonplace in the home and even in cell phones!

### 1.2.1.2 Multiprocessor Topologies

A Symmetric Multiprocessor (SMP) system has the following structure:


The multicore setup is effectively the same as SMP, except that the processors are all on one chip, attached to the bus.

So-called NUMA architectures will be discussed in Chapter 3.

[^0]
### 1.2.1.3 Memory Issues Etc.

Consider the SMP figure above.

- The Ps are processors, e.g. off-the-shelf chips such as Pentiums.
- The Ms are memory modules. These are physically separate objects, e.g. separate boards of memory chips. It is typical that there will be the same number of memory modules as processors. In the shared-memory case, the memory modules collectively form the entire shared address space, but with the addresses being assigned to the memory modules in one of two ways:
- (a)

High-order interleaving. Here consecutive addresses are in the same M (except at boundaries). For example, suppose for simplicity that our memory consists of addresses 0 through 1023, and that there are four Ms. Then M0 would contain addresses 0-255, M1 would have 256-511, M2 would have 512-767, and M3 would have 768-1023.
We need 10 bits for addresses (since $1024=2^{10}$ ). The two most-significant bits would be used to select the module number (since $4=2^{2}$ ); hence the term high-order in the name of this design. The remaining eight bits are used to select the word within a module.

- (b)

Low-order interleaving. Here consecutive addresses are in consecutive memory modules (except when we get to the right end). In the example above, if we used low-order interleaving, then address 0 would be in M0, 1 would be in M1, 2 would be in M2, 3 would be in M3, 4 would be back in M0, 5 in M1, and so on.
Here the two least-significant bits are used to determine the module number.

- To make sure only one processor uses the bus at a time, standard bus arbitration signals and/or arbitration devices are used.
- There may also be coherent caches, which we will discuss later.

All of the above issues can have major on the speed of our program, as will be seen later.

### 1.2.2 Message-Passing Systems

### 1.2.2.1 Basic Architecture

Here we have a number of independent CPUs, each with its own independent memory. The various processors communicate with each other via networks of some kind.

### 1.2.2.2 Example: Clusters

Here one has a set of commodity PCs and networks them for use as a parallel processing system. The PCs are of course individual machines, capable of the usual uniprocessor (or now multiprocessor) applications, but by networking them together and using parallel-processing software environments, we can form very powerful parallel systems.

One factor which can be key to the success of a cluster is the use of a fast network, fast both in terms of hardware and network protocol. Ordinary Ethernet and TCP/IP are fine for the applications envisioned by the original designers of the Internet, e.g. e-mail and file transfer, but is slow in the cluster context. A good network for a cluster is, for instance, Infiniband.

Clusters have become so popular that there are now "recipes" on how to build them for the specific purpose of parallel processing. The term Beowulf come to mean a cluster of PCs, usually with a fast network connecting them, used for parallel processing. Software packages such as ROCKS (http://www.rocksclusters.org/wordpress/) have been developed to make it easy to set up and administer such systems.

### 1.2.3 SIMD

In contrast to MIMD systems, processors in SIMD - Single Instruction, Multiple Data - systems execute in lockstep. At any given time, all processors are executing the same machine instruction on different data.

Some famous SIMD systems in computer history include the ILLIAC and Thinking Machines Corporation's CM-1 and CM-2. Also, DSP ("digital signal processing") chips tend to have an SIMD architecture.

But today the most prominent example of SIMD is that of GPUs-graphics processing units. In addition to powering your PC's video cards, GPUs can now be used for general-purpose computation. The architecture is fundamentally shared-memory, but the individual processors do execute in lockstep, SIMD-fashion.

### 1.3 Programmer World Views

### 1.3.1 Example: Matrix-Vector Multiply

To explain the paradigms, we will use the term nodes, where roughly speaking one node corresponds to one processor, and use the following example:

Suppose we wish to multiply an $n x 1$ vector X by an nxn matrix A, putting the product in an $n x 1$ vector Y , and we have p processors to share the work.

In all the forms of parallelism, each node could be assigned some of the rows of A, and that node would multiply X by those rows, thus forming part of Y .

Note that in typical applications, the matrix A would be very large, say thousands of rows, possibly even millions. Otherwise the computation could be done quite satisfactorily in a serial, i.e. nonparallel manner, making parallel processing unnecessary..

### 1.3.2 Shared-Memory

### 1.3.2.1 Programmer View

In implementing the matrix-vector multiply example of Section 1.3.1 in the shared-memory paradigm, the arrays for $\mathrm{A}, \mathrm{X}$ and Y would be held in common by all nodes. If for instance node 2 were to execute

```
y[3] = 12;
```

and then node 15 were to subsequently execute

```
print("%d\n",Y[3]);
```

then the outputted value from the latter would be 12 .
Computation of the matrix-vector product AX would then involve the nodes somehow deciding which nodes will handle which rows of A. Each node would then multiply its assigned rows of A times X , and place the result directly in the proper section of the shared Y .

Today, programming on shared-memory multiprocessors is typically done via threading. (Or, as we will see in other chapters, by higher-level code that runs threads underneath.) A thread is similar to a process in an operating system (OS), but with much less overhead. Threaded applications have become quite popular in even uniprocessor systems, and Unix ${ }^{2}$ Windows, Python, Java, Perl and now C ++11 and R (via my Rdsm package) all support threaded programming.

In the typical implementation, a thread is a special case of an OS process. But the key difference is that the various threads of a program share memory. (One can arrange for processes to share memory too in some OSs, but they don't do so by default.)

[^1]On a uniprocessor system, the threads of a program take turns executing, so that there is only an illusion of parallelism. But on a multiprocessor system, one can genuinely have threads running in parallel $]^{3}$ Whenever a processor becomes available, the OS will assign some ready thread to it. So, among other things, this says that a thread might actually run on different processors during different turns.

Important note: Effective use of threads requires a basic understanding of how processes take turns executing. See Section A. 1 in the appendix of this book for this material.

One of the most popular threads systems is Pthreads, whose name is short for POSIX threads. POSIX is a Unix standard, and the Pthreads system was designed to standardize threads programming on Unix. It has since been ported to other platforms.

### 1.3.2.2 Example: Pthreads Prime Numbers Finder

Following is an example of Pthreads programming, in which we determine the number of prime numbers in a certain range. Read the comments at the top of the file for details; the threads operations will be explained presently.

```
// PrimesThreads.c
// threads-based program to find the number of primes between 2 and n;
// uses the Sieve of Eratosthenes, deleting all multiples of 2, all
// multiples of 3, all multiples of 5, etc.
// for illustration purposes only; NOT claimed to be efficient
// Unix compilation: gcc -g -o primesthreads PrimesThreads.c -lpthread -lm
// usage: primesthreads n num_threads
#include <stdio.h>
#include <math.h>
#include <pthread.h> // required for threads usage
#define MAX_N 100000000
#define MAX_THREADS 25
// shared variables
int nthreads, // number of threads (not counting main())
    n, // range to check for primeness
    prime[MAX_N+1], // in the end, prime[i] = 1 if i prime, else 0
    nextbase; // next sieve multiplier to be used
// lock for the shared variable nextbase
pthread_mutex_t nextbaselock = PTHREAD_MUTEX_INITIALIZER;
// ID structs for the threads
pthread_t id[MAX_THREADS];
```

[^2]```
// "crosses out" all odd multiples of k
void crossout(int k)
{ int i;
    for (i = 3; i*k <= n; i += 2) {
        prime[i*k] = 0;
    }
}
// each thread runs this routine
void *worker(int tn) // tn is the thread number (0,1,\ldots..)
{ int lim,base,
            work = 0; // amount of work done by this thread
    // no need to check multipliers bigger than sqrt(n)
    lim = sqrt(n);
    do {
        // get next sieve multiplier, avoiding duplication across threads
        // lock the lock
        pthread_mutex_lock(&nextbaselock);
        base = nextbase;
        nextbase += 2;
        // unlock
        pthread_mutex_unlock(&nextbaselock);
        if (base <= lim) {
            // don't bother crossing out if base known composite
            if (prime[base]) {
                crossout(base);
                work++; // log work done by this thread
            }
        }
        else return work;
    } while (1);
}
main(int argc, char **argv)
{ int nprimes, // number of primes found
        i,work;
    n = atoi(argv[1]);
    nthreads = atoi(argv[2]);
    // mark all even numbers nonprime, and the rest "prime until
    // shown otherwise"
    for (i = 3; i <= n; i++) {
        if (i%2 == 0) prime[i] = 0;
        else prime[i] = 1;
    }
    nextbase = 3;
    // get threads started
    for (i = 0; i < nthreads; i++) {
        // this call says create a thread, record its ID in the array
        // id, and get the thread started executing the function worker(),
        // passing the argument i to that function
        pthread_create(&id[i],NULL,worker,i);
    }
    // wait for all done
    for (i = 0; i < nthreads; i++) {
        // this call says wait until thread number id[i] finishes
        // execution, and to assign the return value of that thread to our
```

```
// local variable work here
    pthread_join(id[i],&work);
    printf("%d values of base done\n",work);
    }
    // report results
    nprimes = 1;
    for (i = 3; i <= n; i++)
        if (prime[i]) {
        nprimes++;
    }
    printf("the number of primes found was % \\n",nprimes);
}
```

To make our discussion concrete, suppose we are running this program with two threads. Suppose also the both threads are running simultaneously most of the time. This will occur if they aren't competing for turns with other threads, say if there are no other threads, or more generally if the number of other threads is less than or equal to the number of processors minus two. (Actually, the original thread is main(), but it lies dormant most of the time, as you'll see.)

Note the global variables:

```
int nthreads, // number of threads (not counting main())
    n, // range to check for primeness
    prime[MAX_N+1], // in the end, prime[i] = 1 if i prime, else 0
    nextbase; // next sieve multiplier to be used
pthread_mutex_t nextbaselock = PTHREAD_MUTEX_INITIALIZER;
pthread_t id[MAX_THREADS];
```

This will require some adjustment for those who've been taught that global variables are "evil."
In most threaded programs, all communication between threads is done via global variables ${ }^{T}$ So even if you consider globals to be evil, they are a necessary evil in threads programming.

Personally I have always thought the stern admonitions against global variables are overblown anyway; see http://heather.cs.ucdavis.edu/~matloff/globals.html. But as mentioned, those admonitions are routinely ignored in threaded programming. For a nice discussion on this, see the paper by a famous MIT computer scientist on an Intel Web page, at http://software.intel. com/en-us/articles/global-variable-reconsidered/?wapkw=\%28parallelism\%29.

As mentioned earlier, the globals are shared by all processors $5^{5}$ If one processor, for instance, assigns the value 0 to prime[35] in the function crossout(), then that variable will have the value

[^3]0 when accessed by any of the other processors as well. On the other hand, local variables have different values at each processor; for instance, the variable $\mathbf{i}$ in that function has a different value at each processor.

Note that in the statement
pthread_mutex_t nextbaselock = PTHREAD_MUTEX_INITIALIZER;
the right-hand side is not a constant. It is a macro call, and is thus something which is executed.
In the code

```
pthread_mutex_lock(&nextbaselock);
base = nextbase
nextbase += 2
pthread_mutex_unlock(&nextbaselock);
```

we see a critical section operation which is typical in shared-memory programming. In this context here, it means that we cannot allow more than one thread to execute the code

```
base = nextbase;
```

nextbase $+=2$;
at the same time. A common term used for this is that we wish the actions in the critical section to collectively be atomic, meaning not divisible among threads. The calls to pthread_mutex_lock() and pthread_mutex_unlock() ensure this. If thread A is currently executing inside the critical section and thread B tries to lock the lock by calling pthread_mutex_lock(), the call will block until thread B executes pthread_mutex_unlock().

Here is why this is so important: Say currently nextbase has the value 11. What we want to happen is that the next thread to read nextbase will "cross out" all multiples of 11 . But if we allow two threads to execute the critical section at the same time, the following may occur, in order:

- thread A reads nextbase, setting its value of base to 11
- thread B reads nextbase, setting its value of base to 11
- thread A adds 2 to nextbase, so that nextbase becomes 13
- thread B adds 2 to nextbase, so that nextbase becomes 15

Two problems would then occur:

- Both threads would do "crossing out" of multiples of 11, duplicating work and thus slowing down execution speed.
- We will never "cross out" multiples of 13 .

Thus the lock is crucial to the correct (and speedy) execution of the program.
Note that these problems could occur either on a uniprocessor or multiprocessor system. In the uniprocessor case, thread A's turn might end right after it reads nextbase, followed by a turn by $B$ which executes that same instruction. In the multiprocessor case, A and B could literally be running simultaneously, but still with the action by B coming an instant after A.

This problem frequently arises in parallel database systems. For instance, consider an airline reservation system. If a flight has only one seat left, we want to avoid giving it to two different customers who might be talking to two agents at the same time. The lines of code in which the seat is finally assigned (the commit phase, in database terminology) is then a critical section.

A critical section is always a potential bottleneck in a parallel program, because its code is serial instead of parallel. In our program here, we may get better performance by having each thread work on, say, five values of nextbase at a time. Our line
nextbase $+=2$;
would become
nextbase += 10 ;

That would mean that any given thread would need to go through the critical section only one-fifth as often, thus greatly reducing overhead. On the other hand, near the end of the run, this may result in some threads being idle while other threads still have a lot of work to do.

Note this code.

```
for (i = 0; i < nthreads; i++) {
    pthread_join(id[i],&work);
    printf("%d values of base done\n",work);
}
```

This is a special case of of barrier.
A barrier is a point in the code that all threads must reach before continuing. In this case, a barrier is needed in order to prevent premature execution of the later code

```
for (i = 3; i <= n; i++)
    if (prime[i]) {
        nprimes++;
    }
```

which would result in possibly wrong output if we start counting primes before some threads are done.

Actually, we could have used Pthreads' built-in barrier function. We need to declare a barrier variable, e.g.

```
pthread_barrier_t barr;
```

and then call it like this:

```
pthread_barrier_wait(&barr);
```

The pthread_join() function actually causes the given thread to exit, so that we then "join" the thread that created it, i.e. main(). Thus some may argue that this is not really a true barrier.

Barriers are very common in shared-memory programming, and will be discussed in more detail in Chapter 3 .

### 1.3.2.3 Role of the OS

Let's again ponder the role of the OS here. What happens when a thread tries to lock a lock:

- The lock call will ultimately cause a system call, causing the OS to run.
- The OS keeps track of the locked/unlocked status of each lock, so it will check that status. th
- Say the lock is unlocked (a 0 ). Then the OS sets it to locked (a 1 ), and the lock call returns. The thread enters the critical section.
- When the thread is done, the unlock call unlocks the lock, similar to the locking actions.
- If the lock is locked at the time a thread makes a lock call, the call will block. The OS will mark this thread as waiting for the lock. When whatever thread currently using the critical section unlocks the lock, the OS will relock it and unblock the lock call of the waiting thread. If several threads are waiting, of course only one will be unblock.

Note that main() is a thread too, the original thread that spawns the others. However, it is dormant most of the time, due to its calls to pthread_join().

Finally, keep in mind that although the globals variables are shared, the locals are not. Recall that local variables are stored on a stack. Each thread (just like each process in general) has its own stack. When a thread begins a turn, the OS prepares for this by pointing the stack pointer register to this thread's stack.

### 1.3.2.4 Debugging Threads Programs

Most debugging tools include facilities for threads. Here's an overview of how it works in GDB. First, as you run a program under GDB, the creation of new threads will be announced, e.g.

```
(gdb) r 100 2
Starting program: /debug/primes 100 2
[New Thread 16384 (LWP 28653)]
[New Thread 32769 (LWP 28676)]
[New Thread 16386 (LWP 28677)]
[New Thread 32771 (LWP 28678)]
```

You can do backtrace (bt) etc. as usual. Here are some threads-related commands:

- info threads (gives information on all current threads)
- thread 3 (change to thread 3)
- break 88 thread 3 (stop execution when thread 3 reaches source line 88)
- break 88 thread 3 if $\mathrm{x}==\mathrm{y}$ (stop execution when thread 3 reaches source line 88 and the variables x and y are equal)

Of course, many GUI IDEs use GDB internally, and thus provide the above facilities with a GUI wrapper. Examples are DDD, Eclipse and NetBeans.

### 1.3.2.5 Higher-Level Threads

The OpenMP library gives the programmer a higher-level view of threading. The threads are there, but rather hidden by higher-level abstractions. We will study OpenMP in detail in Chapter 4 , and use it frequently in the succeeding chapters, but below is an introductory example.

### 1.3.2.6 Example: Sampling Bucket Sort

This code implements the sampling bucket sort of Section 12.5

```
// OpenMP introductory example: sampling bucket sort
// compile: gcc -fopenmp -o bsort bucketsort.c
// set the number of threads via the environment variable
// OMPNUM+THREADS, e.g. in the C shell
// setenv OMPNUMTHREADS 8
#include <omp.h> // required
#include <stdlib.h>
// needed for call to qsort()
int cmpints(int *u, int *v)
{ if (*u< *v) return - 1;
    if (*u > *v) return 1;
    return 0;
}
// adds xi to the part array, increments npart, the length of part
void grab(int xi, int *part, int *npart)
{
    part[*npart] = xi;
    *npart += 1;
}
// finds the min and max in y, length ny,
// placing them in miny and maxy
void findminmax(int *y, int ny, int *miny, int *maxy)
{ int i,yi;
    *miny = *maxy = y [0];
    for (i = 1; i < ny; i++) {
        yi}=y[i]
        if (yi< *miny) *miny = yi;
        else if (yi > *maxy) *maxy = yi;
    }
}
// sort the array x of length n
void bsort(int *x, int n)
{ // these are local to this function, but shared among the threads
    float *bdries; int *counts;
    #pragma omp parallel
    // entering this block activates the threads, each executing it
    { // these are local to each thread:
        int me = omp_get_thread_num();
        int nth = omp_get_num_threads();
        int i, xi,minx,maxx, start;
        int *mypart;
        float increm;
```

```
    int SAMPLESIZE;
    // now determine the bucket boundaries; nth - 1 of them, by
    // sampling the array to get an idea of its range
    #pragma omp single // only 1 thread does this, implied barrier at end
    {
        if (n > 1000) SAMPLESIZE = 1000;
        else SAMPLESIZE = n / 2;
        findminmax (x ,SAMPLESIZE,&minx,&maxx );
        bdries = malloc(( nth - 1)*sizeof(float ));
        increm = (maxx - minx) / (float) nth;
        for (i = 0; i < nth-1; i++)
            bdries[i] = minx + (i+1) * increm;
    // array to serve as the count of the numbers of elements of x
    // in each bucket
    counts}=malloc(nth*sizeof(int))
    }
    // now have this thread grab its portion of the array; thread 0
    // takes everything below bdries[0], thread 1 everything between
    // bdries[0] and bdries[1], etc., with thread nth-1 taking
    // everything over bdries[nth-1]
    mypart = malloc(n*sizeof(int)); int nummypart = 0;
    for (i = 0; i < n; i++) {
        if (me=0) {
            if (x[i] <= bdries[0]) grab(x[i],mypart,&nummypart);
        }
        else if (me<nth-1) {
            if (x[i] > bdries[me-1] && x[i] <= bdries[me])
                grab(x[i],mypart,&nummypart);
        } else
            if (x[i] > bdries[me-1]) grab(x[i],mypart,&nummypart );
        }
        // now record how many this thread got
        counts[me] = nummypart;
        // sort my part
        qsort(mypart, nummypart, sizeof(int), cmpints);
        #pragma omp barrier // other threads need to know all of counts
        // copy sorted chunk back to the original array; first find start point
        start = 0;
        for (i = 0; i < me; i++) start += counts[i];
        for (i = 0; i < nummypart; i++) {
        x[start+i] = mypart[i];
        }
    }
    // implied barrier here; main thread won't resume until all threads
    // are done
}
int main(int argc, char **argv)
{
    // test case
```

```
    int n = atoi(argv[1]), *x = malloc(n*sizeof(int));
    int i;
    for (i = 0; i < n; i++) x[i] = rand() % 50;
    if (n < 100)
    for (i = 0; i < n; i++) printf("%d\n",x[i]);
    bsort(x,n);
    if (n <= 100) {
    printf("x after sorting:\n");
    for (i = 0; i < n; i++) printf("%d\n",x[i]);
}
}
```

Details on OpenMP are presented in Chapter 4. Here is an overview of a few of the OpenMP constructs available:

## - \#pragma omp for

In our example above, we wrote our own code to assign specific threads to do specific parts of the work. An alternative is to write an ordinary for loop that iterates over all the work to be done, and then ask OpenMP to assign specific iterations to specific threads. To do this, insert the above pragma just before the loop.

## - \#pragma omp critical

The block that follows is implemented as a critical section. OpenMP sets up the locks etc. for you, alleviating you of work and alleviating your code of clutter.

### 1.3.3 Message Passing

### 1.3.3.1 Programmer View

Again consider the matrix-vector multiply example of Section 1.3.1. In contrast to the sharedmemory case, in the message-passing paradigm all nodes would have separate copies of A, X and Y. Our example in Section 1.3.2.1 would now change. in order for node 2 to send this new value of $\mathrm{Y}[3]$ to node 15 , it would have to execute some special function, which would be something like

```
send(15,12,"Y[3]");
```

and node 15 would have to execute some kind of receive() function.
To compute the matrix-vector product, then, would involve the following. One node, say node 0 , would distribute the rows of A to the various other nodes. Each node would receive a different set
of rows. The vector X would be sent to all nodes ${ }^{6}$ Each node would then multiply X by the node's assigned rows of A, and then send the result back to node 0 . The latter would collect those results, and store them in Y.

### 1.3.3.2 Example: MPI Prime Numbers Finder

Here we use the MPI system, with our hardware being a cluster.
MPI is a popular public-domain set of interface functions, callable from $\mathrm{C} / \mathrm{C}++$, to do message passing. We are again counting primes, though in this case using a pipelining method. It is similar to hardware pipelines, but in this case it is done in software, and each "stage" in the pipe is a different computer.

The program is self-documenting, via the comments.

```
/* MPI sample program; NOT INTENDED TO BE EFFICIENT as a prime
    finder, either in algorithm or implementation
    MPI (Message Passing Interface) is a popular package using
    the "message passing" paradigm for communicating between
    processors in parallel applications; as the name implies,
    processors communicate by passing messages using "send" and
    "receive" functions
    finds and reports the number of primes less than or equal to N
    uses a pipeline approach: node O looks at all the odd numbers (i.e.
    has already done filtering out of multiples of 2) and filters out
    those that are multiples of 3, passing the rest to node 1; node 1
    filters out the multiples of 5, passing the rest to node 2; node 2
    then removes the multiples of 7, and so on; the last node must check
    whatever is left
    note that we should NOT have a node run through all numbers
    before passing them on to the next node, since we would then
    have no parallelism at all; on the other hand, passing on just
    one number at a time isn't efficient either, due to the high
    overhead of sending a message if it is a network (tens of
    microseconds until the first bit reaches the wire, due to
    software delay); thus efficiency would be greatly improved if
    each node saved up a chunk of numbers before passing them to
    the next node */
#include <mpi.h> // mandatory
#define PIPE_MSG 0 // type of message containing a number to be checked
#define END_MSG 1 // type of message indicating no more data will be coming
int NNodes, // number of nodes in computation
```

[^4]```
    N, // find all primes from 2 to N
    Me; // my node number
double T1,T2; // start and finish times
void Init(int Argc,char **Argv)
{ int DebugWait;
    N = atoi(Argv[1]);
    // start debugging section
    DebugWait = atoi(Argv[2]);
    while (DebugWait) ; // deliberate infinite loop; see below
    /* the above loop is here to synchronize all nodes for debugging;
        if DebugWait is specified as 1 on the mpirun command line, all
        nodes wait here until the debugging programmer starts GDB at
        all nodes (via attaching to OS process number), then sets
        some breakpoints, then GDB sets DebugWait to 0 to proceed; */
    // end debugging section
    MPI_Init(&Argc,&Argv); // mandatory to begin any MPI program
    // puts the number of nodes in NNodes
    MPI_Comm_size(MPI_COMM_WORLD,&NNodes);
    // puts the node number of this node in Me
    MPI_Comm_rank(MPI_COMM_WORLD,&Me);
    // OK, get started; first record current time in T1
    if (Me == NNodes-1) T1 = MPI_Wtime();
}
void NodeO()
{ int I,ToCheck,Dummy,Error;
    for (I = 1; I <= N/2; I++) {
        ToCheck = 2* I + 1; // latest number to check for div3
        if (ToCheck > N) break;
        if (ToCheck % 3 > 0) // not divis by 3, so send it down the pipe
            // send the string at ToCheck, consisting of 1 MPI integer, to
            // node 1 among MPI_COMM_WORLD, with a message type PIPE_MSG
            Error = MPI_Send(&ToCheck,1,MPI_INT,1,PIPE_MSG,MPI_COMM_WORLD);
            // error not checked in this code
    }
    // sentinel
    MPI_Send(&Dummy,1,MPI_INT,1,END_MSG,MPI_COMM_WORLD);
}
void NodeBetween()
{ int ToCheck,Dummy,Divisor;
    MPI_Status Status;
    // first received item gives us our prime divisor
    // receive into Divisor 1 MPI integer from node Me-1, of any message
    // type, and put information about the message in Status
    MPI_Recv(&Divisor,1,MPI_INT,Me-1,MPI_ANY_TAG,MPI_COMM_WORLD,&Status);
    while (1) {
        MPI_Recv(&ToCheck,1,MPI_INT,Me-1,MPI_ANY_TAG,MPI_COMM_WORLD,&Status);
        // if the message type was END_MSG, end loop
        if (Status.MPI_TAG == END_MSG) break;
            if (ToCheck % Divisor > 0)
                MPI_Send(&ToCheck,1,MPI_INT,Me+1,PIPE_MSG,MPI_COMM_WORLD);
    }
    MPI_Send(&Dummy,1,MPI_INT,Me+1,END_MSG,MPI_COMM_WORLD);
}
NodeEnd()
```

```
{ int ToCheck,PrimeCount,I,IsComposite,StartDivisor;
    MPI_Status Status;
    MPI_Recv(&StartDivisor,1,MPI_INT,Me-1,MPI_ANY_TAG,MPI_COMM_WORLD,&Status);
    PrimeCount = Me + 2; /* must account for the previous primes, which
        won't be detected below */
    while (1) {
        MPI_Recv(&ToCheck,1,MPI_INT,Me-1,MPI_ANY_TAG,MPI_COMM_WORLD ,&Status);
        if (Status.MPI_TAG == END_MSG) break;
        IsComposite = 0;
        for (I = StartDivisor; I*I <= ToCheck; I += 2)
            if (ToCheck % I == 0) {
                    IsComposite = 1;
                    break;
            }
        if (!IsComposite) PrimeCount++;
    }
    /* check the time again, and subtract to find run time */
    T2 = MPI_Wtime();
    printf("elapsed time = %f\n",(float)(T2-T1));
    /* print results */
    printf("number of primes = % d\n",PrimeCount);
}
int main(int argc,char **argv)
{ Init(argc,argv);
    // all nodes run this same program, but different nodes take
    // different actions
    if (Me == 0) NodeO();
    else if (Me == NNodes-1) NodeEnd();
        else NodeBetween();
    // mandatory for all MPI programs
    MPI_Finalize();
}
/* explanation of "number of items" and "status" arguments at the end
    of MPI_Recv():
    when receiving a message you must anticipate the longest possible
    message, but the actual received message may be much shorter than
    this; you can call the MPI_Get_count() function on the status
    argument to find out how many items were actually received
    the status argument will be a pointer to a struct, containing the
    node number, message type and error status of the received
    message
    say our last parameter is Status; then Status.MPI_SOURCE
    will contain the number of the sending node, and
    Status.MPI_TAG will contain the message type; these are
    important if used MPI_ANY_SOURCE or MPI_ANY_TAG in our
    node or tag fields but still have to know who sent the
    message or what kind it is */
```

The set of machines can be heterogeneous, but MPI "translates" for you automatically. If say one node has a big-endian CPU and another has a little-endian CPU, MPI will do the proper conversion.

### 1.3.4 Scatter/Gather

Technically, the scatter/gather programmer world view is a special case of message passing. However, it has become so pervasive as to merit its own section here.

In this paradigm, one node, say node 0 , serves as a manager, while the others serve as workers. The parcels out work to the workers, who process their respective chunks of the data and return the results to the manager. The latter receives the results and combines them into the final product.

The matrix-vector multiply example in Section 1.3.3.1 is an example of scatter/gather.
As noted, scatter/gather is very popular. Here are some examples of packages that use it:

- MPI includes scatter and gather functions (Section 7.4).
- Hadoop/MapReduce Computing (Chapter 9) is basically a scatter/gather operation.
- The snow package (Section 1.3.4.1) for the R language is also a scatter/gather operation.


### 1.3.4.1 R snow Package

Base R does not include parallel processing facilities, but includes the parallel library for this purpose, and a number of other parallel libraries are available as well. The parallel package arose from the merger (and slight modifcation) of two former user-contributed libraries, snow and multicore. The former (and essentially the latter) uses the scatter/gather paradigm, and so will be introduced in this section; see Section 1.3.4.1 for further details. for convenience, I'll refer to the portion of parallel that came from snow simply as snow.

Let's use matrix-vector multiply as an example to learn from:

```
> library(parallel)
> c2<- makePSOCKcluster(rep("localhost",2))
>c2
socket cluster with 2 nodes on host localhost
> mmul
function(cls,u,v) {
    rowgrps <- splitIndices(nrow(u), length(cls))
    grpmul <- function(grp) u[grp,] %*% v
    mout <- clusterApply(cls, rowgrps,grpmul)
    Reduce(c,mout)
}
>a <- matrix(sample(1:50,16, replace=T), ncol=2)
>a
        [,1] [, 2]
[1,] 34 41
[2,] 10 28
```

```
\begin{tabular}{llrr}
17 & {\([3]\),} & 44 & 23 \\
18 & {\([4]\),} & 7 & 29 \\
19 & {\([5]\),} & 6 & 24 \\
20 & {\([6]\),} & 28 & 29 \\
21 & {\([7]\),} & 21 & 1 \\
22 & {\([8]\),} & 38 & 30 \\
23 & \(>b<-\) & \(\mathrm{c}(5,-2)\)
\end{tabular}
c(5,-2)
> b
[1] 5 -2
> a %*% b # serial multiply
    [,1]
    [1,] 88
    [2,] -6
    [3,] 174
    [4,] -23
    [5,] -18
    [6,] 82
    [7,] 103
    [8,] 130
    > clusterExport(c2,c("a","b")) # send a,b to workers
    > clusterEvalQ(c2,a) # check that they have it
    [[1]]
        [,1] [, 2]
        [1,] 34 41
        [2,] 10 28
        [3,] 44 23
        [4,]
        [5,] 6 24
        [6,] 28 29
        [7,] 21 1
        [8,] 38 30
        [[2]]
        [,1] [, 2]
        [1,] 34 41
        [2,] 10 28
        [3,] 44 23
        [4,] 7 29
        [5,]
        [6,] 28 29
        [7,] 21 1
        [8,] 38 30
        >mmul(c2,a,b) # test our parallel code
        [1] }8
```

What just happened?
First we set up a snow cluster. The term should not be confused with hardware systems we referred to as "clusters" earlier. We are simply setting up a group of R processes that will communicate
with each other via TCP/IP sockets.
In this case, my cluster consists of two R processes running on the machine from which I invoked makePSOCKcluster(). (In TCP/IP terminology, localhost refers to the local machine.) If I were to run the Unix ps command, with appropriate options, say ax, I'd see three R processes. I saved the cluster in c2.

On the other hand, my snow cluster could indeed be set up on a real cluster, e.g.

```
c3<- makePSOCKcluster(c(" pc28"," pc29"," pc29"))
```

where pc28 etc. are machine names.
In preparing to test my parallel code, I needed to ship my matrices $\mathbf{a}$ and $\mathbf{b}$ to the workers:

```
> clusterExport(c2,c("a","b")) # send a,b to workers
```

Note that this function assumes that $\mathbf{a}$ and $\mathbf{b}$ are global variables at the invoking node, i.e. the manager, and it will place copies of them in the global workspace of the worker nodes.

Note that the copies are independent of the originals; if a worker changes, say, $\mathbf{b}[\mathbf{3}]$, that change won't be made at the manager or at the other worker. This is a message-passing system, indeed.

So, how does the mmul code work? Here's a handy copy:

```
mmul<- function(cls,u,v) {
    rowgrps <- splitIndices(nrow(u), length(cls))
    grpmul <- function(grp) u[grp,] %*% v
    mout <- clusterApply(cls, rowgrps,grpmul)
    Reduce(c,mout)
}
```

As discussed in Section 1.3.1, our strategy will be to partition the rows of the matrix, and then have different workers handle different groups of rows. Our call to splitIndices() sets this up for us.

That function does what its name implies, e.g.

```
> splitIndices (12,5)
[[1]]
[1] 1 2 3
[[2]]
[1] 4 5
[[3]]
[1] 6 7
[[4]]
[1] }8
```

```
[[5]]
[1] 10}1011\quad1
```

Here we asked the function to partition the numbers $1, \ldots, 12$ into 5 groups, as equal-sized as possible, which you can see is what it did. Note that the type of the return value is an R list.

So, after executing that function in our mmul() code, rowgrps will be an R list consisting of a partitioning of the row numbers of $\mathbf{u}$, exactly what we need.
The call to clusterApply () is then where the actual work is assigned to the workers. The code

```
mout <- clusterApply(cls,rowgrps,grpmul)
```

instructs snow to have the first worker process the rows in rowgrps[[1]], the second worker to work on rowgrps[[2]], and so on. The clusterApply() function expects its second argument to be an R list, which is the case here.

Each worker will then multiply $\mathbf{v}$ by its row group, and return the product to the manager. However, the product will again be a list, one component for each worker, so we need Reduce() to string everything back together.

Note that R does allow functions defined within functions, which the locals and arguments of the outer function becoming global to the inner function.

Note that a here could have been huge, in which case the export action could slow down our program. If a were not needed at the workers other than for this one-time matrix multiply, we may wish to change to code so that we send each worker only the rows of a that we need:

```
mmul1 <- function(cls,u,v) {
    rowgrps <- splitIndices(nrow(u), length(cls))
    uchunks <- Map(function(grp) u[grp,], rowgrps)
    mulchunk <- function(uc) uc %**% v
    mout <- clusterApply(cls,uchunks,mulchunk)
    Reduce(c,mout)
}
```

Let's test it:

```
> a <- matrix(sample(1:50,16,replace=T),ncol=2)
>b<- c(5,-2)
> clusterExport(c2,"b") # don't send a
    a
    [,1] [, 2]
[1,] 10 26
[2,] 1 34
[3,] 49 30
[4,] 39 41
[5,]
```

```
11 [6,] 2 30
12 [7,] 33 23
13 [8,] 44 5
14>a %*% b
15 [,1]
16 [1,] -2
17 [2,] -63
18 [3,] 185
19 [4,] 113
20 [5,] 32
21 [6,] -50
22 [7,] 119
23 [8,] 210
24> mmul1(c2,a,b)
25 [1] [-2 -63 185 113 [
```

Note that we did not need to use clusterExport() to send the chunks of a to the workers, as the call to clusterApply () does this, since it sends the arguments,

## Chapter 2

## Recurring Performance Issues

Oh no! It's actually slower in parallel! - almost everyone's exclamation the first time they try to parallelize code

The available parallel hardware systems sound wonderful at first. But everyone who uses such systems has had the experience of enthusiastically writing his/her first parallel program, anticipating great speedups, only to find that the parallel code actually runs more slowly than the original nonparallel program.

In this chapter, we highlight some major issues that will pop up throughout the book.

### 2.1 Communication Bottlenecks

Whether you are on a shared-memory, message-passing or other platform, communication is always a potential bottleneck:

- On a shared-memory system, the threads must contend with each other in communicating with memory. And the problem is exacerbated by cache coherency transactions (Section 3.5.1.
- On a NOW, even a very fast network is very slow compared to CPU speeds.
- GPUs are really fast, but their communication with their CPU hosts is slow. There are also memory contention issues as in ordinary shared-memory systems.

Among other things, communication considerations largely drive the load balancing issue, discussed next.

### 2.2 Load Balancing

Arguably the most central performance issue is load balancing, i.e. keeping all the processors busy as much as possible. This issue arises constantly in any discussion of parallel processing.

A nice, easily understandable example is shown in Chapter 7 of the book, Multicore Application Programming: for Windows, Linux and Oracle Solaris, Darryl Gove, 2011, Addison-Wesley. There the author shows code to compute the Mandelbrot set, defined as follows.

Start with any number c in the complex plane, and initialize z to 0 . Then keep applying the transformation

$$
\begin{equation*}
z \leftarrow z^{2}+c \tag{2.1}
\end{equation*}
$$

If the resulting sequence remains bounded (say after a certain number of iterations), we say that c belongs to the Mandelbrot set.

Gove has a rectangular grid of points in the plane, and wants to determine whether each point is in the set or not; a simple but time-consuming computation is used for this determination 1

Gove sets up two threads, one handling all the points in the left half of the grid and the other handling the right half. He finds that the latter thread is very often idle, while the former thread is usually busy - extremely poor load balance. We'll return to this issue in Section 2.4 .

## 2.3 "Embarrassingly Parallel" Applications

The term embarrassingly parallel is heard often in talk about parallel programming.

### 2.3.1 What People Mean by "Embarrassingly Parallel"

Consider a matrix multiplication application, for instance, in which we compute AX for a matrix A and a vector X . One way to parallelize this problem would be to have each processor handle a group of rows of A , multiplying each by X in parallel with the other processors, which are handling other groups of rows. We call the problem embarrassingly parallel, with the word "embarrassing" meaning that the problem is too easy, i.e. there is no intellectual challenge involved. It is pretty obvious that the computation $\mathrm{Y}=\mathrm{AX}$ can be parallelized very easily by splitting the rows of A into groups.

[^5]By contrast, most parallel sorting algorithms require a great deal of interaction. For instance, consider Mergesort. It breaks the vector to be sorted into two (or more) independent parts, say the left half and right half, which are then sorted in parallel by two processes. So far, this is embarrassingly parallel, at least after the vector is broken in half. But then the two sorted halves must be merged to produce the sorted version of the original vector, and that process is not embarrassingly parallel; it can be parallelized, but in a more complex, less obvious manner.

Of course, it's no shame to have an embarrassingly parallel problem! On the contrary, except for showoff academics, having an embarrassingly parallel application is a cause for celebration, as it is easy to program.

In recent years, the term embarrassingly parallel has drifted to a somewhat different meaning. Algorithms that are embarrassingly parallel in the above sense of simplicity tend to have very low communication between processes, key to good performance. That latter trait is the center of attention nowadays, so the term embarrassingly parallel generally refers to an algorithm with low communication needs.

For that reason, many people would NOT considered even our prime finder example in Section 1.3 .2 .2 to be embarrassingly parallel. Yes, it was embarrassingly easy to write, but it has high communication costs, as both its locks and its global array are accessed quite often.

On the other hand, the Mandelbrot computation described in Section 2.2 is truly embarrassingly parallel, in both the old and new sense of the term. There the author Gove just assigned the points on the left to one thread and the rest to the other thread-very simple - and there was no communication between them.

### 2.3.2 Iterative Algorithms

Many parallel algorithms involve iteration, with a rendezvous of the tasks after each iteration. Within each iteration, the nodes act entirely independently of each other, which makes the problem seem embarrassingly parallel.

But unless the granularity of the problem is coarse, i.e. there is a large amount of work to do in each iteration, the communication overhead will be significant, and the algorithm may not be considered embarrassingly parallel.

### 2.4 Static (But Possibly Random) Task Assignment Typically Better Than Dynamic

Say an algorithm generates t independent ${ }^{2}$ tasks and we have p processors to handle them. In our matrix-times-vector example of Section 1.3.1, say, each row of the matrix might be considered one task. A processor's work would then be to multiply the vector by this processor's assigned rows of the matrix.

How do we decide which tasks should be done by which processors? In static assignment, our code would decide at the outset which processors will handle which tasks. The alternative, dynamic assignment, would have processors determine their tasks as the computation proceeds.

In the matrix-times-vector example, say we have 10000 rows and 10 processors. In static task assignment, we could pre-assign processor 0 rows 0-999, processor 1 rows 1000-1999 and so on. On the other hand, we could set up a task farm, a queue consisting here of the numbers 0-9999. Each time a processor finished handling one row, it would remove the number at the head of the queue, and then process the row with that index.

It would at first seem that dynamic assignment is more efficient, as it is more flexible. However, accessing the task farm, for instance, entails communication costs, which might be very heavy. In this section, we will show that it's typically better to use the static approach, though possibly randomized $3^{3}$

### 2.4.1 Example: Matrix-Vector Multiply

Consider again the problem of multiplying a vector X by a large matrix A , yielding a vector Y. Say A has 10000 rows and we have 10 threads. Let's look at little closer at the static/dynamic tradeoff outlined above. For concreteness, assume the shared-memory setting.

There are several possibilities here:

- Method A: We could simply divide the 10000 rows into chunks of $10000 / 10=1000$, and parcel them out to the threads. We would pre-assign thread 0 to work on rows $0-999$ of A , thread 1 to work on rows 1000-1999 and so on.

This is essentially OpenMP's static scheduling policy, with default chunk size $\mathbb{4}^{4}$
There would be no communication between the threads this way, but there could be a problem of load imbalance. Say for instance that by chance thread 3 finishes well before the others.

[^6]
### 2.4. STATIC (BUT POSSIBLY RANDOM) TASK ASSIGNMENT TYPICALLY BETTER THAN DYNAMIC29

Then it will be idle, as all the work had been pre-allocated.

- Method B: We could take the same approach as in Method A, but with a chunk size of, say, 100 instead of 1000 . This is OpenMP's static policy again, but with a chunk size of 100 .
If we didn't use OpenMP (which would internally do the following anyway, in essence), we would have a shared variable named,say, nextchunk similar to nextbase in our prime-finding program in Section 1.3.2.2. Each time a thread would finish a chunk, it would obtain a new chunk to work on, by recording the value of nextchunk and incrementing that variable by 1 (all atomically, of course).

This approach would have better load balance, because the first thread to find there is no work left to do would be idle for at most 100 rows' amount of computation time, rather than 1000 as above. Meanwhile, though, communication would increase, as access to the locks around nextchunk would often make one thread wait for another 5

- Method C: So, Method A above minimizes communication at the possible expense of load balance, while the Method B does the opposite.

OpenMP also offers the guided policy, which is like dynamic except the chunk size decreases over time.

I will now show that in typical settings, the Method A above (or a slight modification) works well. To this end, consider a chunk consisting of $m$ tasks, such as $m$ rows in our matrix example above, with times $T_{1}, T_{2}, \ldots, T_{m}$. The total time needed to process the chunk is then $T_{1}+\ldots, T_{m}$.

The $T_{i}$ can be considered random variables; some tasks take a long time to perform, some take a short time, and so on. As an idealized model, let's treat them as independent and identically distributed random variables. Under that assumption (if you don't have the probability background, follow as best you can), we have that the mean (expected value) and variance of total task time are

$$
E\left(T_{1}+\ldots, T_{m}\right)=m E\left(T_{1}\right)
$$

and

$$
\operatorname{Var}\left(T_{1}+\ldots, T_{m}\right)=m \operatorname{Var}\left(T_{1}\right)
$$

Thus

[^7]$$
\frac{\text { standard deviation of chunk time }}{\text { mean of chunk time }} \sim O\left(\frac{1}{\sqrt{m}}\right)
$$

In other words:

- run time for a chunk is essentially constant if k is large, and
- there is essentially no load imbalance in Method A

Since load imbalance was the only drawback to Method A and we now see it's not a problem after all, then Method A is best.

For more details and timing examples, see N. Matloff, "Efficient Parallel R Loops on Long-Latency Platforms," Proceedings of the $42 n d$ Interface between Statistics and Computer Science, Rice University, June 2012 ${ }^{6}$

### 2.4.2 Load Balance, Revisited

But what about the assumptions behind that reasoning? Consider for example the Mandelbrot problem in Section 2.2. There were two threads, thus two chunks, with the tasks for a given chunk being computations for all the points in the chunk's assigned region of the picture.

Gove noted there was fairly strong load imbalance here, and that the reason was that most of the Mandelbrot points turned out to be in the left half of the picture! The computation for a given point is iterative, and if a point is not in the set, it tends to take only a few iterations to discover this. That's why the thread handling the right half of the picture was idle so often.

So Method A would not work well here, and upon reflection one can see that the problem was that the tasks within a chunk were not independent, but were instead highly correlated, thus violating our mathematical assumptions above. Of course, before doing the computation, Gove didn't know that it would turn out that most of the set would be in the left half of the picture. But, one could certainly anticipate the correlated nature of the points; if one point is not in the Mandelbrot set, its near neighbors are probably not in it either.

But Method A can still be made to work well, via a simple modification: Simply form the chunks randomly. In the matrix-multiply example above, with 10000 rows and chunk size 1000, do NOT assign the chunks contiguously. Instead, generate a random permutation of the numbers $0,1, \ldots, 9999$, naming them $i_{0}, i_{1}, \ldots, i_{9999}$. Then assign thread 0 rows $i_{0}-i_{999}$, thread 1 rows $i_{1000}-i_{1999}$, etc.

[^8]
### 2.4. STATIC (BUT POSSIBLY RANDOM) TASK ASSIGNMENT TYPICALLY BETTER THAN DYNAMIC31

In the Mandelbrot example, we could randomly assign rows of the picture, in the same way, and avoid load imbalance.

So, actually, Method A, or let's call it Method A', will still typically work well.

### 2.4.3 Example: Mutual Web Outlinks

Here's an example that we'll use at various points in this book:

## Mutual outlinks in a graph:

Consider a network graph of some kind, such as Web links. For any two vertices, say any two Web sites, we might be interested in mutual outlinks, i.e. outbound links that are common to two Web sites. Say we want to find the number number of mutual outlinks, averaged over all pairs of Web sites.

Let A be the adjacency matrix of the graph. Then the mean of interest would be found as follows:

```
sum \(=0\)
for \(\mathrm{i}=0 \ldots \mathrm{n}-2\)
    for \(\mathrm{j}=\mathrm{i}+1 \ldots \mathrm{n}-1\)
        count \(=0\)
        for \(\mathrm{k}=0 \ldots \mathrm{n}-1\) count \(+=\mathrm{a}[\mathrm{i}][\mathrm{k}] * \mathrm{a}[\mathrm{j}][\mathrm{k}]\)
mean \(=\operatorname{sum} /(n *(n-1) / 2)\)
```

Say again $n=10000$ and we have 10 threads. We should not simply assign work to the threads by dividing up the i loop, with thread 0 taking the cases $\mathrm{i}=0, \ldots, 999$, thread 1 the cases $1000, \ldots, 1999$ and so on. This would give us a real load balance problem. Thread 8 would have much less work to do than thread 3 , say.

We could randomize as discussed earlier, but there is a much better solution: Just pair the rows of A. Thread 0 would handle rows $0, \ldots, 499$ and $9500, \ldots, 9999$, thread 1 would handle rows 500,999 and $9000, \ldots, 9499$ etc. This approach is taken in our OpenMP implementation, Section 4.12 .

In other words, Method A still works well.

In the mutual outlinks problem, we have a good idea beforehand as to how much time each task needs, but this may not be true in general. An alternative would be to do random pre-assignment of tasks to processors.

On the other hand, if we know beforehand that all of the tasks should take about the same time, we should use static scheduling, as it might yield better cache and virtual memory performance.

### 2.4.4 Work Stealing

There is another variation to Method A that is of interest today, called work stealing. Here a thread that finishes its assigned work and has thus no work left to do will "raid" the work queue of some other thread. This is the approach taken, for example, by the elegant Cilk language. Needless to say, accessing the other work queue is going to be expensive in terms of time and memory contention overhead.

### 2.4.5 Timing Example

I ran the Mandelbrot example on a shared memory machine with four cores, two threads per core, with the following results for eight threads, on an 8000 x 8000 grid:

| policy | time |
| ---: | ---: |
| static | 47.8 |
| dynamic | 21.4 |
| guided | 29.6 |
| random | 15.7 |

Default values were used for chunk size in the first three cases. I did try other chunk sizes for the dynamic policy, but it didn't make much difference. See Section 4.4 for the code.

Needless to say, one shouldn't overly extrapolate from the above timings, but it does illustrate the issues.

### 2.5 Latency and Bandwidth

We've been speaking of communications delays so far as being monolithic, but they are actually (at least) two-dimensional. The key measures are latency and bandwidth:

- Latency is the time it takes for one bit to travel for source to destination, e.g. from a CPU to memory in a shared memory system, or from one computer to another in a cluster.
- Bandwidth is the number of bits per unit time that can be input into the communications channel. This can be affected by factors such as bus width in a shared memory system and number of parallel network paths in a message passing system, and also by the speed of the links.

It's helpful to think of a bridge, with toll booths at its entrance. Latency is the time needed for one car to get from one end of the bridge to the other. Bandwidth is the number of cars that can enter
the bridge per unit time. We can reduce latency by increasing the speed limit, and can increase bandwidth by improving the speed by which toll takers can collect tolls, and increasing the number of toll booths.

## Latency hiding:

One way of dealing with long latencies is known as latency hiding. The idea is to do a long-latency operation in parallel with something else.

For example, GPUs tend to have very long memory access times, but this is solved by having many pending memory accesses at the same time. During the latency of some accesses, earlier ones that have now completed can now be acted upon (Section 5.3.3.2).

### 2.6 Relative Merits: Performance of Shared-Memory Vs. MessagePassing

My own preference is shared-memory, but there are pros and cons to each paradigm.
It is generally believed in the parallel processing community that the shared-memory paradigm produces code that is easier to write, debug and maintain than message-passing. See for instance R. Chandra, Parallel Programming in OpenMP, MKP, 2001, pp.10ff (especially Table 1.1), and M. Hess et al, Experiences Using OpenMP Based on Compiler Directive Software DSM on a PC Cluster, in OpenMP Shared Memory Parallel Programming: International Workshop on OpenMP Applications and Tools, Michael Voss (ed.), Springer, 2003, p.216.

On the other hand, in some cases message-passing can produce faster code. Consider the Odd/Even Transposition Sort algorithm, for instance. Here pairs of processes repeatedly swap sorted arrays with each other. In a shared-memory setting, this might produce a bottleneck at the shared memory, slowing down the code. Of course, the obvious solution is that if you are using a shared-memory machine, you should just choose some other sorting algorithm, one tailored to the shared-memory setting.

There used to be a belief that message-passing was more scalable, i.e. amenable to very large systems. However, GPU has demonstrated that one can achieve extremely good scalability with shared-memory.

As will be seen, though, GPU is hardly a panacea. Where, then, are people to get access to largescale parallel systems? Most people do not (currently) have access to large-scale multicore machines, while most do have access to large-scale message-passing machines, say in cloud computing venues. Thus message-passing plays a role even for those of us who preferred the shared-memory paradigm.

Also, hybrid systems are common, in which a number of shared-memory systems are tied together
by, say, MPI.

### 2.7 Memory Allocation Issues

Many algorithms require large amounts of memory for intermediate storage of data. It may be prohibitive to allocate this memory statically, i.e. at compile time. Yet dynamic allocation, say via malloc() or C++'s new (which probably produces a call to malloc() anyway, is very expensive in time.

Using large amounts of memory also can be a major source of overhead due to cache misses and page faults.

One way to avoid malloc(), of course, is to set up static arrays whenever possible.
There are no magic solutions here. One must simply be aware of the problem, and tweak one's code accordingly, say by adjusting calls to malloc() so that one achieves a balance between allocating too much memory and making too many calls.

### 2.8 Issues Particular to Shared-Memory Systems

This topic is covered in detail in Chapter 3, but is so important that the main points should be mentioned here.

- Memory is typically divided into banks. If more than one thread attempts to access the same bank at the same time, that effectively serializes the program.
- There is typically a cache at each processor. Keeping the contents of these caches consistent with each other, and with the memory itself, adds a lot of overhead, causing slowdown.

In both cases, awareness of these issues should impact how you write your code.
See Sections 3.2 and 3.5.

## Chapter 3

## Shared Memory Parallelism

Shared-memory programming is considered by many in the parallel processing community as being the clearest of the various parallel paradigms available.

Note: To get the most of this section-which is used frequently in the rest of this book-you may wish to read the material on array storage in the appendix of this book, Section A.3.1.

### 3.1 What Is Shared?

The term shared memory means that the processors all share a common address space. Say this is occurring at the hardware level, and we are using Intel Pentium CPUs. Suppose processor P3 issues the instruction
movl 200, \%eabx
which reads memory location 200 and places the result in the EAX register in the CPU. If processor P 4 does the same, they both will be referring to the same physical memory cell. In non-sharedmemory machines, each processor has its own private memory, and each one will then have its own location 200, completely independent of the locations 200 at the other processors' memories.

Say a program contains a global variable $\mathbf{X}$ and a local variable $\mathbf{Y}$ on share-memory hardware (and we use shared-memory software). If for example the compiler assigns location 200 to the variable $\mathbf{X}$, i.e. $\& X=200$, then the point is that all of the processors will have that variable in common, because any processor which issues a memory operation on location 200 will access the same physical memory cell.

On the other hand, each processor will have its own separate run-time stack. All of the stacks are in shared memory, but they will be accessed separately, since each CPU has a different value in its SP (Stack Pointer) register. Thus each processor will have its own independent copy of the local variable $\mathbf{Y}$.

To make the meaning of "shared memory" more concrete, suppose we have a bus-based system, with all the processors and memory attached to the bus. Let us compare the above variables $\mathbf{X}$ and $\mathbf{Y}$ here. Suppose again that the compiler assigns $\mathbf{X}$ to memory location 200. Then in the machine language code for the program, every reference to $\mathbf{X}$ will be there as 200 . Every time an instruction that writes to $\mathbf{X}$ is executed by a CPU, that CPU will put 200 into its Memory Address Register (MAR), from which the 200 flows out on the address lines in the bus, and goes to memory. This will happen in the same way no matter which CPU it is. Thus the same physical memory location will end up being accessed, no matter which CPU generated the reference.

By contrast, say the compiler assigns a local variable $\mathbf{Y}$ to something like ESP +8 , the third item on the stack (on a 32 -bit machine), 8 bytes past the word pointed to by the stack pointer, ESP. The OS will assign a different ESP value to each thread, so the stacks of the various threads will be separate. Each CPU has its own ESP register, containing the location of the stack for whatever thread that CPU is currently running. So, the value of $\mathbf{Y}$ will be different for each thread.

### 3.2 Memory Modules

Parallel execution of a program requires, to a large extent, parallel accessing of memory. To some degree this is handled by having a cache at each CPU, but it is also facilitated by dividing the memory into separate modules or banks. This way several memory accesses can be done simultaneously.

In this section, assume for simplicity that our machine has 32 -bit words. This is still true for many GPUs, in spite of the widespread use of 64 -bit general-purpose machines today, and in any case, the numbers here can easily be converted to the 64 -bit case.

Note that this means that consecutive words differ in address by 4 . Let's thus define the wordaddress of a word to be its ordinary address divided by 4 . Note that this is also its address with the lowest two bits deleted.

### 3.2.1 Interleaving

There is a question of how to divide up the memory into banks. There are two main ways to do this:
(a) High-order interleaving: Here consecutive words are in the same bank (except at boundaries). For example, suppose for simplicity that our memory consists of word-addresses 0 through 1023, and that there are four banks, M0 through M3. Then M0 would contain word-addresses $0-255$, M1 would have $256-511$, M2 would have $512-767$, and M3 would have 768-1023.
(b) Low-order interleaving: Here consecutive addresses are in consecutive banks (except when we get to the right end). In the example above, if we used low-order interleaving, then wordaddress 0 would be in M0, 1 would be in M1, 2 would be in M2, 3 would be in M3, 4 would be back in M0, 5 in M1, and so on.

Say we have eight banks. Then under high-order interleaving, the first three bits of a word-address would be taken to be the bank number, with the remaining bits being address within bank. Under low-order interleaving, the hree least significant bits would be used to determine bank number.

Low-order interleaving has often been used for vector processors. On such a machine, we might have both a regular add instruction, ADD , and a vector version, VADD. The latter would add two vectors together, so it would need to read two vectors from memory. If low-order interleaving is used, the elements of these vectors are spread across the various banks, so fast access is possible.

A more modern use of low-order interleaving, but with the same motivation as with the vector processors, is in GPUs (Chapter 5 ).

High-order interleaving might work well in matrix applications, for instance, where we can partition the matrix into blocks, and have different processors work on different blocks. In image processing applications, we can have different processors work on different parts of the image. Such partitioning almost never works perfectly - e.g. computation for one part of an image may need information from another part - but if we are careful we can get good results.

### 3.2.2 Bank Conflicts and Solutions

Consider an array $\mathbf{x}$ of 16 million elements, whose sum we wish to compute, say using 16 threads. Suppose we have four memory banks, with low-order interleaving.

A naive implementation of the summing code might be

```
parallel for thr = 0 to 15
    localsum = 0
    for j = 0 to 999999
        localsum += x[thr*1000000+j]
    grandsum += localsumsum
```

In other words, thread 0 would sum the first million elements, thread 1 would sum the second million, and so on. After summing its portion of the array, a thread would then add its sum to a
grand total. (The threads could of course add to grandsum directly in each iteration of the loop, but this would cause too much traffic to memory, thus causing slowdowns.)

Suppose for simplicity that the threads run in lockstep, so that they all attempt to access memory at once. On a multicore/multiprocessor machine, this may not occur, but it in fact typically will occur in a GPU setting.

A problem then arises. To make matters simple, suppose that $\mathbf{x}$ starts at an address that is a multiple of 4 , thus in bank 0 . (The reader should think about how to adjust this to the other three cases.) On the very first memory access, thread 0 accesses $\mathbf{x}[0]$ in bank 0 , thread 1 accesses $\mathrm{x}[\mathbf{1 0 0 0 0 0 0}]$, also in bank 0 , and so on-and these will all be in memory bank 0 ! Thus there will be major conflicts, hence major slowdown.

A better approach might be to have any given thread work on every sixteenth element of $\mathbf{x}$, instead of on contiguous elements. Thread 0 would work on $\mathrm{x}[\mathbf{1 0 0 0 0 0 0}], \mathrm{x}[\mathbf{1 0 0 0 0 1 6}], \mathrm{x}[\mathbf{1 0 0 0 0 0 3 2}, \ldots$; thread 1 would handle $\mathbf{x [ 1 0 0 0 0 0 1 ] , ~} \mathbf{x [ 1 0 0 0 0 1 7 ]}, \mathbf{x [ 1 0 0 0 0 0 3 3 , \ldots}$; and so on:

```
parallel for thr = 0 to 15
    localsum = 0
    for j = 0 to 999999
        localsum += x[16*j+thr]
    grandsum += localsumsum
```

Here, consecutive threads work on consecutive elements in $\mathbf{x} \|^{1}$ That puts them in separate banks, thus no conflicts, hence speedy performance.

In general, avoiding bank conflicts is an art, but there are a couple of approaches we can try.

- We can rewrite our algorithm, e.g. use the second version of the above code instead of the first.
- We can add padding to the array. For instance in the first version of our code above, we could lengthen the array from 16 million to 16000016 , placing padding in words 1000000 , 2000001 and so on. We'd tweak our array indices in our code accordingly, and eliminate bank conflicts that way.

In the first approach above, the concept of stride often arises. It is defined to be the distance betwwen array elements in consecutive accesses by a thread. In our original code to compute grandsum, the stride was 1 , since each array element accessed by a thread is 1 past the last access by that thread. In our second version, the stride was 16.

Strides of greater than 1 often arise in code that deals with multidimensional arrays. Say for example we have two-dimensional array with 16 columns. In $\mathrm{C} / \mathrm{C}++$, which uses row-major order,

[^9]access of an entire column will have a stride of 16 . Access down the main diagonal will have a stride of 17 .

Suppose we have b banks, again with low-order interleaving. You should experiment a bit to see that an array access with a stride of $s$ will access $s$ different banks if and only if $s$ and $b$ are relatively prime, i.e. the greatest common divisor of s and b is 1 . This can be proven with group theory.

Another strategy, useful for collections of complex objects, is to set up structs of arrays rather than arrays of structs. Say for instance we are working with data on workers, storing for each worker his name, salary and number of years with the firm. We might naturally write code like this:

```
struct {
    char name[25];
    float salary;
    float yrs;
} x[100];
```

That gives a 100 structs for 100 workers. Again, this is very natural, but it may make for poor memory access patterns. Salary values for the various workers will no longer be contiguous, for instance, even though the structs are contiguous. This could cause excessive cache misses.

One solution would be to add padding to each struct, so that the salary values are a word apart in memory. But another approach would be to replace the above arrays of structs by a struct of arrays:

```
struct {
    char *name[]100;
    float salary[100];
    float yrs[100];
}
```


### 3.2.3 Example: Code to Implement Padding

As discussed above, array padding is used to try to get better parallel access to memory banks. The code below is aimed to provide utilities to assist in this. Details are explained in the comments.

```
// routines to initialize, read and write
// padded versions of a matrix of floats;
// the matrix is nominally mxn, but its
// rows will be padded on the right ends,
// so as to enable a stride of s down each
// column; it is assumed that s >= n
// allocate space for the padded matrix,
```

```
// initially empty
float *padmalloc(int m, int n, int s) {
        return(malloc(m*s*sizeof(float)));
}
// store the value tostore in the matrix q,
// at row i, column j; m, n and
// s are as in padmalloc() above
void setter(float *q, int m, int n, int s,
        int i, int j, float tostore) {
    *(q+i*s+j) = tostore;
}
// fetch the value in the matrix q,
// at row i, column j; m, n and s are
// as in padmalloc() above
float getter(float *q, int m, int n, int s,
        int i, int j) {
        return *(q + i * S + j);
}
```


### 3.3 Interconnection Topologies

### 3.3.1 SMP Systems

A Symmetric Multiprocessor (SMP) system has the following structure:
bus


Here and below:

- The Ps are processors, e.g. off-the-shelf chips such as Pentiums.
- The Ms are memory modules. These are physically separate objects, e.g. separate boards of memory chips. It is typical that there will be the same number of Ms as Ps.
- To make sure only one P uses the bus at a time, standard bus arbitration signals and/or arbitration devices are used.
- There may also be coherent caches, which we will discuss later.


### 3.3.2 NUMA Systems

In a Nonuniform Memory Access (NUMA) architecture, each CPU has a memory module physically next to it, and these processor/memory ( $\mathrm{P} / \mathrm{M}$ ) pairs are connected by some kind of network.

Here is a simple version:


Each $\mathrm{P} / \mathrm{M} / \mathrm{R}$ set here is called a processing element (PE). Note that each PE has its own local bus, and is also connected to the global bus via $R$, the router.

Suppose for example that P3 needs to access location 200, and suppose that high-order interleaving is used. If location 200 is in M3, then P3's request is satisfied by the local bus ${ }^{2}$ On the other hand, suppose location 200 is in M8. Then the R3 will notice this, and put the request on the global bus, where it will be seen by R8, which will then copy the request to the local bus at PE8, where the request will be satisfied. (E.g. if it was a read request, then the response will go back from M8 to R8 to the global bus to R3 to P3.)

It should be obvious now where NUMA gets its name. P8 will have much faster access to M8 than P3 will to M8, if none of the buses is currently in use - and if say the global bus is currently in use, P3 will have to wait a long time to get what it wants from M8.

Today almost all high-end MIMD systems are NUMAs. One of the attractive features of NUMA is that by good programming we can exploit the nonuniformity. In matrix problems, for example, we can write our program so that, for example, P8 usually works on those rows of the matrix which are stored in M8, P3 usually works on those rows of the matrix which are stored in M3, etc. In order to do this, we need to make use of the C language's \& address operator, and have some knowledge of the memory hardware structure, i.e. the interleaving.

[^10]
### 3.3.3 NUMA Interconnect Topologies

The problem with a bus connection, of course, is that there is only one pathway for communication, and thus only one processor can access memory at the same time. If one has more than, say, two dozen processors are on the bus, the bus becomes saturated, even if traffic-reducing methods such as adding caches are used. Thus multipathway topologies are used for all but the smallest systems. In this section we look at two alternatives to a bus topology.

### 3.3.3.1 Crossbar Interconnects

Consider a shared-memory system with n processors and n memory modules. Then a crossbar connection would provide $n^{2}$ pathways. E.g. for $\mathrm{n}=8$ :


Generally serial communication is used from node to node, with a packet containing information on both source and destination address. E.g. if P2 wants to read from M5, the source and destination will be 3 -bit strings in the packet, coded as 010 and 101 , respectively. The packet will also contain bits which specify which word within the module we wish to access, and bits which specify whether we wish to do a read or a write. In the latter case, additional bits are used to specify the value to be written.

Each diamond-shaped node has two inputs (bottom and right) and two outputs (left and top), with buffers at the two inputs. If a buffer fills, there are two design options: (a) Have the node from which the input comes block at that output. (b) Have the node from which the input comes discard the packet, and retry later, possibly outputting some other packet for now. If the packets at the heads of the two buffers both need to go out the same output, the one (say) from the bottom input will be given priority.

There could also be a return network of the same type, with this one being memory $\rightarrow$ processor, to return the result of the read requests ${ }^{3}$

Another version of this is also possible. It is not shown here, but the difference would be that at the bottom edge we would have the PEi and at the left edge the memory modules Mi would be replaced by lines which wrap back around to PEi, similar to the Omega network shown below.

Crossbar switches are too expensive for large-scale systems, but are useful in some small systems. The 16 -CPU Sun Microsystems Enterprise 10000 system includes a $16 x 16$ crossbar.

### 3.3.3.2 Omega (or Delta) Interconnects

These are multistage networks similar to crossbars, but with fewer paths. Here is an example of a NUMA 8x8 system:


Recall that each PE is a processor/memory pair. PE3, for instance, consists of P3 and M3.
Note the fact that at the third stage of the network (top of picture), the outputs are routed back to the PEs, each of which consists of a processor and a memory module $4^{4}$

At each network node (the nodes are the three rows of rectangles), the output routing is done by destination bit. Let's number the stages here 0,1 and 2 , starting from the bottom stage, number the nodes within a stage $0,1,2$ and 3 from left to right, number the PEs from 0 to 7 , left to right, and number the bit positions in a destination address 0,1 and 2 , starting from the most significant bit. Then at stage i , bit i of the destination address is used to determine routing, with a 0 meaning routing out the left output, and 1 meaning the right one.

Say P2 wishes to read from M5. It sends a read-request packet, including $5=101$ as its destination address, to the switch in stage 0 , node 1 . Since the first bit of 101 is 1 , that means that this switch will route the packet out its right-hand output, sending it to the switch in stage 1 , node 3 . The latter switch will look at the next bit in 101, a 0 , and thus route the packet out its left output, to the switch in stage 2, node 2. Finally, that switch will look at the last bit, a 1, and output out

[^11]its right-hand output, sending it to PE5, as desired. M5 will process the read request, and send a packet back to PE2, along the same

Again, if two packets at a node want to go out the same output, one must get priority (let's say it is the one from the left input).

Here is how the more general case of $\mathrm{N}=2^{n}$ PEs works. Again number the rows of switches, and switches within a row, as above. So, $S_{i j}$ will denote the switch in the i-th row from the bottom and j -th column from the left (starting our numbering with 0 in both cases). Row i will have a total of N input ports $I_{i k}$ and N output ports $O_{i k}$, where $\mathrm{k}=0$ corresponds to the leftmost of the N in each case. Then if row i is not the last row $(i<n-1), O_{i k}$ will be connected to $I_{j m}$, where $\mathrm{j}=$ i +1 and

$$
\begin{equation*}
m=(2 k+\lfloor(2 k) / N\rfloor) \bmod N \tag{3.1}
\end{equation*}
$$

If row i is the last row, then $O_{i k}$ will be connected to, PE k .

### 3.3.4 Comparative Analysis

In the world of parallel architectures, a key criterion for a proposed feature is scalability, meaning how well the feature performs as we go to larger and larger systems. Let $n$ be the system size, either the number of processors and memory modules, or the number of PEs. Then we are interested in how fast the latency, bandwidth and cost grow with n:

| criterion | bus | Omega | crossbar |
| :--- | ---: | ---: | ---: |
| latency | $\mathrm{O}(1)$ | $\mathrm{O}\left(\log _{2} \mathrm{n}\right)$ | $\mathrm{O}(\mathrm{n})$ |
| bandwidth | $\mathrm{O}(1)$ | $\mathrm{O}(\mathrm{n})$ | $\mathrm{O}(\mathrm{n})$ |
| cost | $\mathrm{O}(1)$ | $\mathrm{O}\left(\mathrm{n} \log _{2} \mathrm{n}\right)$ | $\mathrm{O}\left(\mathrm{n}^{2}\right)$ |

Let us see where these expressions come from, beginning with a bus: No matter how large $n$ is, the time to get from, say, a processor to a memory module will be the same, thus $\mathrm{O}(1)$. Similarly, no matter how large $n$ is, only one communication can occur at a time, thus again $\mathrm{O}(1) \cdot^{5}$

Again, we are interested only in " $\mathrm{O}(\mathrm{)}$ " measures, because we are only interested in growth rates as the system size n grows. For instance, if the system size doubles, the cost of a crossbar will quadruple; the $O\left(n^{2}\right)$ cost measure tells us this, with any multiplicative constant being irrelevant.

For Omega networks, it is clear that $\log _{2} n$ network rows are needed, hence the latency value given. Also, each row will have $\mathrm{n} / 2$ switches, so the number of network nodes will be $\mathrm{O}\left(\mathrm{n} \log _{2} n\right)$. This

[^12]figure then gives the cost (in terms of switches, the main expense here). It also gives the bandwidth, since the maximum number of simultaneous transmissions will occur when all switches are sending at once.

Similar considerations hold for the crossbar case.
The crossbar's big advantage is that it is guaranteed that n packets can be sent simultaneously, providing they are to distinct destinations.

That is not true for Omega-networks. If for example, PE0 wants to send to PE3, and at the same time PE4 wishes to sent to PE2, the two packets will clash at the leftmost node of stage 1, where the packet from PE0 will get priority.

On the other hand, a crossbar is very expensive, and thus is dismissed out of hand in most modern systems. Note, though, that an equally troublesom aspect of crossbars is their high latency value; this is a big drawback when the system is not heavily loaded.

The bottom line is that Omega-networks amount to a compromise between buses and crossbars, and for this reason have become popular.

### 3.3.5 Why Have Memory in Modules?

In the shared-memory case, the Ms collectively form the entire shared address space, but with the addresses being assigned to the Ms in one of two ways:

- (a)

High-order interleaving. Here consecutive addresses are in the same M (except at boundaries). For example, suppose for simplicity that our memory consists of addresses 0 through 1023, and that there are four Ms. Then M0 would contain addresses $0-255$, M1 would have 256-511, M2 would have 512-767, and M3 would have 768-1023.

- (b)

Low-order interleaving. Here consecutive addresses are in consecutive M's (except when we get to the right end). In the example above, if we used low-order interleaving, then address 0 would be in M0, 1 would be in M1, 2 would be in M2, 3 would be in M3, 4 would be back in M0, 5 in M1, and so on.

The idea is to have several modules busy at once, say in conjunction with a split-transaction bus. Here, after a processor makes a memory request, it relinquishes the bus, allowing others to use it while the memory does the requested work. Without splitting the memory into modules, this wouldn't achieve parallelism. The bus does need extra lines to identify which processor made the request.

### 3.4 Synchronization Hardware

Avoidance of race conditions, e.g. implementation of locks, plays such a crucial role in sharedmemory parallel processing that hardware assistance is a virtual necessity. Recall, for instance, that critical sections can effectively serialize a parallel program. Thus efficient implementation is crucial.

### 3.4.1 Test-and-Set Instructions

Consider a bus-based system. In addition to whatever memory read and memory write instructions the processor included, there would also be a TAS instruction ${ }^{6}$ This instruction would control a TAS pin on the processor chip, and the pin in turn would be connected to a TAS line on the bus.

Applied to a location L in memory and a register R, say, TAS does the following:

```
copy L to R
if R is O then write 1 to L
```

And most importantly, these operations are done in an atomic manner; no bus transactions by other processors may occur between the two steps.

The TAS operation is applied to variables used as locks. Let's say that 1 means locked and 0 unlocked. Then the guarding of a critical section C by a lock variable L , using a register R , would be done by having the following code in the program being run:

```
TRY: TAS R,L
        JNZ TRY
    C: ... ; start of critical section
    ... ; end of critical section
    MOV 0,L ; unlock
```

where of course JNZ is a jump-if-nonzero instruction, and we are assuming that the copying from the Memory Data Register to R results in the processor N and Z flags (condition codes) being affected.

### 3.4.1.1 LOCK Prefix on Intel Processors

On Pentium machines, the LOCK prefix can be used to get atomicity for certain instructions: ADD, ADC, AND, BTC, BTR, BTS, CMPXCHG, DEC, INC, NEG, NOT, OR, SBB, SUB, XOR,

[^13]XADD. The bus will be locked for the duration of the execution of the instruction, thus setting up atomic operations. There is a special LOCK line in the control bus for this purpose. (Locking thus only applies to these instructions in forms in which there is an operand in memory.) By the way, XCHG asserts this LOCK\# bus signal even if the LOCK prefix is not specified.

For example, consider our count-the-2s example on page ??. If we store mycount in a register, say EDX, then

```
lock add %edx, overallcount
```

would replace the code we had earlier,

```
pthread_mutex_lock(&nextbaselock);
overallcount += mylocalcount;
pthread_mutex_unlock(&nextbaselock);
```

without locks!
Here is how we could implement a lock if needed. The lock would be in a variable named, say, lockvar.

```
    movl $lockvar, %ebx
    movl $1, %ecx
top:
    movl $0, %eax
    lock cmpxchg (%ebx), %ecx
    jz top # else leave the loop and enter the critical section
```

The operation CMPXCHG has EAX as an unnamed operand. The instruction basically does (here source is ECX and destination is lockvar)

```
if c(EAX) != c(destination) # sorry, lock is already locked
    c(EAX) <- c(destination)
    ZF <- 0 # the Zero Flag in the EFLAGS register
else
    c(destination) <- c(source) # lock the lock
    ZF <- 1
```

The LOCK prefix locks the bus for the entire duration of the instruction. Note that the ADD instruction here involves two memory transactions - one to read the old value of overallcount, and the second the write the new, incremented value back to overallcount. So, we are locking for a rather long time, potentially compromising performance when other threads want to access memory, but the benefits can be huge.

### 3.4.1.2 Locks with More Complex Interconnects

In crossbar or $\Omega$-network systems, some 2-bit field in the packet must be devoted to transaction type, say 00 for Read, 01 for Write and 10 for TAS. In a sytem with 16 CPUs and 16 memory modules, say, the packet might consist of 4 bits for the CPU number, 4 bits for the memory module number, 2 bits for the transaction type, and 32 bits for the data (for a write, this is the data to be written, while for a read, it would be the requested value, on the trip back from the memory to the CPU).

But note that the atomicity here is best done at the memory, i.e. some hardware should be added at the memory so that TAS can be done; otherwise, an entire processor-to-memory path (e.g. the bus in a bus-based system) would have to be locked up for a fairly long time, obstructing even the packets which go to other memory modules.

### 3.4.2 May Not Need the Latest

Note carefully that in many settings it may not be crucial to get the most up-to-date value of a variable. For example, a program may have a data structure showing work to be done. Some processors occasionally add work to the queue, and others take work from the queue. Suppose the queue is currently empty, and a processor adds a task to the queue, just as another processor is checking the queue for work. As will be seen later, it is possible that even though the first processor has written to the queue, the new value won't be visible to other processors for some time. But the point is that if the second processor does not see work in the queue (even though the first processor has put it there), the program will still work correctly, albeit with some performance loss.

### 3.4.3 Fetch-and-Add Instructions

Another form of interprocessor synchronization is a fetch-and-add (FA) instruction. The idea of FA is as follows. For the sake of simplicity, consider code like

LOCK (K) ;
$\mathrm{Y}=\mathrm{X}++$;
UNLOCK (K) ;

Suppose our architecture's instruction set included an F\&A instruction. It would add 1 to the specified location in memory, and return the old value (to $\mathbf{Y}$ ) that had been in that location before being incremented. And all this would be an atomic operation.

We would then replace the code above by a library call, say,

FETCH_AND_ADD (X,1);

The C code above would compile to, say,

F\&A X,R,1
where $\mathbf{R}$ is the register into which the old (pre-incrementing) value of $\mathbf{X}$ would be returned.
There would be hardware adders placed at each memory module. That means that the whole operation could be done in one round trip to memory. Without F\&A, we would need two round trips to memory just for the

X++;
(we would load $\mathbf{X}$ into a register in the CPU, increment the register, and then write it back to $\mathbf{X}$ in memory), and then the LOCK () and UNLOCK () would need trips to memory too. This could be a huge time savings, especially for long-latency interconnects.

### 3.5 Cache Issues

If you need a review of cache memories or don't have background in that area at all, read Section A.2.1 in the appendix of this book before continuing.

### 3.5.1 Cache Coherency

Consider, for example, a bus-based system. Relying purely on TAS for interprocessor synchronization would be unthinkable: As each processor contending for a lock variable spins in the loop shown above, it is adding tremendously to bus traffic.

An answer is to have caches at each processor ${ }^{7}$ These will to store copies of the values of lock variables. (Of course, non-lock variables are stored too. However, the discussion here will focus on effects on lock variables.) The point is this: Why keep looking at a lock variable L again and again, using up the bus bandwidth? L may not change value for a while, so why not keep a copy in the cache, avoiding use of the bus?

The answer of course is that eventually L will change value, and this causes some delicate problems. Say for example that processor P5 wishes to enter a critical section guarded by L, and that processor P2 is already in there. During the time P2 is in the critical section, P5 will spin around, always getting the same value for L (1) from C5, P5's cache. When P2 leaves the critical section, P2 will

[^14]set L to 0 - and now C5's copy of L will be incorrect. This is the cache coherency problem, inconsistency between caches.

A number of solutions have been devised for this problem. For bus-based systems, snoopy protocols of various kinds are used, with the word "snoopy" referring to the fact that all the caches monitor ("snoop on") the bus, watching for transactions made by other caches.

The most common protocols are the invalidate and update types. This relation between these two is somewhat analogous to the relation between write-back and write-through protocols for caches in uniprocessor systems:

- Under an invalidate protocol, when a processor writes to a variable in a cache, it first (i.e. before actually doing the write) tells each other cache to mark as invalid its cache line (if any) which contains a copy of the variable $]^{8}$ Those caches will be updated only later, the next time their processors need to access this cache line.
- For an update protocol, the processor which writes to the variable tells all other caches to immediately update their cache lines containing copies of that variable with the new value.

Let's look at an outline of how one implementation (many variations exist) of an invalidate protocol would operate:

In the scenario outlined above, when P2 leaves the critical section, it will write the new value 0 to L. Under the invalidate protocol, P2 will post an invalidation message on the bus. All the other caches will notice, as they have been monitoring the bus. They then mark their cached copies of the line containing L as invalid.

Now, the next time P5 executes the TAS instruction-which will be very soon, since it is in the loop shown above - P 5 will find that the copy of L in C 5 is invalid. It will respond to this cache miss by going to the bus, and requesting P2 to supply the "real" (and valid) copy of the line containing L.

But there's more. Suppose that all this time P6 had also been executing the loop shown above, along with P5. Then P5 and P6 may have to contend with each other. Say P6 manages to grab possession of the bus first ${ }^{9} \mathrm{P} 6$ then executes the TAS again, which finds $\mathrm{L}=0$ and changes L back to 1. P6 then relinquishes the bus, and enters the critical section. Note that in changing L to 1, P6 also sends an invalidate signal to all the other caches. So, when P5 tries its execution of the TAS again, it will have to ask P6 to send a valid copy of the block. P6 does so, but L will be 1, so P5 must resume executing the loop. P5 will then continue to use its valid local copy of L each

[^15]time it does the TAS, until P6 leaves the critical section, writes 0 to L , and causes another cache miss at P5, etc.

At first the update approach seems obviously superior, and actually, if our shared, cacheable $\varepsilon^{10}$ variables were only lock variables, this might be true.

But consider a shared, cacheable vector. Suppose the vector fits into one block, and that we write to each vector element sequentially. Under an update policy, we would have to send a new message on the bus/network for each component, while under an invalidate policy, only one message (for the first component) would be needed. If during this time the other processors do not need to access this vector, all those update messages, and the bus/network bandwidth they use, would be wasted.

Or suppose for example we have code like

Sum += X[I];
in the middle of a for loop. Under an update protocol, we would have to write the value of Sum back many times, even though the other processors may only be interested in the final value when the loop ends. (This would be true, for instance, if the code above were part of a critical section.)

Thus the invalidate protocol works well for some kinds of code, while update works better for others. The CPU designers must try to anticipate which protocol will work well across a broad mix of applications ${ }^{11}$

Now, how is cache coherency handled in non-bus shared-memory systems, say crossbars? Here the problem is more complex. Think back to the bus case for a minute: The very feature which was the biggest negative feature of bus systems - the fact that there was only one path between components made bandwidth very limited-is a very positive feature in terms of cache coherency, because it makes broadcast very easy: Since everyone is attached to that single pathway, sending a message to all of them costs no more than sending it to just one - we get the others for free. That's no longer the case for multipath systems. In such systems, extra copies of the message must be created for each path, adding to overall traffic.

A solution is to send messages only to "interested parties." In directory-based protocols, a list is kept of all caches which currently have valid copies of all blocks. In one common implementation, for example, while P2 is in the critical section above, it would be the owner of the block containing L . (Whoever is the latest node to write to L would be considered its current owner.) It would maintain a directory of all caches having valid copies of that block, say C5 and C6 in our story here. As soon as P2 wrote to L , it would then send either invalidate or update packets (depending on which type was being used) to C5 and C6 (and not to other caches which didn't have valid copies).

[^16]There would also be a directory at the memory, listing the current owners of all blocks. Say for example P0 now wishes to "join the club," i.e. tries to access L, but does not have a copy of that block in its cache C 0 . C0 will thus not be listed in the directory for this block. So, now when it tries to access L and it will get a cache miss. P0 must now consult the home of L, say P14. The home might be determined by L's location in main memory according to high-order interleaving; it is the place where the main-memory version of L resides. A table at P 14 will inform P 0 that P 2 is the current owner of that block. P0 will then send a message to P 2 to add C 0 to the list of caches having valid copies of that block. Similarly, a cache might "resign" from the club, due to that cache line being replaced, e.g. in a LRU setting, when some other cache miss occurs.

### 3.5.2 Example: the MESI Cache Coherency Protocol

Many types of cache coherency protocols have been proposed and used, some of them quite complex. A relatively simple one for snoopy bus systems which is widely used is MESI, which for example is the protocol used in the Pentium series.

MESI is an invalidate protocol for bus-based systems. Its name stands for the four states a given cache line can be in for a given CPU :

- Modified
- Exclusive
- Shared
- Invalid

Note that each memory block has such a state at each cache. For instance, block 88 may be in state S at P5's and P12's caches but in state I at P1's cache.

Here is a summary of the meanings of the states:

| state | meaning |
| :---: | :---: |
| M | written to more than once; no other copy valid |
| E | valid; no other cache copy valid; memory copy valid |
| S | valid; at least one other cache copy valid |
| I | invalid (block either not in the cache or present but incorrect) |

Following is a summary of MESI state changes ${ }^{12}$ When reading it, keep in mind again that there is a separate state for each cache/memory block combination.

[^17]In addition to the terms read hit, read miss, write hit, write miss, which you are already familiar with, there are also read snoop and write snoop. These refer to the case in which our CPU observes on the bus a block request by another CPU that has attempted a read or write action but encountered a miss in its own cache; if our cache has a valid copy of that block, we must provide it to the requesting CPU (and in some cases to memory).

So, here are various events and their corresponding state changes:
If our CPU does a read:

| present state | event | new state |
| :---: | :---: | :---: |
| M | read hit | M |
| E | read hit | E |
| S | read hit | S |
| I | read miss; no valid cache copy at any other CPU | E |
| I | read miss; at least one valid cache copy in some other CPU | S |

## If our CPU does a memory write:

| present state | event | new state |
| :---: | :---: | :---: |
| M | write hit; do not put invalidate signal on bus; do not update memory | M |
| E | same as M above | M |
| S | write hit; put invalidate signal on bus; update memory | E |
| I | write miss; update memory but do nothing else | I |

## If our CPU does a read or write snoop:

| present state | event | newstate |
| :---: | :---: | :---: |
| M | read snoop; write line back to memory, picked up by other CPU | S |
| M | write snoop; write line back to memory, signal other CPU now OK to do its write | I |
| E | read snoop; put shared signal on bus; no memory action | S |
| E | write snoop; no memory action | I |
| S | read snoop | S |
| S | write snoop | I |
| I | any snoop | I |

Note that a write miss does NOT result in the associated block being brought in from memory.
Example: Suppose a given memory block has state $M$ at processor A but has state I at processor B , and B attempts to write to the block. B will see that its copy of the block is invalid, so it notifies the other CPUs via the bus that it intends to do this write. CPU A sees this announcement, tells $B$ to wait, writes its own copy of the block back to memory, and then tells $B$ to go ahead with its write. The latter action means that A's copy of the block is not correct anymore, so the block now
has state I at A. B's action does not cause loading of that block from memory to its cache, so the block still has state I at B.

### 3.5.3 The Problem of "False Sharing"

Consider the C declaration
int W,Z;

Since $\mathbf{W}$ and $\mathbf{Z}$ are declared adjacently, most compilers will assign them contiguous memory addresses. Thus, unless one of them is at a memory block boundary, when they are cached they will be stored in the same cache line. Suppose the program writes to $\mathbf{Z}$, and our system uses an invalidate protocol. Then $\mathbf{W}$ will be considered invalid at the other processors, even though its values at those processors' caches are correct. This is the false sharing problem, alluding to the fact that the two variables are sharing a cache line even though they are not related.

This can have very adverse impacts on performance. If for instance our variable $\mathbf{W}$ is now written to, then $\mathbf{Z}$ will suffer unfairly, as its copy in the cache will be considered invalid even though it is perfectly valid. This can lead to a "ping-pong" effect, in which alternate writing to two variables leads to a cyclic pattern of coherency transactions.

One possible solution is to add padding, e.g. declaring $\mathbf{W}$ and $\mathbf{Z}$ like this:
int $Q, U[1000], Z ;$
to separate $\mathbf{Q}$ and $\mathbf{Z}$ so that they won't be in the same cache block. Of course, we must take block size into account, and check whether the compiler really has placed the two variables are in widely separated locations. To do this, we could for instance run the code

```
printf("%x %x\n,&Q,&Z);
```


### 3.6 Memory-Access Consistency Policies

Though the word consistency in the title of this section may seem to simply be a synonym for coherency from the last section, and though there actually is some relation, the issues here are quite different. In this case, it is a timing issue: After one processor changes the value of a shared variable, when will that value be visible to the other processors?

There are various reasons why this is an issue. For example, many processors, especially in multiprocessor systems, have write buffers, which save up writes for some time before actually sending them to memory. (For the time being, let's suppose there are no caches.) The goal is to reduce memory access costs. Sending data to memory in groups is generally faster than sending one at a time, as the overhead of, for instance, acquiring the bus is amortized over many accesses. Reads following a write may proceed, without waiting for the write to get to memory, except for reads to the same address. So in a multiprocessor system in which the processors use write buffers, there will often be some delay before a write actually shows up in memory.

A related issue is that operations may occur, or appear to occur, out of order. As noted above, a read which follows a write in the program may execute before the write is sent to memory. Also, in a multiprocessor system with multiple paths between processors and memory modules, two writes might take different paths, one longer than the other, and arrive "out of order." In order to simplify the presentation here, we will focus on the case in which the problem is due to write buffers, though.

The designer of a multiprocessor system must adopt some consistency model regarding situations like this. The above discussion shows that the programmer must be made aware of the model, or risk getting incorrect results. Note also that different consistency models will give different levels of performance. The "weaker" consistency models make for faster machines but require the programmer to do more work.

The strongest consistency model is Sequential Consistency. It essentially requires that memory operations done by one processor are observed by the other processors to occur in the same order as executed on the first processor. Enforcement of this requirement makes a system slow, and it has been replaced on most systems by weaker models.

One such model is release consistency. Here the processors' instruction sets include instructions ACQUIRE and RELEASE. Execution of an ACQUIRE instruction at one processor involves telling all other processors to flush their write buffers. However, the ACQUIRE won't execute until pending RELEASEs are done. Execution of a RELEASE basically means that you are saying, "I'm done writing for the moment, and wish to allow other processors to see what I've written." An ACQUIRE waits for all pending RELEASEs to complete before it executes ${ }^{13}$

A related model is scope consistency. Say a variable, say Sum, is written to within a critical section guarded by LOCK and UNLOCK instructions. Then under scope consistency any changes made by one processor to Sum within this critical section would then be visible to another processor when the latter next enters this critical section. The point is that memory update is postponed until it is actually needed. Also, a barrier operation (again, executed at the hardware level) forces all pending memory writes to complete.

All modern processors include instructions which implement consistency operations. For example,

[^18]Sun Microsystems' SPARC has a MEMBAR instruction. If used with a STORE operand, then all pending writes at this processor will be sent to memory. If used with the LOAD operand, all writes will be made visible to this processor.

Now, how does cache coherency fit into all this? There are many different setups, but for example let's consider a design in which there is a write buffer between each processor and its cache. As the processor does more and more writes, the processor saves them up in the write buffer. Eventually, some programmer-induced event, e.g. a MEMBAR instruction. ${ }^{14}$ will cause the buffer to be flushed. Then the writes will be sent to "memory" - actually meaning that they go to the cache, and then possibly to memory.

The point is that (in this type of setup) before that flush of the write buffer occurs, the cache coherency system is quite unaware of these writes. Thus the cache coherency operations, e.g. the various actions in the MESI protocol, won't occur until the flush happens.

To make this notion concrete, again consider the example with Sum above, and assume release or scope consistency. The CPU currently executing that code (say CPU 5) writes to Sum, which is a memory operation-it affects the cache and thus eventually the main memory-but that operation will be invisible to the cache coherency protocol for now, as it will only be reflected in this processor's write buffer. But when the unlock is finally done (or a barrier is reached), the write buffer is flushed and the writes are sent to this CPU's cache. That then triggers the cache coherency operation (depending on the state). The point is that the cache coherency operation would occur only now, not before.

What about reads? Suppose another processor, say CPU 8, does a read of Sum, and that page is marked invalid at that processor. A cache coherency operation will then occur. Again, it will depend on the type of coherency policy and the current state, but in typical systems this would result in Sum's cache block being shipped to CPU 8 from whichever processor the cache coherency system thinks has a valid copy of the block. That processor may or may not be CPU 5, but even if it is, that block won't show the recent change made by CPU 5 to Sum.

The analysis above assumed that there is a write buffer between each processor and its cache. There would be a similar analysis if there were a write buffer between each cache and memory.

Note once again the performance issues. Instructions such as ACQUIRE or MEMBAR will use a substantial amount of interprocessor communication bandwidth. A consistency model must be chosen carefully by the system designer, and the programmer must keep the communication costs in mind in developing the software.

The recent Pentium models use Sequential Consistency, with any write done by a processor being immediately sent to its cache as well.

[^19]
### 3.7 Fetch-and-Add Combining within Interconnects

In addition to read and write operations being specifiable in a network packet, an F\&A operation could be specified as well (a 2-bit field in the packet would code which operation was desired). Again, there would be adders included at the memory modules, i.e. the addition would be done at the memory end, not at the processors. When the F\&A packet arrived at a memory module, our variable $\mathbf{X}$ would have 1 added to it, while the old value would be sent back in the return packet (and put into R).

Another possibility for speedup occurs if our system uses a multistage interconnection network such as a crossbar. In that situation, we can design some intelligence into the network nodes to do packet combining: Say more than one CPU is executing an F\&A operation at about the same time for the same variable $\mathbf{X}$. Then more than one of the corresponding packets may arrive at the same network node at about the same time. If each one requested an incrementing of $\mathbf{X}$ by 1 , the node can replace the two packets by one, with an increment of 2 . Of course, this is a delicate operation, and we must make sure that different CPUs get different return values, etc.

### 3.8 Multicore Chips

A recent trend has been to put several CPUs on one chip, termed a multicore chip. As of March 2008, dual-core chips are common in personal computers, and quad-core machines are within reach of the budgets of many people. Just as the invention of the integrated circuit revolutionized the computer industry by making computers affordable for the average person, multicore chips will undoubtedly revolutionize the world of parallel programming.

A typical dual-core setup might have the two CPUs sharing a common L2 cache, with each CPU having its own L3 cache. The chip may interface to the bus or interconnect network of via an L1 cache.

Multicore is extremely important these days. However, they are just SMPs, for the most part, and thus should not be treated differently.

### 3.9 Optimal Number of Threads

A common question involves the best number of threads to run in a shared-memory setting. Clearly there is no general magic answer, but here are some considerations ${ }^{15}$

[^20]- If your application does a lot of I/O, CPUs or cores may stay idle while waiting for I/O events. It thus makes to have many threads, so that computation threads can run when the I/O threads are tied up.
- In a purely computational application, one generally should not have more threads than cores. However, a program with a lot of virtual memory page faults may benefit from setting up extra threads, as page replacement involves (disk) I/O.
- Applications in which there is heavy interthread communication, say due to having a lot of lock variable, access, may benefit from setting up fewer threads than the number of cores.
- Many Intel processors include hardware for hypertheading. These are not full threads in the sense of having separate cores, but rather involve a limited amount of resource duplication within a core. The performance gain from this is typically quite modest. In any case, be aware of it; some software systems count these as threads, and assume for instance that there are 8 cores when the machine is actually just quad core.
- With GPUs (Chapter 5), most memory accesses have long latency and thus are I/O-like. Typically one needs very large numbers of threads for good performance.


### 3.10 Processor Affinity

With a timesharing OS, a given thread may run on different cores during different timeslices. If so, the cache for a given core may need a lot of refreshing, each time a new thread runs on that core. To avoid this slowdown, one might designate a preferred core for each thread, in the hope of reusing cache contents. Setting this up is dependent on the chip and the OS. OpenMP 3.1 has some facility for this.

### 3.11 Illusion of Shared-Memory through Software

### 3.11.0.1 Software Distributed Shared Memory

There are also various shared-memory software packages that run on message-passing hardware such as NOWs, called software distributed shared memory (SDSM) systems. Since the platforms do not have any physically shared memory, the shared-memory view which the programmer has is just an illusion. But that illusion is very useful, since the shared-memory paradigm is believed to be the easier one to program in. Thus SDSM allows us to have "the best of both worlds" - the convenience of the shared-memory world view with the inexpensive cost of some of the messagepassing hardware systems, particularly networks of workstations (NOWs).

SDSM itself is divided into two main approaches, the page-based and object-based varieties. The page-based approach is generally considered clearer and easier to program in, and provides the programmer the "look and feel" of shared-memory programming better than does the object-based type ${ }^{16}$ We will discuss only the page-based approach here. The most popular SDSM system today is the page-based Treadmarks (Rice University). Another excellent page-based system is JIAJIA (Academy of Sciences, China).

To illustrate how page-paged SDSMs work, consider the line of JIAJIA code

Prime $=($ int $*)$ jia_alloc(N*sizeof(int));

The function jia_alloc() is part of the JIAJIA library, libjia.a, which is linked to one's application program during compilation.

At first this looks a little like a call to the standard malloc() function, setting up an array Prime of size $\mathbf{N}$. In fact, it does indeed allocate some memory. Note that each node in our JIAJIA group is executing this statement, so each node allocates some memory at that node. Behind the scenes, not visible to the programmer, each node will then have its own copy of Prime.

However, JIAJIA sets things up so that when one node later accesses this memory, for instance in the statement

Prime [I] = 1;
this action will eventually trigger a network transaction (not visible to the programmer) to the other JIAJIA nodes ${ }^{17}$ This transaction will then update the copies of Prime at the other nodes ${ }^{18}$

How is all of this accomplished? It turns out that it relies on a clever usage of the nodes' virtual memory (VM) systems. To understand this, you need a basic knowledge of how VM systems work. If you lack this, or need review, read Section A.2.2 in the appendix of this book before continuing.

Here is how VM is exploited to develop SDSMs on Unix systems. The SDSM will call a system function such as mprotect(). This allows the SDSM to deliberately mark a page as nonresident (even if the page is resident). Basically, anytime the SDSM knows that a node's local copy of a variable is invalid, it will mark the page containing that variable as nonresident. Then, the next time the program at this node tries to access that variable, a page fault will occur.

As mentioned in the review above, normally a page fault causes a jump to the OS. However, technically any page fault in Unix is handled as a signal, specifically SIGSEGV. Recall that Unix allows the programmer to write his/her own signal handler for any signal type. In this case, that

[^21]means that the programmer-meaning the people who developed JIAJIA or any other page-based SDSM-writes his/her own page fault handler, which will do the necessary network transactions to obtain the latest valid value for $\mathbf{X}$.

Note that although SDSMs are able to create an illusion of almost all aspects of shared memory, it really is not possible to create the illusion of shared pointer variables. For example on shared memory hardware we might have a variable like $\mathbf{P}$ :

```
int Y,*P;
*
P = &Y;
```

There is no simple way to have a variable like $\mathbf{P}$ in an SDSM. This is because a pointer is an address, and each node in an SDSM has its own memory separate address space. The problem is that even though the underlying SDSM system will keep the various copies of $\mathbf{Y}$ at the different nodes consistent with each other, $\mathbf{Y}$ will be at a potentially different address on each node.

All SDSM systems must deal with a software analog of the cache coherency problem. Whenever one node modifies the value of a shared variable, that node must notify the other nodes that a change has been made. The designer of the system must choose between update or invalidate protocols, just as in the hardware case ${ }^{19}$ Recall that in non-bus-based shared-memory multiprocessors, one needs to maintain a directory which indicates at which processor a valid copy of a shared variable exists. Again, SDSMs must take an approach similar to this.

Similarly, each SDSM system must decide between sequential consistency, release consistency etc. More on this later.

Note that in the NOW context the internode communication at the SDSM level is typically done by TCP/IP network actions. Treadmarks uses UDP, which is faster than TCP. but still part of the slow TCP/IP protocol suite. TCP/IP was simply not designed for this kind of work. Accordingly, there have been many efforts to use more efficient network hardware and software. The most popular of these is the Virtual Interface Architecture (VIA).

Not only are coherency actions more expensive in the NOW SDSM case than in the shared-memory hardware case due to network slowness, there is also expense due to granularity. In the hardware case we are dealing with cache blocks, with a typical size being 512 bytes. In the SDSM case, we are dealing with pages, with a typical size being 4096 bytes. The overhead for a cache coherency transaction can thus be large.

[^22]
### 3.11.0.2 Case Study: JIAJIA

## Programmer Interface

We will not go into detail on JIAJIA programming here. There is a short tutorial on JIAJIA at http://heather.cs.ucdavis.edu/~matloff/jiajia.html, but here is an overview:

- One writes in C/C++ (or FORTRAN), making calls to the JIAJIA library, which is linked in upon compilation.
- The library calls include standard shared-memory operations for lock, unlock, barrier, processor number, etc., plus some calls aimed at improving performance.

Following is a JIAJIA example program, performing Odd/Even Transposition Sort. This is a variant on Bubble Sort, sometimes useful in parallel processing contexts. ${ }^{20}$ The algorithm consists of n phases, in which each processor alternates between trading with its left and right neighbors.

```
// JIAJIA example program: Odd-Even Tranposition Sort
// array is of size n, and we use n processors; this would be more
// efficient in a "chunked" versions, of course (and more suited for a
// message-passing context anyway)
#include <stdio.h>
#include <stdlib.h>
#include <jia.h> // required include; also must link via -ljia
// pointer to shared variable
int *x; // array to be sorted
int n, // range to check for primeness
    debug; // 1 for debugging, O else
// if first arg is bigger, then replace it by the second
void cpsmaller(int *p1,int *p2)
{ int tmp;
}
// if first arg is smaller, then replace it by the second
void cpbigger(int *p1,int *p2)
{ int tmp;
if (*p1 < *p2) *p1 = *p2;
}
// does sort of m-element array y
void oddeven(int *y, int m)
{ int i,left=jiapid-1,right=jiapid+1,newval;
```

[^23]```
    for (i=0; i < m; i++) {
        if ((i+jiapid)%2 == 0) {
            if (right < m)
                if (y[jiapid] > y[right]) newval = y[right];
    }
    else {
            if (left >= 0)
                if (y[jiapid] < y[left]) newval = y[left];
    }
    jia_barrier();
    if ((i+jiapid)%2 == 0 && right < m || (i+jiapid)%2 == 1 && left >= 0)
                y[jiapid] = newval;
    jia_barrier();
    }
}
main(int argc, char **argv)
{ int i,mywait=0;
    jia_init(argc,argv); // required init call
    // get command-line arguments (shifted for nodes > 0)
    if (jiapid == 0) {
        n = atoi(argv[1]);
        debug = atoi(argv[2]);
    }
    else {
        n = atoi(argv[2]);
        debug = atoi(argv[3]);
    }
    jia_barrier();
    // create a shared array x of length n
    x = (int *) jia_alloc(n*sizeof(int));
    // barrier recommended after allocation
    jia_barrier();
    // node O gets simple test array from command-line
    if (jiapid == 0) {
        for (i = 0; i < n; i++)
            x[i] = atoi(argv[i+3]);
    }
    jia_barrier();
    if (debug && jiapid == 0)
        while (mywait == 0) { ; }
    jia_barrier();
    oddeven(x,n);
    if (jiapid == 0) {
        printf("\nfinal array\n");
        for (i = 0; i < n; i++)
            printf("%d\n",x[i]);
    }
    jia_exit();
}
```


## System Workings

JIAJIA's main characteristics as an SDSM are:

- page-based
- scope consistency
- home-based
- multiple writers

Let's take a look at these.
As mentioned earlier, one first calls jia_alloc() to set up one's shared variables. Note that this will occur at each node, so there are multiple copies of each variable; the JIAJIA system ensures that these copies are consistent with each other, though of course subject to the laxity afforded by scope consistency.

Recall that under scope consistency, a change made to a shared variable at one processor is guaranteed to be made visible to another processor if the first processor made the change between lock/unlock operations and the second processor accesses that variable between lock/unlock operations on that same lock ${ }^{211}$

Each page - and thus each shared variable - has a home processor. If another processor writes to a page, then later when it reaches the unlock operation it must send all changes it made to the page back to the home node. In other words, the second processor calls jia_unlock(), which sends the changes to its sister invocation of jia_unlock() at the home processor ${ }^{22}$ Say later a third processor calls jia_lock() on that same lock, and then attempts to read a variable in that page. A page fault will occur at that processor, resulting in the JIAJIA system running, which will then obtain that page from the first processor.

Note that all this means the JIAJIA system at each processor must maintain a page table, listing where each home page resides ${ }^{233}$ At each processor, each page has one of three states: Invalid, Read-Only, Read-Write. State changes, though, are reported when lock/unlock operations occur. For example, if CPU 5 writes to a given page which had been in Read-Write state at CPU 8, the latter will not hear about CPU 5's action until some CPU does a lock. This CPU need not be CPI 8. When one CPU does a lock, it must coordinate with all other nodes, at which time state-change messages will be piggybacked onto lock-coordination messages.

[^24]Note also that JIAJIA allows the programmer to specify which node should serve as the home of a variable, via one of several forms of the jia_alloc() call. The programmer can then tailor his/her code accordingly. For example, in a matrix problem, the programmer may arrange for certain rows to be stored at a given node, and then write the code so that most writes to those rows are done by that processor.

The general principle here is that writes performed at one node can be made visible at other nodes on a "need to know" basis. If for instance in the above example with CPUs 5 and 8 , CPU 2 does not access this page, it would be wasteful to send the writes to CPU 2, or for that matter to even inform CPU 2 that the page had been written to. This is basically the idea of all nonSequential consistency protocols, even though they differ in approach and in performance for a given application.

JIAJIA allows multiple writers of a page. Suppose CPU 4 and CPU 15 are simultaneously writing to a particular page, and the programmer has relied on a subsequent barrier to make those writes visible to other processors ${ }^{24}$ When the barrier is reached, each will be informed of the writes of the other ${ }^{25}$ Allowing multiple writers helps to reduce the performance penalty due to false sharing.

### 3.12 Barrier Implementation

Recall that a barrier is program cod ${ }^{26}$ which has a processor do a wait-loop action until all processors have reached that point in the program ${ }^{27}$

A function Barrier() is often supplied as a library function; here we will see how to implement such a library function in a correct and efficient manner. Note that since a barrier is a serialization point for the program, efficiency is crucial to performance.

Implementing a barrier in a fully correct manner is actually a bit tricky. We'll see here what can go wrong, and how to make sure it doesn't.

In this section, we will approach things from a shared-memory point of view. But the methods apply in the obvious way to message-passing systems as well, as will be discused later.

[^25]
### 3.12.1 A Use-Once Version

```
struct BarrStruct {
    int NNodes, // number of threads participating in the barrier
        Count, // number of threads that have hit the barrier so far
        pthread_mutex_t Lock = PTHREAD_MUTEX_INITIALIZER;
} ;
Barrier(struct BarrStruct *PB)
{ pthread_mutex_lock(&PB->Lock);
    PB->Count++;
    pthread_mutex_unlock(&PB->Lock);
    while (PB->Count < PB->NNodes) ;
}
```

This is very simple, actually overly so. This implementation will work once, so if a program using it doesn't make two calls to Barrier() it would be fine. But not otherwise. If, say, there is a call to $\operatorname{Barrier}()$ in a loop, we'd be in trouble.

What is the problem? Clearly, something must be done to reset Count to 0 at the end of the call, but doing this safely is not so easy, as seen in the next section.

### 3.12.2 An Attempt to Write a Reusable Version

Consider the following attempt at fixing the code for Barrier():

```
Barrier(struct BarrStruct *PB)
{ int OldCount;
    pthread_mutex_lock(&PB->Lock);
    OldCount = PB->Count++;
    pthread_mutex_unlock(&PB->Lock);
    if (OldCount == PB->NNodes-1) PB->Count = 0;
    while (PB->Count < PB->NNodes) ;
}
```

Unfortunately, this doesn't work either. To see why, consider a loop with a barrier call at the end:

```
struct BarrStruct B; // global variable
while (.......) {
    .........
    Barrier(&B);
}
```

At the end of the first iteration of the loop, all the processors will wait at the barrier until everyone catches up. After this happens, one processor, say 12, will reset B.Count to 0 , as desired. But
if we are unlucky, some other processor, say processor 3, will then race ahead, perform the second iteration of the loop in an extremely short period of time, and then reach the barrier and increment the Count variable before processor 12 resets it to 0 . This would result in disaster, since processor 3 's increment would be canceled, leaving us one short when we try to finish the barrier the second time.

Another disaster scenario which might occur is that one processor might reset B.Count to 0 before another processor had a chance to notice that B.Count had reached B.NNodes.

### 3.12.3 A Correct Version

One way to avoid this would be to have two Count variables, and have the processors alternate using one then the other. In the scenario described above, processor 3 would increment the other Count variable, and thus would not conflict with processor 12's resetting. Here is a safe barrier function based on this idea:

```
struct BarrStruct {
    int NNodes, // number of threads participating in the barrier
        Count[2], // number of threads that have hit the barrier so far
        pthread_mutex_t Lock = PTHREAD_MUTEX_INITIALIZER;
};
Barrier(struct BarrStruct *PB)
{ int Par,OldCount;
    Par = PB->EvenOdd;
    pthread_mutex_lock(&PB->Lock);
    OldCount = PB->Count[Par]++;
    if (OldCount == PB->NNodes-1) {
        PB}->\mathrm{ Count[Par] = 0;
        PB->EvenOdd = 1 - Par;
        pthread_mutex_unlock(&PB->Lock);
    }
    else {
        pthread_mutex_unlock(&PB->Lock);
        while (PB->Count[Par] > 0) ;
    }
}
```


### 3.12.4 Refinements

### 3.12.4.1 Use of Wait Operations

The code

```
else while (PB->Count[Par] > 0) ;
```

is harming performance, since it has the processor spining around doing no useful work. In the Pthreads context, we can use a condition variable:

```
struct BarrStruct {
    int NNodes, // number of threads participating in the barrier
        Count[2], // number of threads that have hit the barrier so far
        pthread_mutex_t Lock = PTHREAD_MUTEX_INITIALIZER;
        pthread_cond_t CV = PTHREAD_COND_INITIALIZER;
} ;
Barrier(struct BarrStruct *PB)
{ int Par,I;
    Par = PB->EvenOdd;
    pthread_mutex_lock(&PB->Lock);
    PB->Count[Par]++;
    if (PB->Count < PB->NNodes)
        pthread_cond_wait(&PB->CV,&PB->Lock);
    else {
        PB->Count[Par] = 0;
        PB->EvenOdd = 1 - Par;
        for (I = 0; I < PB->NNodes-1; I++)
            pthread_cond_signal(&PB->CV);
    }
    pthread_mutex_unlock(&PB->Lock);
}
```

Here, if a thread finds that not everyone has reached the barrier yet, it still waits for the rest, but does so passively, via the wait for the condition variable CV. This way the thread is not wasting valuable time on that processor, which can run other useful work.

Note that the call to pthread_cond_wait() requires use of the lock. Your code must lock the lock before making the call. The call itself immediately unlocks that lock after it registers the wait with the threads manager. But the call blocks until awakened when another thread calls pthread_cond_signal() or pthread_cond_broadcast().

It is required that your code lock the lock before calling pthread_cond_signal(), and that it unlock the lock after the call.

By using pthread_cond_wait() and placing the unlock operation later in the code, as seen above, we actually could get by with just a single Count variable, as before.

Even better, the for loop could be replaced by a single call
pthread_cond_broadcast(\&PB->PB->CV) ;

This still wakes up the waiting threads one by one, but in a much more efficient way, and it makes for clearer code.

### 3.12.4.2 Parallelizing the Barrier Operation

3.12.4.2 1 Tree Barriers It is clear from the code above that barriers can be costly to performance, since they rely so heavily on critical sections, i.e. serial parts of a program. Thus in many settings it is worthwhile to parallelize not only the general computation, but also the barrier operations themselves.

Consider for instance a barrier in which 16 threads are participating. We could speed things up by breaking this barrier down into two sub-barriers, with eight threads each. We would then set up three barrier operations: one of the first group of eight threads, another for the other group of eight threads, and a third consisting of a "competition" between the two groups. The variable NNodes above would have the value 8 for the first two barriers, and would be equal to 2 for the third barrier.

Here thread 0 could be the representative for the first group, with thread 4 representing the second group. After both groups's barriers were hit by all of their members, threads 0 and 4 would participated in the third barrier.

Note that then the notification phase would the be done in reverse: When the third barrier was complete, threads 0 and 4 would notify the members of their groups.

This would parallelize things somewhat, as critical-section operations could be executing simultaneously for the first two barriers. There would still be quite a bit of serial action, though, so we may wish to do further splitting, by partitioning each group of four threads into two subroups of two threads each.

In general, for $n$ threads (with $n$, say, equal to a power of 2 ) we would have a tree structure, with $\log _{2} n$ levels in the tree. The $i^{\text {th }}$ level (starting with the root as level 0 ) with consist of $2^{i}$ parallel barriers, each one representing $n / 2^{i}$ threads.
3.12.4.2.2 Butterfly Barriers Another method basically consists of each node "shaking hands" with every other node. In the shared-memory case, handshaking could be done by having a global array ReachedBarrier. When thread 3 and thread 7 shake hands, for instance, would reach the barrier, thread 3 would set ReachedBarrier [3] to 1, and would then wait for ReachedBarrier[7] to become 1. The wait, as before, could either be a while loop or a call to pthread_cond_wait(). Thread 7 would do the opposite.

If we have n nodes, again with n being a power of 2 , then the barrier process would consist of $\log _{2} n$ phases, which we'll call phase 0 , phase 1 , etc. Then the process works as follows.

For any node i , let $\mathrm{i}(\mathrm{k})$ be the number obtained by inverting bit k in the binary representation of i, with bit 0 being the least significant bit. Then in the $k^{t h}$ phase, node i would shake hands with node $\mathrm{i}(\mathrm{k})$.

For example, say $n=8$. In phase 0 , node $5=101_{2}$, say, would shake hands with node $4=100_{2}$. Actually, a butterfly exchange amounts to a number of simultaneously tree operations.

## Chapter 4

## Introduction to OpenMP

OpenMP has become the de facto standard for shared-memory programming.

### 4.1 Overview

OpenMP has become the environment of choice for many, if not most, practitioners of sharedmemory parallel programming. It consists of a set of directives which are added to one's C/C++/FORTRAN code that manipulate threads, without the programmer him/herself having to deal with the threads directly. This way we get "the best of both worlds" - the true parallelism of (nonpreemptive) threads and the pleasure of avoiding the annoyances of threads programming.

Most OpenMP constructs are expressed via pragmas, i.e. directives. The syntax is

```
#pragma omp . .....
```

The number sign must be the first nonblank character in the line.

### 4.2 Example: Dijkstra Shortest-Path Algorithm

The following example, implementing Dijkstra's shortest-path graph algorithm, will be used throughout this tutorial, with various OpenMP constructs being illustrated later by modifying this code:

```
// Dijkstra.c
// OpenMP example program: Dijkstra shortest-path finder in a
```

```
// bidirectional graph; finds the shortest path from vertex O to all
// others
// usage: dijkstra nv print
// where nv is the size of the graph, and print is 1 if graph and min
// distances are to be printed out, O otherwise
#include <omp.h>
// global variables, shared by all threads by default
int nv, // number of vertices
    *notdone, // vertices not checked yet
    nth, // number of threads
    chunk, // number of vertices handled by each thread
    md, // current min over all threads
    mv, // vertex which achieves that min
    largeint = -1; // max possible unsigned int
unsigned *ohd, // 1-hop distances between vertices; "ohd[i][j]" is
        // ohd[i*nv+j]
        *mind; // min distances found so far
void init(int ac, char **av)
{ int i,j,tmp;
    nv = atoi(av[1]);
    ohd = malloc(nv*nv*sizeof(int));
    mind = malloc(nv*sizeof(int));
    notdone = malloc(nv*sizeof(int));
    // random graph
    for (i = 0; i < nv; i++)
        for (j = i; j < nv; j++) {
            if (j == i) ohd[i*nv+i] = 0;
            else {
                ohd[nv*i+j] = rand() % 20;
                ohd[nv*j+i] = ohd[nv*i+j];
            }
        }
    for (i = 1; i < nv; i++) {
        notdone[i] = 1;
        mind[i] = ohd[i];
    }
}
// finds closest to 0 among notdone, among s through e
void findmymin(int s, int e, unsigned *d, int *v)
{ int i;
    *d = largeint;
    for (i = s; i <= e; i++)
        if (notdone[i] && mind[i] < *d) {
            *d = mind[i];
            *v = i;
        }
}
// for each i in [s,e], ask whether a shorter path to i exists, through
// mv
```

```
void updatemind(int s, int e)
{ int i;
    for (i = s; i <= e; i++)
        if (mind[mv] + ohd[mv*nv+i] < mind[i])
        mind[i] = mind[mv] + ohd[mv*nv+i];
}
void dowork()
{
    #pragma omp parallel
    { int startv,endv, // start, end vertices for my thread
            step, // whole procedure goes nv steps
            mymv, // vertex which attains the min value in my chunk
            me = omp_get_thread_num();
            unsigned mymd; // min value found by this thread
        #pragma omp single
        { nth = omp_get_num_threads(); // must call inside parallel block
            if (nv % nth != 0) {
                    printf("nv must be divisible by nth\n");
                    exit(1);
            }
            chunk = nv/nth;
            printf("there are %d threads\n",nth);
        }
        startv = me * chunk;
        endv = startv + chunk - 1;
        for (step = 0; step < nv; step++) {
            // find closest vertex to 0 among notdone; each thread finds
            // closest in its group, then we find overall closest
            #pragma omp single
            { md = largeint; mv = 0; }
            findmymin(startv,endv,&mymd,&mymv);
            // update overall min if mine is smaller
            #pragma omp critical
            { if (mymd < md)
                    { md = mymd; mv = mymv; }
            }
            #pragma omp barrier
            // mark new vertex as done
            #pragma omp single
            { notdone[mv] = 0; }
            // now update my section of mind
            updatemind(startv,endv);
            #pragma omp barrier
        }
    }
}
int main(int argc, char **argv)
{ int i,j,print;
    double startime,endtime;
    init(argc,argv);
    startime = omp_get_wtime();
    // parallel
    dowork();
    // back to single thread
    endtime = omp_get_wtime();
    printf("elapsed time: %f\n",endtime-startime);
```

```
    print = atoi(argv[2]);
    if (print) {
        printf("graph weights:\n");
        for (i = 0; i < nv; i++) {
            for (j = 0; j < nv; j++)
                printf("%u ",ohd[nv*i+j]);
            printf("\n");
        }
        printf("minimum distances:\n");
        for (i = 1; i < nv; i++)
            printf("%u\n",mind[i]);
}
}
```

The constructs will be presented in the following sections, but first the algorithm will be explained.

### 4.2.1 The Algorithm

The code implements the Dijkstra algorithm for finding the shortest paths from vertex 0 to the other vertices in an N-vertex undirected graph. Pseudocode for the algorithm is shown below, with the array G assumed to contain the one-hop distances between vertices.

```
Done = {0} # vertices checked so far
NewDone = None # currently checked vertex
NonDone = {1,2,...,N-1} # vertices not checked yet
for J = 0 to N-1 Dist[J] = G(0,J) # initialize shortest-path lengths
for Step = 1 to N-1
    find J such that Dist[J] is min among all J in NonDone
    transfer J from NonDone to Done
    NewDone = J
    for K = 1 to N-1
        if K is in NonDone
            # check if there is a shorter path from O to K through NewDone
            # than our best so far
            Dist[K] = min(Dist[K],Dist [NewDone] +G[NewDone,K])
```

At each iteration, the algorithm finds the closest vertex J to 0 among all those not yet processed, and then updates the list of minimum distances to each vertex from 0 by considering paths that go through J. Two obvious potential candidate part of the algorithm for parallelization are the "find J " and "for K" lines, and the above OpenMP code takes this approach.

### 4.2.2 The OpenMP parallel Pragma

As can be seen in the comments in the lines

```
// parallel
dowork();
// back to single thread
```

the function main() is run by a master thread, which will then branch off into many threads running dowork() in parallel. The latter feat is accomplished by the directive in the lines

```
void dowork()
{
    #pragma omp parallel
    { int startv,endv, // start, end vertices for this thread
            step, // whole procedure goes nv steps
            mymv, // vertex which attains that value
            me = omp_get_thread_num();
```

That directive sets up a team of threads (which includes the master), all of which execute the block following the directive in parallel. ${ }^{1}$ Note that, unlike the for directive which will be discussed below, the parallel directive leaves it up to the programmer as to how to partition the work. In our case here, we do that by setting the range of vertices which this thread will process:

```
startv = me * chunk;
endv = startv + chunk - 1;
```

Again, keep in mind that all of the threads execute this code, but we've set things up with the variable me so that different threads will work on different vertices. This is due to the OpenMP call

```
me = omp_get_thread_num();
```

which sets me to the thread number for this thread.

### 4.2.3 Scope Issues

Note carefully that in

```
#pragma omp parallel
{ int startv,endv, // start, end vertices for this thread
    step, // whole procedure goes nv steps
    mymv, // vertex which attains that value
    me = omp_get_thread_num();
```

[^26]the pragma comes before the declaration of the local variables. That means that all of them are "local" to each thread, i.e. not shared by them. But if a work sharing directive comes within a function but after declaration of local variables, those variables are actually "global" to the code in the directive, i.e. they are shared in common among the threads.

This is the default, but you can change these properties, e.g. using the private keyword and its cousins. For instance,
\#pragma omp parallel private( $\mathrm{x}, \mathrm{y}$ )
would make $\mathbf{x}$ and $\mathbf{y}$ nonshared even if they were declared above the directive line. You may wish to modify that a bit, so that $\mathbf{x}$ and $\mathbf{y}$ have initial values that were shared before the directive; use firstprivate for this.

It is crucial to keep in mind that variables which are global to the program (in the $\mathrm{C} / \mathrm{C}++$ sense) are automatically global to all threads. This is the primary means by which the threads communicate with each other.

### 4.2.4 The OpenMP single Pragma

In some cases we want just one thread to execute some code, even though that code is part of a parallel or other work sharing block $\|^{2}$ We use the single directive to do this, e.g.:

```
#pragma omp single
{ nth = omp_get_num_threads();
    if (nv % nth != 0) {
        printf("nv must be divisible by nth\n");
        exit(1);
    }
    chunk = nv/nth;
    printf("there are %d threads\n",nth); }
```

Since the variables nth and chunk are global and thus shared, we need not have all threads set them, hence our use of single.

### 4.2.5 The OpenMP barrier Pragma

As see in the example above, the barrier implements a standard barrier, applying to all threads.

[^27]
### 4.2.6 Implicit Barriers

Note that there is an implicit barrier at the end of each single block, which is also the case for parallel, for, and sections blocks. This can be overridden via the nowait clause, e.g.
\#pragma omp for nowait

Needless to say, the latter should be used with care, and in most cases will not be usable. On the other hand, putting in a barrier where it is not needed would severely reduce performance.

### 4.2.7 The OpenMP critical Pragma

The last construct used in this example is critical, for critical sections.

```
#pragma omp critical
{ if (mymd < md)
    { md = mymd; mv = mymv; }
}
```

It means what it says, allowing entry of only one thread at a time while others wait. Here we are updating global variables $\mathbf{m d}$ and $\mathbf{m v}$, which has to be done atomically, and critical takes care of that for us. This is much more convenient than setting up lock variables, etc., which we would do if we were programming threads code directly.

### 4.3 The OpenMP for Pragma

This one breaks up a $\mathrm{C} / \mathrm{C}++$ for loop, assigning various iterations to various threads. (The threads, of course, must have already been set up via the omp parallel pragma.) This way the iterations are done in parallel. Of course, that means that they need to be independent iterations, i.e. one iteration cannot depend on the result of another.

### 4.3.1 Example: Dijkstra with Parallel for Loops

Here's how we could use this construct in the Dijkstra program :
// Dijkstra.c
// OpenMP example program (OMPi version): Dijkstra shortest-path finder

```
// in a bidirectional graph; finds the shortest path from vertex 0 to
// all others
// usage: dijkstra nv print
// where nv is the size of the graph, and print is 1 if graph and min
// distances are to be printed out, O otherwise
#include <omp.h>
// global variables, shared by all threads by default
int nv, // number of vertices
    *notdone, // vertices not checked yet
    nth, // number of threads
    chunk, // number of vertices handled by each thread
    md, // current min over all threads
    mv, // vertex which achieves that min
    largeint = -1; // max possible unsigned int
unsigned *ohd, // 1-hop distances between vertices; "ohd[i][j]" is
            // ohd[i*nv+j]
        *mind; // min distances found so far
void init(int ac, char **av)
{ int i,j,tmp;
    nv = atoi(av[1]);
    ohd = malloc(nv*nv*sizeof(int));
    mind = malloc(nv*sizeof(int));
    notdone = malloc(nv*sizeof(int));
    // random graph
    for (i = 0; i < nv; i++)
        for (j = i; j < nv; j++) {
            if (j == i) ohd[i*nv+i] = 0;
            else {
                ohd[nv*i+j] = rand() % 20;
                ohd[nv*j+i] = ohd[nv*i+j];
            }
        }
    for (i = 1; i < nv; i++) {
        notdone[i] = 1;
        mind[i] = ohd[i];
    }
}
void dowork()
{
    #pragma omp parallel
    { int step, // whole procedure goes nv steps
            mymv, // vertex which attains that value
            me = omp_get_thread_num(),
            i;
        unsigned mymd; // min value found by this thread
        #pragma omp single
        { nth = omp_get_num_threads();
            printf("there are %d threads\n",nth); }
        for (step = 0; step < nv; step++) {
            // find closest vertex to 0 among notdone; each thread finds
```

```
            // closest in its group, then we find overall closest
            #pragma omp single
            { md = largeint; mv = 0; }
            mymd = largeint;
            #pragma omp for nowait
            for (i = 1; i < nv; i++) {
            if (notdone[i] && mind[i] < mymd) {
                    mymd = ohd[i];
            mymv = i;
            }
            }
            // update overall min if mine is smaller
            #pragma omp critical
            { if (mymd < md)
            { md = mymd; mv = mymv; }
            }
            // mark new vertex as done
            #pragma omp single
            { notdone[mv] = 0; }
            // now update ohd
            #pragma omp for
            for (i = 1; i < nv; i++)
            if (mind[mv] + ohd[mv*nv+i] < mind[i])
                mind[i] = mind[mv] + ohd[mv*nv+i];
            }
    }
}
int main(int argc, char **argv)
{ int i,j,print;
    init(argc,argv);
    // parallel
    dowork();
    // back to single thread
    print = atoi(argv[2]);
    if (print) {
        printf("graph weights:\n");
        for (i = 0; i < nv; i++) {
            for (j = 0; j < nv; j++)
                printf("%u ",ohd[nv*i+j]);
            printf("\n");
        }
        printf("minimum distances:\n");
        for (i = 1; i < nv; i++)
            printf("%u\n",mind[i]);
    }
}
```

The work which used to be done in the function findmymin() is now done here:

```
#pragma omp for
for (i = 1; i < nv; i++) {
    if (notdone[i] && mind[i] < mymd) {
        mymd = ohd[i];
        mymv = i;
```

```
    }
}
```

Each thread executes one or more of the iterations, i.e. takes responsibility for one or more values of $i$. This occurs in parallel, so as mentioned earlier, the programmer must make sure that the iterations are independent; there is no predicting which threads will do which values of $\mathbf{i}$, in which order. By the way, for obvious reasons OpenMP treats the loop index, $\mathbf{i}$ here, as private even if by context it would be shared.

### 4.3.2 Nested Loops

If we use the for pragma to nested loops, by default the pragma applies only to the outer loop. We can of course insert another for pragma inside, to parallelize the inner loop.

Or, starting with OpenMP version 3.0, one can use the collapse clause, e.g.

```
#pragma omp parallel for collapse(2)
```

to specify two levels of nesting in the assignment of threads to tasks.

### 4.3.3 Controlling the Partitioning of Work to Threads: the schedule Clause

In this default version of the for construct, iterations are executed by threads in unpredictable order; the OpenMP standard does not specify which threads will execute which iterations in which order. But this can be controlled by the programmer, using the schedule clause. OpenMP provides three choices for this:

- static: The iterations are grouped into chunks, and assigned to threads in round-robin fashion. Default chunk size is approximately the number of iterations divided by the number of threads.
- dynamic: Again the iterations are grouped into chunks, but here the assignment of chunks to threads is done dynamically. When a thread finishes working on a chunk, it asks the OpenMP runtime system to assign it the next chunk in the queue. Default chunk size is 1 .
- guided: Similar to dynamic, but with the chunk size decreasing as execution proceeds.

For instance, our original version of our program in Section 4.2 broke the work into chunks, with chunk size being the number vertices divided by the number of threads.

For the Dijkstra algorithm, for instance, we could get the same operation with less code by asking OpenMP to do the chunking for us, say with a chunk size of 8 :

```
#pragma omp for schedule(static)
for (i = 1; i < nv; i++) {
    if (notdone[i] && mind[i] < mymd) {
        mymd = ohd[i];
        mymv = i;
    }
}
#pragma omp for schedule(static)
for (i = 1; i < nv; i++)
    if (mind[mv] + ohd[mv*nv+i] < mind[i])
        mind[i] = mind[mv] + ohd[mv*nv+i];
```

Note again that this would have the same effect as our original code, which each thread handling one chunk of contiguous iterations within a loop. So it's just a programming convenience for us in this case. (If the number of threads doesn't evenly divide the number of iterations, OpenMP will fix that up for us too.)

The more general form is

```
#pragma omp for schedule(static,chunk)
```

Here static is still a keyword but chunk is an actual argument. However, setting the chunk size in the schedule() clause is a compile-time operation. If you wish to have the chunk size set at run time, call omp_set_schedule() in conjunction with the runtime clause. Example:

```
int main(int argc, char **argv)
{
    n = atoi(argv[1]);
    int chunk = atoi(argv[2]);
    omp_set_schedule(omp_sched_static, chunk);
    #pragma omp parallel
    {
        #pragma omp for schedule(runtime)
        for (i = 1; i < n; i++) {
        }
    }
}
```

Or set the OMP_SCHEDULE environment variable.
The syntax is the same for dynamic and guided.
As discussed in Section 2.4, on the one hand, large chunks are good, due to there being less overhead - every time a thread finishes a chunk, it must go through the critical section, which serializes our parallel program and thus slows things down. On the other hand, if chunk sizes are large, then toward the end of the work, some threads may be working on their last chunks while others have finished and are now idle, thus foregoing potential speed enhancement. So it would be nice to have large chunks at the beginning of the run, to reduce the overhead, but smaller chunks at the end. This can be done using the guided clause.

For the Dijkstra algorithm, for instance, we could have this:

```
#pragma omp for schedule(guided)
for (i = 1; i < nv; i++) {
    if (notdone[i] && mind[i] < mymd) {
        mymd = ohd[i];
        mymv = i;
    }
}
#pragma omp for schedule(guided)
for (i = 1; i < nv; i++)
    if (mind[mv] + ohd[mv*nv+i] < mind[i])
        mind[i] = mind[mv] + ohd[mv*nv+i];
```

There are other variations of this available in OpenMP. However, in Section 2.4, I showed that these would seldom be necessary or desirable; having each thread handle a single chunk would be best.

See Section 2.4 for a timing example.

```
1 setenv OMP_SCHEDULE "static ,20"
```


### 4.3.4 Example: In-Place Matrix Transpose

This method works in-place, a virtue if we are short on memory. Its cache performance is probably poor, though. It may be better to look at horizontal slabs above the diagonal, say, and trade them with vertical ones below the diagonal.

```
#include <omp.h>
// translate from 2-D to 1-D indices
int onedim(int n, int i, int j) { return n * i + j; }
```

```
void transp(int *m, int n)
{
    #pragma omp parallel
    { int i,j,tmp;
        // walk through all the above-diagonal elements, swapping them
        // with their below-diagonal counterparts
        #pragma omp for
        for (i = 0; i < n; i++) {
            for (j = i+1; j < n; j++) {
                tmp =m[onedim (n, i, j)];
                m[onedim(n, i, j)] =m[onedim(n,j,i)];
            m[onedim(n,j,i)] = tmp;
            }
        }
    }
}
```


### 4.3.5 The OpenMP reduction Clause

The name of this OpenMP clause alludes to the term reduction in functional programming. Many parallel programming languages include such operations, to enable the programmer to more conveniently (and often more efficiently) have threads/processors cooperate in computing sums, products, etc. OpenMP does this via the reduction clause.

For example, consider

```
int z;
#pragma omp for reduction(+:z)
for (i = 0; i < n; i++) z += x[i];
```

The pragma says that the threads will share the work as in our previous discussion of the for pragma. In addition, though, there will be independent copies of $\mathbf{z}$ maintained for each thread, each initialized to 0 before the loop begins. When the loop is entirely done, the values of $\mathbf{z}$ from the various threads will be summed, of course in an atomic manner.

Note that the + operator not only indicates that the values of $\mathbf{z}$ are to be summed, but also that their initial values are to be 0 . If the operator were ${ }^{*}$, say, then the product of the values would be computed, and their initial values would be 1 .

One can specify several reduction variables to the right of the colon, separated by commas.
Our use of the reduction clause here makes our programming much easier. Indeed, if we had old serial code that we wanted to parallelize, we would have to make no change to it! OpenMP is taking
care of both the work splitting across values of $\mathbf{i}$, and the atomic operations. Moreover-note this carefully-it is efficient, because by maintaining separate copies of $\mathbf{z}$ until the loop is done, we are reducing the number of serializing atomic actions, and are avoiding time-costly cache coherency transactions and the like.

Without this construct, we would have to do

```
int z,myz=0;
#pragma omp for private(myz)
for (i = 0; i < n; i++) myz += x[i];
#pragma omp critical
{ z += myz; }
```

Here are the eligible operators and the corresponding initial values:
In C/C++, you can use reduction with $+,-,{ }^{*}, \&, \mid, \& \&$ and || (and the exclusive-or operator).

| operator | initial value |
| :--- | :--- |
| + | 0 |
| - | 0 |
| $*$ | 1 |
| $\&$ | bit string of 1 s |
| $\\|$ | bit string of 0 s |
| - | 0 |
| $\& \&$ | 1 |
| $\\|$ | 0 |

The lack of other operations typically found in other parallel programming languages, such as min and max, is due to the lack of these operators in $\mathrm{C} / \mathrm{C}++$. The FORTRAN version of OpenMP does have min and max ${ }^{3}$

Note that the reduction variables must be shared by the threads, and apparently the only acceptable way to do so in this case is to declare them as global variables.

A reduction variable must be scalar, in $\mathrm{C} / \mathrm{C}++$. It can be an array in FORTRAN.

### 4.4 Example: Mandelbrot Set

Here's the code for the timings in Section 2.4.5

[^28]```
// compile with -D, e.g.
//
// gcc -fopenmp -o manddyn Gove.c -DDYNAMIC
//
// to get the version that uses dynamic scheduling
#include <omp.h>
#include <complex.h>
#include <time.h>
float timediff(struct timespec t1, struct timespec t2)
{ if (t1.tv_nsec > t2.tv_nsec) {
            t2.tv_sec -= 1;
            t2.tv_nsec += 1000000000;
    }
    return t2.tv_sec-t1.tv_sec + 0.000000001 * (t2.tv_nsec-t1.tv_nsec);
}
#ifdef RC
// finds chunk among 0,\ldots,n-1 to assign to thread number me among nth
// threads
void findmyrange(int n,int nth,int me,int *myrange)
{ int chunksize = n / nth;
    myrange [0] = me * chunksize;
    if (me < nth-1) myrange [1] = (me+1) * chunksize - 1;
    else myrange[1] = n - 1;
}
#include <stdlib.h>
#include <stdio.h>
// from http://www.cis.temple.edu/~ ingargio/cis71/code/randompermute.c
// It returns a random permutation of 0..n-1
int * rpermute(int n) {
    int *a=(int *)(int *) malloc(n*sizeof(int));
    // int *a= malloc(n*sizeof(int));
    int k;
    for (k = 0; k < n; k++)
        a[k] = k;
    for (k = n-1; k> 0; k--) {
        int j = rand() % (k+1);
        int temp = a[j];
        a[j] = a[k];
        a[k] = temp;
    }
    return a;
}
#endif
#define MAXITERS 1000
```

```
// globals
int count = 0;
int nptsside;
float side2;
float side4;
int inset(double complex c) {
    int iters;
    float rl,im;
    double complex z = c;
    for (iters = 0; iters < MAXITERS; iters++) {
            z = z*z + c;
            rl = creal(z);
            im = cimag(z);
            if (rl*rl + im*im > 4) return 0;
        }
    return 1;
}
int *scram;
void dowork()
{
    #ifdef RC
    #pragma omp parallel reduction(+:count)
    #else
    #pragma omp parallel
    #endif
    {
            int x,y; float xv,yv;
            double complex z;
            #ifdef STATIC
            #pragma omp for reduction(+:count) schedule(static)
            #elif defined DYNAMIC
            #pragma omp for reduction(+:count) schedule(dynamic)
            #elif defined GUIDED
            #pragma omp for reduction(+:count) schedule(guided)
            #endif
            #ifdef RC
            int myrange[2];
            int me = omp_get_thread_num();
            int nth = omp_get_num_threads();
            int i;
            findmyrange(nptsside, nth,me, myrange);
            for (i = myrange[0]; i <= myrange[1]; i++) {
            x = scram[i];
            #else
            for (x=0; x<nptsside; x++) {
            #endif
            for ( y=0; y<nptsside; y++) {
```

```
            xv = (x - side2) / side4;
            yv = (y - side2) / side4;
            z = xv + yv*I;
            if (inset(z)) {
                        count++;
            }
            }
        }
    }
}
int main(int argc, char **argv)
{
    nptsside = atoi(argv[1]);
    side2 = nptsside / 2.0;
    side4 = nptsside / 4.0;
    struct timespec bgn,nd;
    clock_gettime(CLOCKREALTIME, &bgn);
    #ifdef RC
    scram = rpermute(nptsside);
    #endif
    dowork();
    // implied barrier
    printf("%d\n",count);
    clock_gettime(CLOCK_REALTIME,&nd);
    printf("%f\n",timediff(bgn, nd ));
}
```

The code is similar to that of a number of books and Web sites, such as the Gove book cited in Section 2.2. Here RC is the random chunk method discussed in Section 2.4.

### 4.5 The Task Directive

This is new to OpenMP 3.0. The basic idea is to set up a task queue: When a thread encounters a task directive, it arranges for some thread to execute the associated block - at some time. The first thread can continue. Note that the task might not execute right away; it may have to wait for some thread to become free after finishing another task. Also, there may be more tasks than threads, also causing some threads to wait.

Note that we could arrange for all this ourselves, without task. We'd set up our own work queue, as a shared variable, and write our code so that whenever a thread finished a unit of work, it would
delete the head of the queue. Whenever a thread generated a unit of work, it would add it to the que. Of course, the deletion and addition would have to be done atomically. All this would amount to a lot of coding on our part, so task really simplifies the programming.

### 4.5.1 Example: Quicksort

```
// OpenMP example program: quicksort; not necessarily efficient
void swap(int *yi, int *yj)
{ int tmp = *yi;
    *yi = *yj;
    *yj = tmp;
}
int *separate(int *x, int low, int high)
{ int i,pivot,last;
    pivot = x[low]; // would be better to take, e.g., median of 1st 3 elts
    swap(x+low,x+high);
    last = low;
    for (i = low; i < high; i++) {
        if (x[i] <= pivot) {
            swap(x+last,x+i);
            last += 1;
        }
    }
    swap(x+last,x+high);
    return last;
}
// quicksort of the array z, elements zstart through zend; set the
// latter to 0 and m-1 in first call, where m is the length of z;
// firstcall is 1 or 0, according to whether this is the first of the
// recursive calls
void qs(int *z, int zstart, int zend, int firstcall)
{
    #pragma omp parallel
    { int part;
        if (firstcall == 1) {
            #pragma omp single nowait
            qs(z,0,zend,0);
        } else {
            if (zstart < zend) {
                        part = separate(z,zstart,zend);
                        #pragma omp task
                        qs(z,zstart,part-1,0);
                        #pragma omp task
                        qs(z,part+1,zend,0);
            }
        }
    }
}
// test code
```

```
main(int argc, char**argv)
{ int i,n,*w;
    n = atoi(argv[1]);
    w = malloc(n*sizeof(int));
    for (i = 0; i < n; i++) w[i] = rand();
    qs(w,0,n-1,1);
    if (n < 25)
        for (i = 0; i < n; i++) printf("%d\n",w[i]);
}
```

The code

```
if (firstcall == 1) {
    #pragma omp single nowait
    qs(z,0,zend,0);
```

gets things going. We want only one thread to execute the root of the recursion tree, hence the need for the single clause. After that, the code

```
part = separate(z,zstart,zend);
#pragma omp task
qs(z,zstart,part-1,0);
```

sets up a call to a subtree, with the task directive stating, "OMP system, please make sure that this subtree is handled by some thread."

There are various refinements, such as the barrier-like taskwait clause.

### 4.6 Other OpenMP Synchronization Issues

Earlier we saw the critical and barrier constructs. There is more to discuss, which we do here.

### 4.6.1 The OpenMP atomic Clause

The critical construct not only serializes your program, but also it adds a lot of overhead. If your critical section involves just a one-statement update to a shared variable, e.g.
x += $y$;
etc., then the OpenMP compiler can take advantage of an atomic hardware instruction, e.g. the LOCK prefix on Intel, to set up an extremely efficient critical section, e.g.

```
#pragma omp atomic
x += y;
```

Since it is a single statement rather than a block, there are no braces.
The eligible operators are:
$++,--,+=, *=, \ll=, ~ \&=, 1=$

### 4.6.2 Memory Consistency and the flush Pragma

Consider a shared-memory multiprocessor system with coherent caches, and a shared, i.e. global, variable $\mathbf{x}$. If one thread writes to $\mathbf{x}$, you might think that the cache coherency system will ensure that the new value is visible to other threads. But as discussed in Section 3.6, it is is not quite so simple as this.

For example, the compiler may store $\mathbf{x}$ in a register, and update $\mathbf{x}$ itself at certain points. In between such updates, since the memory location for $\mathbf{x}$ is not written to, the cache will be unaware of the new value, which thus will not be visible to other threads. If the processors have write buffers etc., the same problem occurs.

In other words, we must account for the fact that our program could be run on different kinds of hardware with different memory consistency models. Thus OpenMP must have its own memory consistency model, which is then translated by the compiler to mesh with the hardware.

OpenMP takes a relaxed consistency approach, meaning that it forces updates to memory ("flushes") at all synchronization points, i.e. at:

## - barrier

- entry/exit to/from critical
- entry/exit to/from ordered
- entry/exit to/from parallel
- exit from parallel for
- exit from parallel sections
- exit from single

In between synchronization points, one can force an update to $\mathbf{x}$ via the flush pragma:
\#pragma omp flush (x)

The flush operation is obviously architecture-dependent. OpenMP compilers will typically have the proper machine instructions available for some common architectures. For the rest, it can force a flush at the hardware level by doing lock/unlock operations, though this may be costly in terms of time.

### 4.7 Combining Work-Sharing Constructs

In our examples of the for pragma above, that pragma would come within a block headed by a parallel pragma. The latter specifies that a team of theads is to be created, with each one executing the given block, while the former specifies that the various iterations of the loop are to be distributed among the threads. As a shortcut, we can combine the two pragmas:
\#pragma omp parallel for

This also works with the sections pragma.

### 4.8 The Rest of OpenMP

There is much, much more to OpenMP than what we have seen here. To see the details, there are many Web pages you can check, and there is also the excellent book, Using OpenMP: Portable Shared Memory Parallel Programming, by Barbara Chapman, Gabriele Jost and Ruud Van Der Pas, MIT Press, 2008. The book by Gove cited in Section 2.2 also includes coverage of OpenMP.

### 4.9 Compiling, Running and Debugging OpenMP Code

### 4.9.1 Compiling

There are a number of open source compilers available for OpenMP, including:

- Omni: This is available at (http://phase.hpcc.jp/Omni/). To compile an OpenMP program in x.c and create an executable file $\mathbf{x}$, run

```
omcc -g -o x x.c
```

Note: Apparently declarations of local variables cannot be made in the midst of code; they must precede all code within a block.

- Ompi: You can download this at http://www.cs.uoi.gr/~ompi/index.html. Compile x.c by
ompicc -g -o x x.c
- GCC, version 4.2 or later $\sqrt{4}^{\text {Compile x.c via }}$
gcc -fopenmp -g -o x x.c
You can also use -lgomp instead of -fopenmp.


### 4.9.2 Running

Just run the executable as usual.
The number of threads will be the number of processors, by default. To change that value, set the OMP_NUM_THREADS environment variable. For example, to get four threads in the C shell, type
setenv OMP_NUM_THREADS 4

### 4.9.3 Debugging

Since OpenMP is essentially just an interface to threads, your debugging tool's threads facilities should serve you well. See Section 1.3.2.4 for the GDB case.

A possible problem, though, is that OpenMP's use of pragmas makes it difficult for the compilers to maintain your original source code line numbers, and your function and variable names. But with a little care, a symbolic debugger such as GDB can still be used. Here are some tips for the compilers mentioned above, using GDB as our example debugging tool:

- GCC: GCC maintains line numbers and names well. In earlier versions, it had a problem in that it did not not retain names of local variables within blocks controlled by omp parallel at all. That problem was fixed in version 4.4 of the GCC suite, but seems to have slipped back in with some later versions! This may be due to compiler optimizations that place variables in registers.

[^29]- Omni: The function main() in your executable is actually in the OpenMP library, and your function main() is renamed _ompc_main(). So, when you enter GDB, first set a breakpoint at your own code:
(gdb) b _ompc_main

Then run your program to this breakpoint, and set whatever other breakpoints you want.
You should find that your other variable and function names are unchanged.

- Ompi: Older versions also changed your function names, but the current version (1.2.0) doesn't. Works fine in GDB.


### 4.10 Performance

As is usually the case with parallel programming, merely parallelizing a program won't necessarily make it faster, even on shared-memory hardware. Operations such as critical sections, barriers and so on serialize an otherwise-parallel program, sapping much of its speed. In addition, there are issues of cache coherency transactions, false sharing etc.

### 4.10.1 The Effect of Problem Size

To illustrate this, I ran our original Dijkstra example (Section 4.2 on various graph sizes, on a quad core machine. Here are the timings:

| nv | nth | time |
| :--- | :--- | :--- |
| 1000 | 1 | 0.005472 |
| 1000 | 2 | 0.011143 |
| 1000 | 4 | 0.029574 |

The more parallelism we had, the slower the program ran! The synchronization overhead was just too much to be compensated by the parallel computation.

However, parallelization did bring benefits on larger problems:

| nv | nth | time |
| :--- | :--- | :--- |
| 25000 | 1 | 2.861814 |
| 25000 | 2 | 1.710665 |
| 25000 | 4 | 1.453052 |

### 4.10.2 Some Fine Tuning

How could we make our Dijkstra code faster? One idea would be to eliminate the critical section. Recall that in each iteration, the threads compute their local minimum distance values md and $\mathbf{m v}$, and then update the global values md and mv. Since the update must be atomic, this causes some serialization of the program. Instead, we could have the threads store their values mymd and mymv in a global array mymins, with each thread using a separate pair of locations within that array, and then at the end of the iteration we could have just one task scan through mymins and update md and mv.

Here is the resulting code:

```
// Dijkstra.c
// OpenMP example program: Dijkstra shortest-path finder in a
// bidirectional graph; finds the shortest path from vertex 0 to all
// others
// **** in this version, instead of having a critical section in which
// each thread updates md and mv, the threads record their mymd and mymv
// values in a global array mymins, which one thread then later uses to
// update md and mv
// usage: dijkstra nv print
// where nv is the size of the graph, and print is 1 if graph and min
// distances are to be printed out, O otherwise
#include <omp.h>
// global variables, shared by all threads by default
int nv, // number of vertices
    *notdone, // vertices not checked yet
    nth, // number of threads
    chunk, // number of vertices handled by each thread
    md, // current min over all threads
    mv, // vertex which achieves that min
    largeint = -1; // max possible unsigned int
int *mymins; // (mymd,mymv) for each thread; see dowork()
unsigned *ohd, // 1-hop distances between vertices; "ohd[i][j]" is
    // ohd[i*nv+j]
    *mind; // min distances found so far
void init(int ac, char **av)
{ int i,j,tmp;
    nv = atoi(av[1]);
    ohd = malloc(nv*nv*sizeof(int));
    mind = malloc(nv*sizeof(int));
    notdone = malloc(nv*sizeof(int));
    // random graph
```

```
    for (i = 0; i < nv; i++)
        for (j = i; j < nv; j++) {
            if (j == i) ohd[i*nv+i] = 0;
            else {
                ohd[nv*i+j] = rand() % 20;
                ohd[nv*j+i] = ohd[nv*i+j];
            }
        }
    for (i = 1; i < nv; i++) {
        notdone[i] = 1;
        mind[i] = ohd[i];
    }
}
// finds closest to 0 among notdone, among s through e
void findmymin(int s, int e, unsigned *d, int *v)
{ int i;
    *d = largeint;
    for (i = s; i <= e; i++)
        if (notdone[i] && mind[i] < *d) {
            *d = ohd[i];
            *v = i;
        }
}
// for each i in [s,e], ask whether a shorter path to i exists, through
// mv
void updatemind(int s, int e)
{ int i;
    for (i = s; i <= e; i++)
        if (mind[mv] + ohd[mv*nv+i] < mind[i])
            mind[i] = mind[mv] + ohd[mv*nv+i];
}
void dowork()
{
    #pragma omp parallel
    { int startv,endv, // start, end vertices for my thread
            step, // whole procedure goes nv steps
            me,
            mymv; // vertex which attains the min value in my chunk
            unsigned mymd; // min value found by this thread
        int i;
        me = omp_get_thread_num();
        #pragma omp single
        { nth = omp_get_num_threads();
            if (nv % nth != 0) {
                printf("nv must be divisible by nth\n");
                exit(1);
            }
            chunk = nv/nth;
            mymins = malloc(2*nth*sizeof(int));
        }
        startv = me * chunk;
        endv = startv + chunk - 1;
        for (step = 0; step < nv; step++) {
            // find closest vertex to 0 among notdone; each thread finds
            // closest in its group, then we find overall closest
```

```
            findmymin(startv,endv,&mymd,&mymv);
            mymins[2*me] = mymd;
            mymins[2*me+1] = mymv;
            #pragma omp barrier
            // mark new vertex as done
            #pragma omp single
            { md = largeint; mv = 0;
                    for (i = 1; i < nth; i++)
                    if (mymins[2*i] < md) {
                    md = mymins[2*i];
                mv = mymins[2*i+1];
                }
            notdone[mv] = 0;
            }
            // now update my section of mind
            updatemind(startv,endv);
            #pragma omp barrier
        }
    }
}
int main(int argc, char **argv)
{ int i,j,print;
    double startime,endtime;
    init(argc,argv);
    startime = omp_get_wtime();
    // parallel
    dowork();
    // back to single thread
    endtime = omp_get_wtime();
    printf("elapsed time: %f\n",endtime-startime);
    print = atoi(argv[2]);
    if (print) {
        printf("graph weights:\n");
        for (i = 0; i < nv; i++) {
            for (j = 0; j < nv; j++)
                printf("%u ",ohd[nv*i+j]);
            printf("\n");
        }
        printf("minimum distances:\n");
        for (i = 1; i < nv; i++)
            printf("%u\n",mind[i]);
    }
}
```

Let's take a look at the latter part of the code for one iteration;

```
findmymin(startv, endv,&mymd,&mymv);
mymins[2*me] = mymd;
mymins[2*me+1] = mymv;
#pragma omp barrier
// mark new vertex as done
#pragma omp single
{ notdone[mv] = 0;
    for (i = 1; i < nth; i++)
            if (mymins[2*i] < md) {
```

```
        md = mymins[2*i];
        mv = mymins[2*i+1];
    }
}
// now update my section of mind
updatemind(startv,endv);
#pragma omp barrier
```

The call to findmymin() is as before; this thread finds the closest vertex to 0 among this thread's range of vertices. But instead of comparing the result to md and possibly updating it and mv, the thread simply stores its mymd and mymv in the global array mymins. After all threads have done this and then waited at the barrier, we have just one thread update md and mv.

Let's see how well this tack worked:

| nv | nth | time |
| :--- | :--- | :--- |
| 25000 | 1 | 2.546335 |
| 25000 | 2 | 1.449387 |
| 25000 | 4 | 1.411387 |

This brought us about a $15 \%$ speedup in the two-thread case, though less for four threads.
What else could we do? Here are a few ideas:

- False sharing could be a problem here. To address it, we could make mymins much longer, changing the places at which the threads write their data, leaving most of the array as padding.
- We could try the modification of our program in Section 4.3.1, in which we use the OpenMP for pragma, as well as the refinements stated there, such as schedule.
- We could try combining all of the ideas here.


### 4.10.3 OpenMP Internals

We may be able to write faster code if we know a bit about how OpenMP works inside.
You can get some idea of this from your compiler. For example, if you use the -t option with the Omni compiler, or $-\mathbf{k}$ with Ompi, you can inspect the result of the preprocessing of the OpenMP pragmas.

Here for instance is the code produced by Omni from the call to findmymin() in our Dijkstra program:

```
# 93 "Dijkstra.c"
findmymin(startv,endv,&(mymd),&(mymv)) ; {
```

```
_ompc_enter_critical(&__ompc_lock_critical);
# 96 "Dijkstra.c"
if((mymd)<(((unsigned )(md)))){
# 97 "Dijkstra.c"
(md)=(((int ) (mymd)));
# 97 "Dijkstra.c"
(mv)=(mymv);
}_ompc_exit_critical(&__ompc_lock_critical);
```

Fortunately Omni saves the line numbers from our original source file, but the pragmas have been replaced by calls to OpenMP library functions.
With Ompi, while preprocessing of your file $\mathbf{x . c}$, the compiler produces an intermediate file $\mathbf{x} \_$ompi.c, and the latter is what is actually compiled. Your function main is renamed to _ompi_originalMain(). Your other functions and variables are renamed. For example in our Dijkstra code, the function dowork() is renamed to dowork_parallel_0. And by the way, all indenting is lost! So it's a bit hard to read, but can be very instructive.

The document, The GNU OpenMP Implementation, http://pl.postech.ac.kr/~gla/cs700-07f/ ref/openMp/libgomp.pdf, includes good outline of how the pragmas are translated.

### 4.11 Example: Root Finding

The application is described in the comments, but here are a couple of things to look for in particular:

- The variables curra and currb are shared by all the threads, but due to the nature of the application, no critical sections are needed.
- On the other hand, the barrier is essential. The reader should ponder what calamities would occur without it.

Note the disclaimer in the comments, to the effect that parallelizing this application will be fruitful only if the functioin $f()$ is very time-consuming to evaluate. It might be the output of some complex simulation, for instance, with the argument to $f()$ being some simulation parameter.

```
#include<omp.h>
#include<math.h>
// OpenMP example: root finding
// the function f() is known to be negative
// at a, positive at b, and thus has at
```

```
// least one root in (a,b); if there are
// multiple roots, only one is found;
// the procedure runs for niters iterations
// strategy: in each iteration, the current
// interval is split into nth equal parts,
// and each thread checks its subinterval
// for a sign change of f(); if one is
// found, this subinterval becomes the
// new current interval; the current guess
// for the root is the left endpoint of the
// current interval
// of course, this approach is useful in
// parallel only if f() is very expensive
// to evaluate
// for simplicity, assumes that no endpoint
// of a subinterval will ever exactly
// coincide with a root
float root(float(*f)(float),
    float inita, float initb, int niters) {
    float curra = inita;
    float currb = initb;
    #pragma omp parallel
    {
        int nth = omp_get_num_threads();
        int me = omp_get_thread_num();
        int iter;
        for (iter = 0; iter < niters; iter++) {
            #pragma omp barrier
            float subintwidth =
                (currb - curra) / nth;
            float myleft =
                curra + me * subintwidth;
            float myright = myleft + subintwidth;
            if ((*f)(myleft) < 0 &&
                    (*f)(myright) > 0) {
                    curra = myleft;
                currb = myright;
            }
        }
    }
    return curra;
}
float testf(float x) {
    return pow(x-2.1,3);
}
```

```
58
59 int main(int argc, char **argv)
60 { printf("%f\n",root(testf, - 4.1,4.1,1000)); }
```


### 4.12 Example: Mutual Outlinks

Consider the example of Section 2.4.3. We have a network graph of some kind, such as Web links. For any two vertices, say any two Web sites, we might be interested in mutual outlinks, i.e. outbound links that are common to two Web sites.

The OpenMP code below finds the mean number of mutual outlinks, among all pairs of sites in a set of Web sites. Note that it uses the method for load balancing presented in Section 2.4.3.

```
#include <omp.h>
#include <stdio.h>
// OpenMP example: finds mean number of mutual outlinks, among all
// pairs of Web sites in our set
int n, // number of sites (will assume n is even)
    nth, // number of threads (will assume n/2 divisible by nth)
    *m, // link matrix
    tot = 0; // grand total of matches
// processes row pairs (i,i+1), (i,i+2), ...
int procpairs(int i)
{ int j,k,sum=0;
    for (j = i+1; j < n; j++) {
        for (k = 0; k < n; k++)
                sum += m[n*i+k] * m[n*j+k];
    }
    return sum;
}
float dowork()
{
    #pragma omp parallel
    { int pn1,pn2,i,mysum=0;
        int me = omp_get_thread_num();
        nth = omp_get_num_threads();
        // in checking all (i,j) pairs, partition the work according to i;
        // to get good load balance, this thread me will handle all i that equal
        // me mod nth
        for (i = me; i < n; i += nth) {
            mysum += procpairs(i);
        }
        #pragma omp atomic
        tot += mysum;
        #pragma omp barrier
    }
    int divisor = n * (n-1) / 2;
```

```
    return ((float) tot)/divisor;
}
int main(int argc, char **argv)
{ int n2 = n/2,i,j;
    n = atoi(argv[1]); // number of matrix rows/cols
    int msize = n * n * sizeof(int);
    m = (int *) malloc(msize);
    // as a test, fill matrix with random 1s and 0s
    for (i = 0; i < n; i++) {
        m[n*i+i] = 0;
        for (j = 0; j < n; j++) {
            if (j != i) m[i*n+j] = rand() % 2;
        }
    }
    if (n < 10) {
        for (i = 0; i < n; i++) {
            for (j = 0; j < n; j++) printf("%d ",m[n*i+j]);
            printf("\n");
        }
    }
    tot = 0;
    float meanml = dowork();
    printf("mean = %f\n",meanml);
}
```


### 4.13 Example: Transforming an Adjacency Matrix

Say we have a graph with adjacency matrix

$$
\left(\begin{array}{llll}
0 & 1 & 0 & 0  \tag{4.1}\\
1 & 0 & 0 & 1 \\
0 & 1 & 0 & 1 \\
1 & 1 & 1 & 0
\end{array}\right)
$$

with row and column numbering starting at 0 , not 1 . We'd like to transform this to a two-column matrix that displays the links, in this case

$$
\left(\begin{array}{ll}
0 & 1  \tag{4.2}\\
1 & 0 \\
1 & 3 \\
2 & 1 \\
2 & 3 \\
3 & 0 \\
3 & 1 \\
3 & 2
\end{array}\right)
$$

For instance, there is a 1 on the far right, second row of the above matrix, meaning that in the graph there is an edge from vertex 1 to vertex 3 . This results in the row (1,3) in the transformed matrix seen above.

Suppose further that we require this listing to be in lexicographical order, sorted on source vertex and then on destination vertex. Here is code to do this computation in OpenMP:

```
// takes a graph adjacency matrix for a directed graph, and converts it
// to a 2-column matrix of pairs (i,j), meaning an edge from vertex i to
// vertex j; the output matrix must be in lexicographical order
// not claimed efficient, either in speed or in memory usage
#include <omp.h>
// needs -lrt link flag for C++
#include <time.h>
float timediff(struct timespec t1, struct timespec t2)
{ if (t1.tv_nsec > t2.tv_nsec) {
    t2.tv_sec -= 1;
    t2.tv_nsec += 1000000000;
    }
    return t2.tv_sec-t1.tv_sec + 0.000000001 * (t2.tv_nsec-t1.tv_nsec);
}
// transgraph() does this work
// arguments:
// adjm: the adjacency matrix (NOT assumed symmetric), 1 for edge, 0
// otherwise; note: matrix is overwritten by the function
// n: number of rows and columns of adjm
// nout: output, number of rows in returned matrix
// return value: pointer to the converted matrix
int *transgraph(int *adjm, int n, int *nout)
{
    int *outm, // to become the output matrix
        *num1s, // i-th element will be the number of 1s in row i of adjm
        *cumul1s; // cumulative sums in num1s
    #pragma omp parallel
    { int i,j,m;
        int me = omp_get_thread_num(),
            nth = omp_get_num_threads();
        int myrows[2];
        int tot1s;
        int outrow, num1si;
        #pragma omp single
        {
            num1s = malloc(n*sizeof(int));
            cumul1s = malloc((n+1)*sizeof(int));
        }
        // determine the rows in adjm to be handled by this thread
```

```
        findmyrange(n, nth ,me, myrows);
        // start the action
        for (i = myrows[0]; i <= myrows[1]; i++) {
            tot1s = 0;
            for (j = 0; j < n; j++)
                if (adjm[n*i+j] = 1) {
                    adjm[n*i+(tot1s++)]= j;
            }
        num1s[i] = tot1s;
        }
        #pragma omp barrier
        #pragma omp single
        {
            cumul1s [0] = 0;
            // now calculate where the output of each row in adjm
            // should start in outm
            for (m=1; m<= n; m++) {
                cumul1s [m] = cumul1s[m-1] + num1s[m-1];
            }
            *nout = cumul1s[n];
            outm = malloc(2*(* nout) * sizeof(int));
        }
        // now fill in this thread's portion
        for (i = myrows[0]; i <= myrows[1]; i++) {
            outrow = cumul1s[i];
            num1si = num1s[i];
            for (j = 0; j < num1si; j++) {
                outm [2*(outrow+j)] = i;
            outm[2*(outrow+j)+1] = adjm[n*i+j];
            }
        }
        #pragma omp barrier
    }
    return outm;
}
int main(int argc, char **argv)
{ int i,j;
    int *adjm;
    int n = atoi(argv[1]);
    int nout;
    int *outm;
    adjm = malloc(n*n*sizeof(int));
    for (i = 0; i < n; i++)
        for (j = 0; j < n; j++)
            if (i == j) adjm[n*i+j] = 0;
            else adjm[n*i+j]= rand() % 2;
    struct timespec bgn,nd;
    clock_gettime(CLOCK_REALTIME, &bgn);
```

```
    outm = transgraph(adjm,n,&nout);
    printf("number of output rows: %d\n", nout);
    clock_gettime(CLOCK_REALTIME, &nd);
    printf("%f\n",timediff(bgn, nd ));
    if (n <= 10)
        for (i = 0; i < nout; i++)
        printf("%d %d\n",outm[2*i],outm[2*i+1]);
}
// finds chunk among 0,\ldots,n-1 to assign to thread number me among nth
// threads
void findmyrange(int n, int nth,int me,int *myrange)
{ int chunksize = n / nth;
    myrange [0] = me * chunksize;
    if (me < nth-1) myrange[1] = (me+1) * chunksize - 1;
    else myrange[1] = n - 1;
}
```


### 4.14 Locks with OpenMP

Though one of OpenMP's best virtues is that you can avoid working with those pesky lock variables needed for straight threads programming, there are still some instances in which lock variables may be useful. OpenMP does provide for locks:

- declare your locks to be of type omp_lock_t
- call omp_set_lock() to lock the lock
- call omp_unset_lock() to unlock the lock


### 4.15 Other Examples of OpenMP Code in This Book

There are additional OpenMP examples in later sections of this book, such as $5^{5}$

- sampling bucket sort, Section 1.3.2.6

[^30]- parallel prefix sum/run-length decoding, Section 10.3 .
- matrix multiplication, Section 11.3.2.1.
- Jacobi algorithm for solving systems of linear equations, with a good example of the OpenMP reduction clause, Section 11.5.4
- another implementation of Quicksort, Section 12.1.2


## Chapter 5

## Introduction to GPU Programming with CUDA

Even if you don't play video games, you can be grateful to the game players, as their numbers have given rise to a class of highly powerful parallel processing devices - graphics processing units (GPUs). Yes, you program right on the video card in your computer, even though your program may have nothing to do with graphics or games.

### 5.1 Overview

The video game market is so lucrative that the industry has developed ever-faster GPUs, in order to handle ever-faster and ever-more visually detailed video games. These actually are parallel processing hardware devices, so around 2003 some people began to wonder if one might use them for parallel processing of nongraphics applications.

Originally this was cumbersome. One needed to figure out clever ways of mapping one's application to some kind of graphics problem, i.e. ways to disguising one's problem so that it appeared to be doing graphics computations. Though some high-level interfaces were developed to automate this transformation, effective coding required some understanding of graphics principles.

But current-generation GPUs separate out the graphics operations, and now consist of multiprocessor elements that run under the familiar shared-memory threads model. Thus they are easily programmable. Granted, effective coding still requires an intimate knowledge of the hardwre, but at least it's (more or less) familiar hardware, not requiring knowledge of graphics.

Moreover, unlike a multicore machine, with the ability to run just a few threads at one time, e.g. four threads on a quad core machine, GPUs can run hundreds or thousands of threads at once.

There are various restrictions that come with this, but you can see that there is fantastic potential for speed here.

NVIDIA has developed the CUDA language as a vehicle for programming on their GPUs. It's basically just a slight extension of C, and has become very popular. More recently, the OpenCL language has been developed by Apple, AMD and others (including NVIDIA). It too is a slight extension of C , and it aims to provide a uniform interface that works with multicore machines in addition to GPUs. OpenCL is not yet in as broad use as CUDA, so our discussion here focuses on CUDA and NVIDIA GPUs.

Also, the discussion will focus on NVIDIA's Tesla line. There is also a newer, more versatile line called Fermi, but unless otherwise stated, all statements refer to Tesla.

Some terminology:

- A CUDA program consists of code to be run on the host, i.e. the CPU, and code to run on the device, i.e. the GPU.
- A function that is called by the host to execute on the device is called a kernel.
- Threads in an application are grouped into blocks. The entirety of blocks is called the grid of that application.


### 5.2 Example: Calculate Row Sums

Here's a sample program. And I've kept the sample simple: It just finds the sums of all the rows of a matrix.

```
#include <stdio.h>
#include <stdlib.h>
#include <cuda.h>
// CUDA example: finds row sums of an integer matrix m
// findielt() finds the rowsum of one row of the nxn matrix m, storing the
// result in the corresponding position in the rowsum array rs; matrix
// stored as 1-dimensional, row-major order
__global_- void find1elt(int *m, int *rs, int n)
{
    int rownum = blockIdx.x; // this thread will handle row # rownum
    int sum = 0;
    for (int k = 0; k < n; k++)
        sum += m[rownum*n+k];
    rs[rownum] = sum;
```

```
}
int main(int argc, char **argv)
{
    int n = atoi(argv[1]); // number of matrix rows/cols
    int *hm, // host matrix
            *dm, // device matrix
            *hrs, // host rowsums
            *drs; // device rowsums
    int msize = n * n * sizeof(int); // size of matrix in bytes
    // allocate space for host matrix
    hm = (int *) malloc(msize);
    // as a test, fill matrix with consecutive integers
    int t = 0,i,j;
    for (i = 0; i < n; i++) {
        for (j = 0; j < n; j++) {
            hm[i*n+j] = t++;
        }
    }
    // allocate space for device matrix
    cudaMalloc((void **)&dm, msize);
    // copy host matrix to device matrix
    cudaMemcpy (dm,hm, msize , cudaMemcpyHostToDevice);
    // allocate host, device rowsum arrays
    int rssize = n * sizeof(int);
    hrs = (int *) malloc(rssize);
    cudaMalloc((void **)&drs, rssize);
    // set up parameters for threads structure
    dim3 dimGrid(n,1); // n blocks
    dim3 dimBlock (1,1,1); // 1 thread per block
    // invoke the kernel
    find1elt<<<dimGrid, dimBlock>>>(dm, drs,n);
    // wait for kernel to finish
    cudaThreadSynchronize();
    // copy row vector from device to host
    cudaMemcpy(hrs, drs, rssize, cudaMemcpyDeviceToHost);
    // check results
    if (n< 10) for(int i=0; i<n; i++) printf("%d\n",hrs[i]);
    // clean up
    free (hm);
    cudaFree(dm);
    free(hrs);
    cudaFree(drs);
}
```

This is mostly C, with a bit of CUDA added here and there. Here's how the program works:

- Our main() runs on the host.
- Kernel functions are identified by _-global_- void. They are called by the host and run on the device, thus serving as entries to the device.
We have only one kernel invocation here, but could have many, say with the output of one serving as input to the next.
- Other functions that will run on the device, called by functions running on the device, must be identified by __device_-, e.g.
__device__ int sumvector(float *x, int n)
Note that unlike kernel functions, device functions can have return values, e.g. int above.
- When a kernel is called, each thread runs it. Each thread receives the same arguments.
- Each block and thread has an ID, stored in programmer-accessible structs blockIdx and threadIdx. We'll discuss the details later, but for now, we'll just note that here the statement

```
int rownum = blockIdx.x;
```

picks up the block number, which our code in this example uses to determine which row to sum.

- One calls cudaMalloc() on the host to dynamically allocate space on the device's memory ${ }^{1}$ Execution of the statement

```
cudaMalloc((void **)&drs,rssize);
```

allocates space on the device, pointed to by drs, a variable in the host's address space.
The space allocated by a cudaMalloc() call on the device is global to all kernels, and resides in the global memory of the device (details on memory types later).
One can also allocate device memory statically. For example, the statement
__device int $z[100]$;
appearing outside any function definition would allocate space on device global memory, with scope global to all kernels. However, it is not accessible to the host.

- Data is transferred to and from the host and device memories via cudaMemcpy(). The fourth argument specifies the direction, e.g. cudaMemcpyHostToDevice, cudaMemcpyDeviceToHost or cudaMemcpyDeviceToDevice.
- Kernels return void values, so values are returned via a kernel's arguments.

[^31]- Device functions (which we don't have here) can return values. They are called only by kernel functions or other device functions.
- Note carefully that a call to the kernel doesn't block; it returns immediately. For that reason, the code above has a host barrier call, to avoid copying the results back to the host from the device before they're ready:
cudaThreadSynchronize();

On the other hand, if our code were to have another kernel call, say on the next line after

```
find1elt<<<dimGrid,dimBlock>>>(dm,drs,n);
```

and if some of the second call's input arguments were the outputs of the first call, there would be an implied barrier betwwen the two calls; the second would not start execution before the first finished.

Calls like cudaMemcpy () do block until the operation completes.
There is also a thread barrier available for the threads themselves, at the block level. The call is
__syncthreads();

This can only be invoked by threads within a block, not across blocks. In other words, this is barrier synchronization within blocks.

- I've written the program so that each thread will handle one row of the matrix. I've chosen to store the matrix in one-dimensional form in row-major order, and the matrix is of size n x n, so the loop

```
for (int k = 0; k < n; k++)
```

    sum \(+=\mathrm{m}[\) rownum*n+k];
    will indeed traverse the n elements of row number rownum, and compute their sum. That sum is then placed in the proper element of the output array:
rs[rownum] = sum;

- After the kernel returns, the host must copy the result back from the device memory to the host memory, in order to access the results of the call.


### 5.3 Understanding the Hardware Structure

Scorecards, get your scorecards here! You can't tell the players without a scorecard - classic cry of vendors at baseball games

Know thy enemy-Sun Tzu, The Art of War
The enormous computational potential of GPUs cannot be unlocked without an intimate understanding of the hardware. This of course is a fundamental truism in the parallel processing world, but it is acutely important for GPU programming. This section presents an overview of the hardware.

### 5.3.1 Processing Units

A GPU consists of a large set of streaming multiprocessors (SMs). Since each SM is essentially a multicore machine in its own right, you might say the GPU is a multi-multiprocessor machine.

Each SM consists of a number of streaming processors (SPs), individual cores. The cores run threads, as with ordinary cores, but threads in an SM run in lockstep, to be explained below.

It is important to understand the motivation for this SM/SP hierarchy: Two threads located in different SMs cannot synchronize with each other in the barrier sense. Though this sounds like a negative at first, it is actually a great advantage, as the independence of threads in separate SMs means that the hardware can run faster. So, if the CUDA application programmer can write his/her algorithm so as to have certain independent chunks, and those chunks can be assigned to different SMs (we'll see how, shortly), then that's a "win."

Note that at present, word size is 32 bits. Thus for instance floating-point operations in hardware were originally in single precision only, though newer devices are capable of double precision.

### 5.3.2 Thread Operation

GPU operation is highly threaded, and again, understanding of the details of thread operation is key to good performance.

### 5.3.2.1 SIMT Architecture

When you write a CUDA application program, you partition the threads into groups called blocks. The hardware will assign an entire block to a single SM, though several blocks can run in the same SM. The hardware will then divide a block into warps, 32 threads to a warp. Knowing that the
hardware works this way, the programmer controls the block size and the number of blocks, and in general writes the code to take advantage of how the hardware works.

The central point is that all the threads in a warp run the code in lockstep. During the machine instruction fetch cycle, the same instruction will be fetched for all of the threads in the warp. Then in the execution cycle, each thread will either execute that particular instruction or execute nothing. The execute-nothing case occurs in the case of branches; see below. This is the classical single instruction, multiple data (SIMD) pattern used in some early special-purpose computers such as the ILLIAC; here it is called single instruction, multiple thread (SIMT).

The syntactic details of grid and block configuration will be presented in Section 5.3.4.

### 5.3.2.2 The Problem of Thread Divergence

The SIMT nature of thread execution has major implications for performance. Consider what happens with if/then/else code. If some threads in a warp take the "then" branch and others go in the "else" direction, they cannot operate in lockstep. That means that some threads must wait while others execute. This renders the code at that point serial rather than parallel, a situation called thread divergence. As one CUDA Web tutorial points out, this can be a "performance killer." (On the other hand, threads in the same block but in different warps can diverge with no problem.)

### 5.3.2.3 "OS in Hardware"

Each SM runs the threads on a timesharing basis, just like an operating system (OS). This timesharing is implemented in the hardware, though, not in software as in the OS case.

The "hardware OS" runs largely in analogy with an ordinary OS:

- A process in an ordinary OS is given a fixed-length timeslice, so that processes take turns running. In a GPU's hardware OS, warps take turns running, with fixed-length timeslices.
- With an ordinary OS, if a process reaches an input/output operation, the OS suspends the process while I/O is pending, even if its turn is not up. The OS then runs some other process instead, so as to avoid wasting CPU cycles during the long period of time needed for the I/O.

With an SM, though, the analogous situation occurs when there is a long memory operation, to global memory; if a a warp of threads needs to access global memory (including local memory; see below), the SM will schedule some other warp while the memory access is pending.

The hardware support for threads is extremely good; a context switch takes very little time, quite a contrast to the OS case. Moreover, as noted above, the long latency of global memory may be solvable by having a lot of threads that the hardware can timeshare to hide that latency; while one warp is fetching data from memory, another warp can be executing, thus not losing time due to the long fetch delay. For these reasons, CUDA programmers typically employ a large number of threads, each of which does only a small amount of work-again, quite a contrast to something like OpenMP.

### 5.3.3 Memory Structure

The GPU memory hierarchy plays a key role in performance. Let's discuss the most important two types of memory first - shared and global.

### 5.3.3.1 Shared and Global Memory

Here is a summary:

| type | shared | global |
| ---: | ---: | ---: |
| scope | glbl. to block | glbl. to app. |
| size | small | large |
| location | on-chip | off-chip |
| speed | blinding | molasses |
| lifetime | kernel | application |
| host access? | no | yes |
| cached? | no | no |

In prose form:

- Shared memory: All the threads in an SM share this memory, and use it to communicate among themselves, just as is the case with threads in CPUs. Access is very fast, as this memory is on-chip. It is declared inside the kernel, or in the kernel call (details below).
On the other hand, shared memory is small, currently 16 K bytes per SM, and the data stored in it are valid only for the life of the currently-executing kernel. Also, shared memory cannot be accessed by the host.
- Global memory: This is shared by all the threads in an entire application, and is persistent across kernel calls, throughout the life of the application, i.e. until the program running on the host exits. It is usually much larger than shared memory. It is accessible from the host. Pointers to global memory can (but do not have to) be declared outside the kernel.

On the other hand, global memory is off-chip and very slow, taking hundreds of clock cycles per access instead of just a few. As noted earlier, this can be ameliorated by exploiting latency hiding; we will elaborate on this in Section 5.3.3.2.

The reader should pause here and reread the above comparison between shared and global memories. The key implication is that shared memory is used essentially as a programmer-managed cache. Data will start out in global memory, but if a variable is to be accessed multiple times by the GPU code, it's probably better for the programmer to write code that copies it to shared memory, and then access the copy instead of the original. If the variable is changed and is to be eventually transmitted back to the host, the programmer must include code to copy it back to global memory.

Accesses to global and shared memory are done via half-warps, i.e. an attempt is made to do all memory accesses in a half-warp simultaneously. In that sense, only threads in a half-warp run simultaneously, but the full warp is scheduled to run contemporaneously by the hardware OS, first one half-warp and then the other.

The host can access global memory via cudaMemcpy(), as seen earlier. It cannot access shared memory. Here is a typical pattern:

```
__global__ void abckernel(int *abcglobalmem)
{
    __shared__ int abcsharedmem[100];
    // ... code to copy some of abcglobalmem to some of abcsharedmem
    // ... code for computation
    // ... code to copy some of abcsharedmem to some of abcglobalmem
}
```

Typically you would write the code so that each thread deals with its own portion of the shared data, e.g. its own portion of abcsharedmem and abcglobalmem above. However, all the threads in that block can read/write any element in abcsharedmem.
Shared memory consistency (recall Section 3.6) is sequential within a thread, but relaxed among threads in a block: A write by one thread is not guaranteed to be visible to the others in a block until __syncthreads() is called. On the other hand, writes by a thread will be visible to that same thread in subsequent reads without calling _-syncthreads(). Among the implications of this is that if each thread writes only to portions of shared memory that are not read by other threads in the block, then _-syncthreads() need not be called.

In the code fragment above, we allocated the shared memory through a C-style declaration:
__shared__ int abcsharedmem[100];

It is also possible to allocate shared memory in the kernel call, along with the block and thread configuration. Here is an example:

```
#include <stdio.h>
#include <stdlib.h>
#include <cuda.h>
// CUDA example: illustrates kernel-allocated shared memory; does
// nothing useful, just copying an array from host to device global,
// then to device shared, doubling it there, then copying back to device
// global then host
_-global__ void doubleit(int *dv, int n)
{ extern _-shared_- int sv[];
        int me = threadIdx.x;
        // threads share in copying dv to sv, with each thread copying one
        // element
        sv [me] = 2 * dv[me];
        dv[me] = sv[me];
}
int main(int argc, char **argv)
{
    int n = atoi(argv[1]); // number of matrix rows/cols
    int *hv, // host array
            *dv; // device array
        int vsize = n * sizeof(int); // size of array in bytes
        // allocate space for host array
        hv = (int *) malloc(vsize);
        // fill test array with consecutive integers
        int t = 0, i;
        for (i = 0; i < n; i++)
            hv[i] = t++;
        // allocate space for device array
        cudaMalloc((void **)&dv, vsize);
        // copy host array to device array
        cudaMemcpy(dv,hv, vsize, cudaMemcpyHostToDevice);
        // set up parameters for threads structure
        dim3 dimGrid (1,1);
        dim3 dimBlock(n,1,1); // all n threads in the same block
        // invoke the kernel; third argument is amount of shared memory
        doubleit <<<<dimGrid, dimBlock, vsize >>> (dv,n);
        // wait for kernel to finish
        cudaThreadSynchronize();
        // copy row array from device to host
        cudaMemcpy(hv,dv, vsize, cudaMemcpyDeviceToHost);
        // check results
        if (n< 10) for(int i=0; i<n; i++) printf("%d\n",hv[i]);
        // clean up
        free(hv);
        cudaFree(dv);
}
```

Here the variable sv is kernel allocated. It's declared in the statement
extern _-shared_- int sv[];
but actually allocated during the kernel invocation
doubleit<<<dimGrid,dimBlock,vsize>>>(dv,n);
in that third argument within the chevrons, vsize.
Note that one can only directly declare one region of space in this manner. This has two implications:

- Suppose we have two __device__ functions, each declared an extern __shared__ array like this. Those two arrays will occupy the same place in memory!
- Suppose within one __device__ function, we wish to have two extern __shared__ arrays. We cannot do that literally, but we can share the space via subarrays, e.g.:
int $* x=8 s v[120]$;
would set up $\mathbf{x}$ as a subarray of $\mathbf{s v}$ above, starting at element 120 .

One can also set up shared arrays of fixed length in the same code. Declare them before the variable-length one.

In our example above, the array sv is syntactically local to the function doubleit(), but is shared by all invocations of that function in the block, thus acting "global" to them in a sense. But the point is that it is not accessible from within other functions running in that block. In order to achieve the latter situation, a shared array can be declared outside any function.

### 5.3.3.2 Global-Memory Performance Issues

As noted, the latency (Section 2.5) for global memory is quite high, on the order of hundreds of clock cycles. However, the hardware attempts to ameliorate this problem in a couple of ways.

First, as mentioned earlier, if a warp has requested a global memory access that will take a long time, the harware will schedule another warp to run while the first is waiting for the memory access to complete. This is an example of a common parallel processing technique called latency hiding.

Second, the bandwidth (Section 2.5) to global memory can be high, due to hardware actions called coalescing. This simply means that if the hardware sees that the threads in this half-warp (or at
least the ones currently accessing global memory) are accessing consecutive words, the hardware can execute the memory requests in groups of up to 32 words at a time. This works because the memory is low-order interleaved (Section 3.2.1), and is true for both reads and writes.

The newer GPUs go even further, coalescing much more general access patterns, not just to consecutive words.

The programmer may be able to take advantage of coalescing, by a judicious choice of algorithms and/or by inserting padding into arrays (Section 3.2.2).

### 5.3.3.3 Shared-Memory Performance Issues

Shared memory is divided into banks, in a low-order interleaved manner (recall Section 3.2): Words with consecutive addresses are stored in consecutive banks, mod the number of banks, i.e. wrapping back to 0 when hitting the last bank. If for instance there are 8 banks, addresses $0,8,16, \ldots$ will be in bank 0 , addresses $1,9,17, \ldots$ will be in bank 1 and so on. (Actually, older devices have 16 banks, while newer ones have 32.) The fact that all memory accesses in a half-warp are attempted simultaneously implies that the best access to shared memory arises when the accesses are to different banks, just as for the case of global memory.

An exception occurs in broadcast. If all threads in the block wish to read from the same word in the same bank, the word will be sent to all the requestors simultaneously without conflict. However, if only some theads try to read the same word, there may or may not be a conflict, as the hardware chooses a bank for broadcast in some unspecified way.

As in the discussion of global memory above, we should write our code to take advantage of these structures.

The biggest performance issue with shared memory is its size, as little as 16 K per SM in many GPU cards.

### 5.3.3.4 Host/Device Memory Transfer Performance Issues

Copying data between host and device can be a major bottleneck. One way to ameliorate this is to use cudaMallocHost() instead of malloc() when allocating memory on the host. This sets up page-locked memory, meaning that it cannot be swapped out by the OS' virtual memory system. This allows the use of DMA hardware to do the memory copy, said to make cudaMemcpy () twice as fast.

### 5.3.3.5 Other Types of Memory

There are also other types of memory. Again, let's start with a summary:

| type | registers | local | constant | texture |
| ---: | ---: | ---: | ---: | ---: |
| scope | single thread | single thread | glbl. to app. | glbl. to app. |
| location | device | device | host+device cache | host+device cache |
| speed | fast | molasses | fast if cache hit | fast if cache hit |
| lifetime | kernel | kernel | application | application |
| host access? | no | no | yes | yes |
| device access? | read/write | read/write | read | read |

## - Registers:

Each SM has a set of registers, much more numerous than in a CPU. Access to them is very fast, said to be slightly faster than to shared memory.
The compiler normally stores the local variables for a device function in registers, but there are exceptions. An array won't be placed in registers if the array is too large, or if the array has variable index values, such as

```
int z[20],i;
y = z[i];
```

Since registers are not indexable by the hardware, the compiler cannot allocate $\mathbf{z}$ to registers in this case. If on the other hand, the only code accessing $\mathbf{z}$ has constant indices, e.g. $\mathbf{z}[8]$, the compiler may put $\mathbf{z}$ in registers.

## - Local memory:

This is physically part of global memory, but is an area within that memory that is allocated by the compiler for a given thread. As such, it is slow, and accessible only by that thread. The compiler allocates this memory for local variables in a device function if the compiler cannot store them in registers. This is called register spill.

## - Constant memory:

As the name implies, it's read-only from the device (read/write by the host), for storing values that will not be changed by device code. It is off-chip, thus potentially slow, but has a cache on the chip. At present, the size is 64 K .
One designates this memory with __constant__, as a global variable in the source file. One sets its contents from the host via cudaMemcpyToSymbol(), whose (simple form for the) call is

```
cudaMemcpyToSymbol(var_name,pointer_to_source,number_bytes_copy,cudaMemcpyHostToDevice)
```

For example:

```
__constant__ int x; // not contained in any function
```

// host code
int $\mathrm{y}=3$;
cudaMemcpyToSymbol("x",\&y,sizeof(int));
// device code
int z;
$\mathrm{z}=\mathrm{x}$;

Note again that the name Constant refers to the fact that device code cannot change it. But host code certainly can change it between kernel calls. This might be useful in iterative algorithms like this:

```
/ host code
for 1 to number of iterations
    set Constant array x
    call kernel (do scatter op)
    cudaThreadSynchronize()
    do gather op, using kernel results to form new x
// device code
use x together with thread-specific data
return results to host
```


## - Texture:

This is similar to constant memory, in the sense that it is read-only and cached. The difference is that the caching is two-dimensional. The elements $\mathbf{a}[\mathbf{i}][\mathbf{j}]$ and $\mathbf{a}[\mathbf{i}+\mathbf{1}][\mathbf{j}]$ are far from each other in the global memory, but since they are "close" in a two-dimensional sense, they may reside in the same cache line.

### 5.3.4 Threads Hierarchy

Following the hardware, threads in CUDA software follow a hierarchy:

- The entirety of threads for an application is called a grid.
- A grid consists of one or more blocks of threads.
- Each block has its own ID within the grid, consisting of an "x coordinate" and a "y coordinate."
- Likewise each thread has $\mathrm{x}, \mathrm{y}$ and z coordinates within whichever block it belongs to.
- Just as an ordinary CPU thread needs to be able to sense its ID, e.g. by calling omp_get_thread_num() in OpenMP, CUDA threads need to do the same. A CUDA thread can access its block ID via the built-in variables blockIdx.x and blockIdx.y, and can access its thread ID within its block via threadIdx.x, threadIdx.y and threadIdx.z.
- The programmer specifies the grid size (the numbers of rows and columns of blocks within a grid) and the block size (numbers of rows, columns and layers of threads within a block). In the first example above, this was done by the code
dim3 dimGrid(n,1);
dim3 dimBlock $(1,1,1)$;
find1elt<<<dimGrid, dimBlock>>>(dm,drs,n);
Here the grid is specified to consist of $\mathrm{n}(n \times 1)$ blocks, and each block consists of just one $(1 \times 1 \times 1)$ thread.
That last line is of course the call to the kernel. As you can see, CUDA extends C syntax to allow specifying the grid and block sizes. CUDA will store this information in structs of type dim3, in this case our variables gridDim and blockDim, accessible to the programmer, again with member variables for the various dimensions, e.g. blockDim.x for the size of the X dimension for the number of threads per block.
- All threads in a block run in the same SM, though more than one block might be on the same SM.
- The "coordinates" of a block within the grid, and of a thread within a block, are merely abstractions. If for instance one is programming computation of heat flow across a twodimensional slab, the programmer may find it clearer to use two-dimensional IDs for the threads. But this does not correspond to any physical arrangement in the hardware.

As noted, the motivation for the two-dimensional block arrangment is to make coding conceptually simpler for the programmer if he/she is working an application that is two-dimensional in nature.

For example, in a matrix application one's parallel algorithm might be based on partitioning the matrix into rectangular submatrices (tiles), as we'll do in Section 11.2. In a small example there, the matrix

$$
A=\left(\begin{array}{ccc}
1 & 5 & 12  \tag{5.1}\\
0 & 3 & 6 \\
4 & 8 & 2
\end{array}\right)
$$

is partitioned as

$$
A=\left(\begin{array}{ll}
A_{00} & A_{01}  \tag{5.2}\\
A_{10} & A_{11}
\end{array}\right)
$$

where

$$
\begin{align*}
& A_{00}=\left(\begin{array}{ll}
1 & 5 \\
0 & 3
\end{array}\right),  \tag{5.3}\\
& A_{01}=\binom{12}{6},  \tag{5.4}\\
& A_{10}=\left(\begin{array}{ll}
4 & 8
\end{array}\right) \tag{5.5}
\end{align*}
$$

and

$$
\begin{equation*}
A_{11}=(2) . \tag{5.6}
\end{equation*}
$$

We might then have one block of threads handle $A_{00}$, another block handle $A_{01}$ and so on. CUDA's two-dimensional ID system for blocks makes life easier for programmers in such situations.

### 5.3.5 What's NOT There

We're not in Kansas anymore, Toto - character Dorothy Gale in The Wizard of Oz
It looks like C, it feels like C, and for the most part, it is C. But in many ways, it's quite different from what you're used to:

- You don't have access to the C library, e.g. printf() (the library consists of host machine language, after all). There are special versions of math functions, however, e.g. _-sin().
- No recursion.
- No stack. Functions are essentially inlined, rather than their calls being handled by pushes onto a stack.
- No pointers to functions.


### 5.4 Synchronization, Within and Between Blocks

As mentioned earlier, a barrier for the threads in the same block is available by calling __syncthreads(). Note carefully that if one thread writes a variable to shared memory and another then reads that variable, one must call this function (from both threads) in order to get the latest value. Keep in mind that within a block, different warps will run at different times, making synchronization vital.

Remember too that threads across blocks cannot sync with each other in this manner. There are, though, several atomic operations-read/modify/write actions that a thread can execute without pre-emption, i.e. without interruption-available on both global and shared memory. For example, atomicAdd() performs a fetch-and-add operation, as described in Section 3.4.3 of this book. The call is
atomicAdd(address of integer variable,inc);
where address of integer variable is the address of the (device) variable to add to, and inc is the amount to be added. The return value of the function is the value originally at that address before the operation.

There are also atomicExch() (exchange the two operands), atomicCAS() (if the first operand equals the second, replace the first by the third), atomicMin(), atomicMax(), atomicAnd(), atomicOr(), and so on.

Use -arch=sm_11 when compiling, e.g.
nvcc -g -G yoursrc.cu -arch=sm_11

Though a barrier could in principle be constructed from the atomic operations, its overhead would be quite high, possibly near a microsecond. This would not be not much faster than attaining interblock synchronization by returning to the host and calling cudaThreadSynchronize() there. Recall that the latter is a possible way to implement a barrier, since global memory stays intact in between kernel calls, but again, it would be slow.

So, what if synchronization is really needed? This is the case, for instance, for iterative algorithms, where all threads must wait at the end of each iteration.

If you have a small problem, maybe you can get satisfactory performance by using just one block. You'll have to use a larger granularity, i.e. more work assigned to each thread. But using just one block means you're using only one SM, thus only a fraction of the potential power of the machine.

If you use multiple blocks, though, your only feasible option for synchronization is to rely on returns to the host, where synchronization occurs via cudaThreadSynchronize(). You would then have the situation outlined in the discussion of Constant memory in Section 5.3.3.5.

### 5.5 More on the Blocks/Threads Tradeoff

Resource size considerations must be kept in mind when you design your code and your grid configuration. In particular, note the following:

- Each block in your code is assigned to some SM. It will be tied to that SM during the entire execution of your kernel, though of course it will not constantly be running during that time.
- Within a block, threads execute by the warp, 32 threads. At any give time, the SM is running one warp, chosen by the GPU OS.
- The GPU has a limit on the number of threads that can run on a single block, typically 512, and on the total number of threads running on an SM, 786.
- If a block contains fewer than 32 threads, only part of the processing power of the SM it's running on will be used. So block size should normally be at least 32 . Moreover, for the same reason, block size should ideally be a multiple of 32 .
- If your code makes used of shared memory, or does within-block synchronization, the larger the block size, the better.
- We want to use the full power of the GPU, with its many SMs, thus implying a need to use at least as many blocks as there are SMs (which may require smaller blocks).
- Moreover, due to the need for latency hiding in memory access, we want to have lots of warps, so that some will run while others are doing memory access.
- Two threads doing unrelated work, or the same work but with many if/elses, would cause a lot of thread divergence if they were in the same block.
- A commonly-cited rule of thumb is to have between 128 and 256 threads per block.

Though there is a limit on the number of blocks, this limit will be much larger than the number of SMs. So, you may have multiple blocks running on the same SM. Since execution is scheduled by the warp anyway, there appears to be no particular drawback to having more than one block on the same SM.

### 5.6 Hardware Requirements, Installation, Compilation, Debugging

You do need a suitable NVIDIA video card. There is a list at http://www.nvidia.com/object/ cuda_gpus.html. If you have a Linux system, run lspci to determine what kind you have.

Download the CUDA toolkit from NVIDIA. Just plug "CUDA download" into a Web search engine to find the site. Install as directed.

You'll need to set your search and library paths to include the CUDA bin and lib directories.
To compile x.cu (and yes, use the .cu suffix), type
\$ nvcc -g -G x.cu

The -g -G options are for setting up debugging, the first for host code, the second for device code. You may also need to specify
-I/your_CUDA_include_path
to pick up the file cuda.h. Run the code as you normally would.
You may need to take special action to set your library path properly. For example, on Linux machines, set the environment variable LD_LIBRARY_PATH to include the CUDA library.

To determine the limits, e.g. maximum number of threads, for your device, use code like this:

```
cudaDeviceProp Props;
cudaGetDeviceProperties(&Props,0);
```

The 0 is for device 0 , assuming you only have one device. The return value of cudaGetDeviceProperties() is a complex C struct whose components are listed at http://developer. download. nvidia.com/compute/cuda/2_3/toolkit/docs/online/group__CUDART__DEVICE_g5aa4f47938af8276f08074d html.

Here's a simple program to check some of the properties of device 0 :

```
#include<cuda.h>
#include <stdio.h>
int main()
{
    cudaDeviceProp Props;
    cudaGetDeviceProperties( &Props,0);
    printf("shared mem: %d)\n", Props.sharedMemPerBlock);
    printf("max threads/block: %d\n",Props.maxThreadsPerBlock);
    printf("max blocks: %d\n",Props.maxGridSize[0]);
    printf("total Const mem: %d\n",Props.totalConstMem);
}
```

Under older versions of CUDA, such as 2.3, one can debug using GDB as usual. You must compile your program in emulation mode, using the -deviceemu command-line option. This is no longer available as of version 3.2. CUDA also includes a special version of GDB, CUDA-GDB (invoked as cuda-gdb) for real-time debugging. However, on Unix-family platforms it runs only if X11 is not running. Short of dedicating a machine for debugging, you may find it useful to install a version 2.3 in addition to the most recent one to use for debugging.

### 5.7 Example: Improving the Row Sums Program

The issues involving coalescing in Section 5.3.3.2 would suggest that our rowsum code might run faster with column sums, to take advantage of the memory banking. (So the user would either need to take the transpose first, or have his code set up so that the matrix is in transpose form to begin with.) As two threads in the same half-warp march down adjoining columns in lockstep, they will always be accessing adjoining words in memory.

So, I modified the program accordingly (not shown), and compiled the two versions, as rs and cs, the row- and column-sum versions of the code, respectively.

This did produce a small improvement (confirmed in subsequent runs, needed in any timing experiment):

```
pc5:~/CUDA% time rs 20000
2.585u 1.753s 0:04.54 95.3% 0+0k 7104+0io 54pf+0w
pc5:~/CUDA% time cs }2000
2.518u 1.814s 0:04.40 98.1% 0+0k 536+0io 5pf+0w
```

But let's compare it to a version running only on the CPU,

```
#include <stdio.h>
#include <stdlib.h>
// non-CUDA example: finds col sums of an integer matrix m
// findlelt() finds the colsum of one col of the nxn matrix m, storing the
// result in the corresponding position in the colsum array cs; matrix
// stored as 1-dimensional, row-major order
void find1elt(int *m, int *cs, int n)
{
    int sum=0;
    int topofcol;
    int col,k;
    for (col = 0; col < n; col++) {
        topofcol = col;
```

```
        sum = 0;
        for (k = 0; k < n; k++)
            sum += m[topofcol+k*n];
        cs[col] = sum;
    }
}
int main(int argc, char **argv)
{
    int n = atoi(argv[1]); // number of matrix cols/cols
    int *hm, // host matrix
        *hcs; // host colsums
    int msize = n * n * sizeof(int); // size of matrix in bytes
    // allocate space for host matrix
    hm = (int *) malloc(msize);
    // as a test, fill matrix with consecutive integers
    int t = 0,i,j;
    for (i = 0; i < n; i++) {
        for (j = 0; j < n; j++) {
            hm[i*n+j] = t++;
        }
    }
    int cssize = n * sizeof(int);
    hcs = (int *) malloc(cssize);
    find1elt(hm, hcs,n);
    if (n< 10) for(i=0; i<n; i++) printf("%d\n",hcs[i]);
    // clean up
    free(hm);
    free(hcs);
}
```

How fast does this non-CUDA version run?
pc5:~/CUDA\% time csc 20000
$61.110 \mathrm{u} 1.719 \mathrm{~s} 1: 02.8699 .9 \% \quad 0+0 \mathrm{k} 0+0$ io $0 \mathrm{pf}+0 \mathrm{w}$

Very impressive! No wonder people talk of CUDA in terms like "a supercomputer on our desktop." And remember, this includes the time to copy the matrix from the host to the device (and to copy the output array back). And we didn't even try to optimize thread configuration, memory coalescing and bank usage, making good use of memory hierarchy, etc. ${ }^{2}$

On the other hand, remember that this is an "embarrassingly parallel" application, and in many applications we may have to settle for a much more modest increase, and work harder to get it.

[^32]
### 5.8 Example: Finding the Mean Number of Mutual Outlinks

As in Sections 2.4.3 and 4.12, consider a network graph of some kind, such as Web links. For any two vertices, say any two Web sites, we might be interested in mutual outlinks, i.e. outbound links that are common to two Web sites. The CUDA code below finds the mean number of mutual outlinks, among all pairs of sites in a set of Web sites.

```
#include <cuda.h>
#include <stdio.h>
// CUDA example: finds mean number of mutual outlinks, among all pairs
// of Web sites in our set; in checking all (i,j) pairs, thread k will
// handle all i such that i mod totth = k, where totth is the number of
// threads
// procpairs() processes all pairs for a given thread
__global__ void procpairs(int *m, int *tot, int n)
{ int totth = gridDim.x * blockDim.x, // total number of threads
            me = blockIdx.x * blockDim.x + threadIdx.x; // my thread number
        int i,j,k,sum = 0;
        for (i = me; i < n; i += totth) { // do various rows i
            for (j = i +1; j < n; j++) { // do all rows j > i
                for (k = 0; k < n; k++)
                    sum +=m[n*i+k] * m[n*j+k];
            }
    }
    atomicAdd(tot,sum);
}
int main(int argc, char **argv)
{ int n = atoi(argv[1]), // number of vertices
            nblk = atoi(argv[2]); // number of blocks
        int *hm, // host matrix
            *dm, // device matrix
            htot, // host grand total
            *dtot; // device grand total
    int msize = n * n * sizeof(int); // size of matrix in bytes
    // allocate space for host matrix
    hm = (int *) malloc(msize);
    // as a test, fill matrix with random 1s and 0s
    int i,j;
    for (i = 0; i < n; i++) {
        hm[n*i+i] = 0;
        for (j = 0; j < n; j++) {
            if (j != i) hm[i*n+j] = rand() % 2;
        }
    }
    // allocate space for device matrix
    cudaMalloc ((void **)&dm, msize );
```

```
// copy host matrix to device matrix
cudaMemcpy(dm,hm, msize , cudaMemcpyHostToDevice);
htot = 0;
// set up device total and initialize it
cudaMalloc((void **)&dtot, sizeof(int));
cudaMemcpy(dtot,&htot, sizeof(int), cudaMemcpyHostToDevice);
// set up parameters for threads structure
dim3 dimGrid(nblk,1);
dim3 dimBlock (192,1,1);
// invoke the kernel
procpairs<<<<dimGrid, dimBlock}>>>>(dm, dtot,n)
// wait for kernel to finish
cudaThreadSynchronize();
// copy total from device to host
cudaMemcpy(&htot, dtot, sizeof(int),cudaMemcpyDeviceToHost);
// check results
if (n<= 15) {
        for (i = 0; i < n; i++) {
        for (j = 0; j < n; j++)
            printf("%d ",hm[n*i+j]);
        printf("\n");
    }
}
printf("mean = %f\n",htot/float ((n*(n-1))/2));
// clean up
free(hm);
cudaFree(dm);
cudaFree(dtot);
```

\}

Again we've used the method in Section 2.4.3 to partition the various pairs (i, j ) to the different threads. Note the use of atomicAdd().

The above code is hardly optimal. The reader is encouraged to find improvements.

### 5.9 Example: Finding Prime Numbers

The code below finds all the prime numbers from 2 to $\mathbf{n}$.

```
#include <stdio.h>
#include <stdlib.h>
#include <cuda.h>
// CUDA example: illustration of shared memory allocation at run time;
// finds primes using classical Sieve of Erathosthenes: make list of
// numbers 2 to n, then cross out all multiples of 2 (but not 2 itself),
// then all multiples of 3, etc.; whatever is left over is prime; in our
```

```
// array, 1 will mean "not crossed out" and 0 will mean "crossed out"
// IMPORTANT NOTE: uses shared memory, in a single block, without
// rotating parts of array in and out of shared memory; thus limited to
// n <= 4000 if have 16K shared memory
// initialize sprimes, 1s for the odds, 0s for the evens; see sieve()
// for the nature of the arguments
_-device_- void initsp(int *sprimes, int n, int nth, int me)
{
    int chunk, startsetsp,endsetsp,val,i;
    sprimes[2] = 1;
    // determine sprimes chunk for this thread to init
    chunk = (n-1) / nth;
    startsetsp = 2 + me*chunk;
    if (me < nth-1) endsetsp = startsetsp + chunk - 1;
    else endsetsp = n;
    // now do the init
    val = startsetsp % 2;
    for (i = startsetsp; i <= endsetsp; i++) {
        sprimes[i] = val;
        val = 1 - val;
    }
    // make sure sprimes up to date for all
    _-syncthreads();
}
// copy sprimes back to device global memory; see sieve() for the nature
// of the arguments
__device__ void cpytoglb(int *dprimes, int *sprimes, int n, int nth, int me)
{
    int startcpy, endcpy, chunk,i;
    chunk = (n-1) / nth;
    startcpy = 2 + me*chunk;
    if (me < nth -1) endcpy = startcpy + chunk - 1;
    else endcpy = n;
    for (i = startcpy; i <= endcpy; i++) dprimes[i] = sprimes[i];
    _-syncthreads();
}
// finds primes from 2 to n, storing the information in dprimes, with
// dprimes[i] being 1 if i is prime, 0 if composite; nth is the number
// of threads (threadDim somehow not recognized)
__global__ void sieve(int *dprimes, int n, int nth)
{
    extern _-shared_- int sprimes[];
    int me = threadIdx.x;
    int nth1 = nth - 1;
    // initialize sprimes array, 1s for odds, 0 for evens
    initsp(sprimes,n,nth,me);
```

```
    // "cross out" multiples of various numbers m, with each thread doing
    // a chunk of m's; always check first to determine whether m has
    // already been found to be composite; finish when m*m > n
    int maxmult,m,startmult,endmult,chunk,i;
    for (m = 3; m*m <= n; m++) {
        if (sprimes[m] != 0) {
            // find largest multiple of m that is <= n
            maxmult = n / m;
            // now partition 2,3,\ldots,maxmult among the threads
            chunk = (maxmult - 1) / nth;
            startmult = 2 + me*chunk;
            if (me < nth1) endmult = startmult + chunk - 1;
            else endmult = maxmult;
        }
        // OK, cross out my chunk
        for (i = startmult; i <= endmult; i++) sprimes[i*m] = 0;
    }
    _-syncthreads();
    // copy back to device global memory for return to host
    cpytoglb(dprimes,sprimes,n,nth,me);
}
int main(int argc, char **argv)
{
    int n = atoi(argv[1]), // will find primes among 1,\ldots.,n
        nth = atoi(argv[2]); // number of threads
    int *hprimes, // host primes list
        *dprimes; // device primes list
    int psize = (n+1) * sizeof(int); // size of primes lists in bytes
    // allocate space for host list
    hprimes = (int *) malloc(psize);
    // allocate space for device list
    cudaMalloc((void **)&dprimes, psize);
    dim3 dimGrid (1,1);
    dim3 dimBlock(nth,1,1);
    // invoke the kernel, including a request to allocate shared memory
    sieve<<<<dimGrid, dimBlock, psize>>>(dprimes,n,nth );
    // check whether we asked for too much shared memory
    cudaError_t err = cudaGetLastError();
    if(err != cudaSuccess) printf("%s\n",cudaGetErrorString(err));
    // wait for kernel to finish
    cudaThreadSynchronize();
    // copy list from device to host
    cudaMemcpy(hprimes, dprimes, psize,cudaMemcpyDeviceToHost);
    // check results
    if (n<= 1000) for(int i=2; i<=n; i++)
        if (hprimes[i] = 1) printf("%d\n",i);
    // clean up
    free(hprimes);
    cudaFree(dprimes);
```

This code has been designed with some thought as to memory speed and thread divergence. Ideally, we would like to use device shared memory if possible, and to exploit the lockstep, SIMD nature of the hardware.

The code uses the classical Sieve of Erathosthenes, "crossing out" multiples of 2, 3, 5, 7 and so on to get rid of all the composite numbers. However, the code here differs from that in Section 1.3.2.1, even though both programs use the Sieve of Erathosthenes.

Say we have just two threads, A and B. In the earlier version, thread A might cross out all multiples of 19 while B handles multiples of 23 . In this new version, thread A deals with only some multiples of 19 and B handles the others for 19. Then they both handle their own portions of multiples of 23 , and so on. The thinking here is that the second version will be more amenable to lockstep execution, thus causing less thread divergence.

Thus in this new version, each thread handles a chunk of multiples of the given prime. Note the contrast of this with many CUDA examples, in which each thread does only a small amount of work, such as computing a single element in the product of two matrices.

In order to enhance memory performance, this code uses device shared memory. All the "crossing out" is done in the shared memory array sprimes, and then when we are all done, that is copied to the device global memory array dprimes, which is in turn copies to host memory. By the way, note that the amount of shared memory here is determined dynamically.

However, device shared memory consists only of 16 K bytes, which would limit us here to values of n up to about 4000. Moreover, by using just one block, we are only using a small part of the CPU. Extending the program to work for larger values of $\mathbf{n}$ would require some careful planning if we still wish to use shared memory.

### 5.10 Example: Finding Cumulative Sums

Here we wish to compute cumulative sums. For instance, if the original array is (3,1,2,0,3,0,1,2), then it is changed to $(3,4,6,6,9,9,10,12)$.
(Note: This is a special case of the prefix scan problem, covered in Chapter ??.)
The general plan is for each thread to operate on one chunk of the array. A thread will find cumulative sums for its chunk, and then adjust them based on the high values of the chunks that precede it. In the above example, for instance, say we have 4 threads. The threads will first produce $(3,4),(2,2),(3,3)$ and $(1,3)$. Since thread 0 found a cumulative sum of 4 in the end, we must add 4 to each element of $(2,2)$, yielding $(6,6)$. Thread 1 had found a cumulative sum of 2 in the end,
which together with the 4 found by thread 0 makes 6 . Thus thread 2 must add 6 to each of its elements, i.e. add 6 to $(3,3)$, yielding $(9,9)$. The case of thread 3 is similar.

Below is code for the special case of a single block:

```
// for this simple illustration, it is assumed that the code runs in
// just one block, and that the number of threads evenly divides n
#include <cuda.h>
#include <stdio.h>
__global__ void cumulker(int *dx, int n)
{
    int me = threadIdx.x;
    int csize = n / blockDim.x;
    int start = me * csize;
    int i,j,base;
    for (i = 1; i < csize; i++) {
        j = start + i;
        dx[j] = dx[j-1] + dx[j];
    }
    __syncthreads();
    if (me > 0) {
        base = 0;
        for (j = 0; j < me; j++)
            base += dx[(j+1)* csize - 1];
    }
    __syncthreads();
    if (me> 0) {
        for (i = start; i < start + csize; i++)
            dx[i] += base;
    }
}
```


### 5.11 Example: Transforming an Adjacency Matrix

Here is a CUDA version of the code in Section 4.13.

```
// CUDA example
// takes a graph adjacency matrix for a directed graph, and converts it
// to a 2-column matrix of pairs (i,j), meaning an edge from vertex i to
// vertex j; the output matrix must be in lexicographical order
// not claimed efficient, either in speed or in memory usage
#include <cuda.h>
#include <stdio.h>
```

```
// needs -lrt link flag for C++
#include <time.h>
float timediff(struct timespec t1, struct timespec t2)
{ if (t1.tv_nsec > t2.tv_nsec) {
    t2.tv_sec -= 1;
    t2.tv_nsec += 1000000000;
    }
    return t2.tv_sec-t1.tv_sec + 0.000000001 * (t2.tv_nsec-t1.tv_nsec);
}
// kernel transgraph() does this work
// arguments:
// adjm: the adjacency matrix (NOT assumed symmetric), 1 for edge, 0
// otherwise; note: matrix is overwritten by the function
// n: number of rows and columns of adjm
// adjmout: output matrix
// nout: number of rows in adjmout
--global_- void tgkernel1(int *dadjm, int n, int *dcounts)
{ int tot1s,j;
    int me = blockDim.x * blockIdx.x + threadIdx.x;
    tot1s=0;
    for ( j = 0; j < n; j++) {
            if (dadjm[n*me+j] = 1) {
            dadjm[n*me+tot1s++] = j;
        }
    dcounts[me] = tot1s;
    }
}
_-global_- void tgkernel2(int *dadjm, int n,
    int *dcounts, int *dstarts, int *doutm)
{ int outrow, num1si,j;
    // int me = threadIdx.x;
    int me = blockDim.x * blockIdx.x + threadIdx.x;
    // fill in this thread's portion of doutm
    outrow = dstarts[me];
    num1si = dcounts[me];
    if (num1si > 0) {
            for (j = 0; j < num1si; j++) {
            doutm [2* outrow +2*j] = me;
            doutm}[2*\mathrm{ outrow }+2*\textrm{j}+1]=\operatorname{dadjm}[\textrm{n}*\textrm{me}+\textrm{j}]
        }
    }
}
// replaces counts by cumulative counts
void cumulcounts(int *c, int *s, int n)
```

```
{ int i;
    s[0] = 0;
    for (i = 1; i < n; i++) {
        s[i] = s[i-1] + c[i-1];
    }
}
int *transgraph(int *hadjm, int n, int * nout, int gsize, int bsize)
{ int *dadjm; // device adjacency matrix
    int *houtm; // host output matrix
    int *doutm; // device output matrix
    int *hcounts; // host counts vector
    int *dcounts; // device counts vector
    int *hstarts; // host starts vector
    int *dstarts; // device starts vector
    hcounts = (int *) malloc(n*sizeof(int));
    hstarts = (int *) malloc(n*sizeof(int));
    cudaMalloc((void **)&dadjm,n*n*sizeof(int));
    cudaMalloc((void **)&dcounts, n*sizeof(int));
    cudaMalloc((void **)&dstarts,n*sizeof(int));
    houtm = (int *) malloc(n*n*sizeof(int));
    cudaMalloc((void **)&doutm, n*n*sizeof(int));
    cudaMemcpy(dadjm, hadjm, n*n*sizeof(int), cudaMemcpyHostToDevice);
    dim3 dimGrid(gsize, 1);
    dim3 dimBlock(bsize, 1,1);
    // calculate counts and starts first
    tgkernel1<<<<dimGrid, dimBlock>>>(dadjm,n, dcounts );
    // cudaMemcpy(hadjm, dadjm,n*n*sizeof(int),cudaMemcpyDeviceToHost);
    cudaMemcpy(hcounts, dcounts, n*sizeof(int), cudaMemcpyDeviceToHost);
    cumulcounts(hcounts, hstarts,n);
    *nout = hstarts[n-1] + hcounts[n-1];
    cudaMemcpy(dstarts, hstarts, n*sizeof(int), cudaMemcpyHostToDevice);
    tgkernel2<<<dimGrid, dimBlock>>>(dadjm,n, dcounts, dstarts, doutm);
    cudaMemcpy(houtm, doutm, 2*(* nout)*sizeof(int), cudaMemcpyDeviceToHost);
    free(hcounts);
    free(hstarts);
    cudaFree(dadjm);
    cudaFree(dcounts);
    cudaFree(dstarts);
    return houtm;
}
int main(int argc, char **argv)
{ int i,j;
    int *adjm; // host adjacency matrix
    int *outm; // host output matrix
    int n = atoi(argv[1]);
    int gsize = atoi(argv[2]);
    int bsize = atoi(argv[3]);
    int nout;
```

```
    adjm =(int *) malloc(n*n*sizeof(int));
    for (i = 0; i < n; i++)
        for (j = 0; j < n; j++)
            if (i= j) adjm[n*i+j] = 0;
            else adjm[n*i+j] = rand() % 2;
    if (n< 10) {
        printf("adjacency matrix: \n");
        for (i = 0; i < n; i++) {
            for (j = 0; j< n; j++) printf("%d ",adjm[n*i+j]);
            printf("\n");
        }
    }
    struct timespec bgn,nd;
    clock_gettime(CLOCK_REALTIME, &bgn);
    outm = transgraph(adjm,n,&nout,gsize,bsize);
    printf("num rows in out matrix = %d\n", nout);
    if (nout < 50) {
        printf("out matrix: \n");
        for (i = 0; i < nout; i++)
            printf("%d %d\n",outm[2*i],outm[2*i+1]);
    }
    clock_gettime(CLOCK_REALTIME, &nd);
    printf("%f\n",timediff(bgn, nd ));
```

\}

### 5.12 Error Checking

Every CUDA call (except for kernel invocations) returns an error code of type cudaError_t. One can view the nature of the error by calling cudaGetErrorString() and printing its output.

For kernel invocations, one can call cudaGetLastError(), which does what its name implies. A call would typically have the form

```
cudaError_t err = cudaGetLastError();
if(err != cudaSuccess) printf("%s\n",cudaGetErrorString(err));
```

You may also wish to cutilSafeCall(), which is used by wrapping your regular CUDA call. It automatically prints out error messages as above.

Each CUBLAS call returns a potential error code, of type cublasStatus, not checked here.

### 5.13 Loop Unrolling

Loop unrolling is an old technique used on uniprocessor machines to achieve speedup due to branch elimination and the like. Branches make it difficult to do instruction or data prefetching, so eliminating them may speed things up.

The CUDA compiler provides the programmer with the unroll pragma to request loop unrolling. Here an n-iteration for loop is changed to k copies of the body of the loop, each working on about $\mathrm{n} / \mathrm{k}$ iterations. If n and k are known constant, GPU registers can be used to implement the unrolled loop.

For example, the loop

```
for (i = 0; i < 2; i++ {
    sum += x[i];
    sum2 += x[i]*x[i];
}
```

could be unrolled to

```
sum += x[1];
sum2 += x[1]*x[1];
sum += x[2];
sum2 += x[2]*x[2];
```

Here $\mathrm{n}=\mathrm{k}=2$. If x is local to this function, then unrolling will allow the compiler to store it in a register, which could be a great performance enhancer.

The compiler will try to do loop unrolling even if the programmer doesn't request it, but the programmer can try to control things by using the pragma:
\#pragma unroll k
suggest to the compiler a k -fold unrolling. Setting $\mathrm{k}=1$ will instruct the compiler not to unroll.

### 5.14 Short Vectors

In CUDA, there are types such as int4, char2 and so on, up to four elements each. So, an uint4 type is a set of four unsigned ints. These are called short vectors.

The key point is that a short vector can be treated as a single word in terms of memory access and GPU instructions. It may be possible to reduce time by a factor of 4 by dividing arrays into chunks of four contiguous words and making short vectors from them.

### 5.15 The New Generation

The latest GPU architecture from NVIDIA is called Fermi ${ }^{3}$ Many of the advances are of the "bigger and faster than before" type. These are important, but be sure to note the significant architectural changes, including:

- Host memory, device global memory and device shared memory share a unifed address space.
- On-chip memory can be apportioned to both shared memory and cache memory. Since shared memory is in essence a programmer-managed cache, this gives the programmer access to a real cache. (Note, however, that this cache is aimed at spatial locality, not temporal locality.)


### 5.16 CUDA from a Higher Level

CUDA programming can involve a lot of work, and one is never sure that one's code is fully efficient. Fortunately, a number of libraries of tight code have been developed for operations that arise often in parallel programming.

You are of course using CUDA code at the bottom, but without explicit kernel calls. And again, remember, the contents of device global memory are persistent across kernel calls in the same application. Therefore you can mix explicit CUDA code and calls to these libraries. Your program might have multiple kernel invocations, some CUDA and others to the libraries, with each using data in device global memory that was written by earlier kernels. In some cases, you may need to do a conversion to get the proper type.

These packages can be deceptively simple. Remember, each call to a function in these packages involves a CUDA kernel call-with the associated overhead.

Programming in these libraries is typically much more convenient than in direct CUDA. Note, though, that even though these libraries have been highly optimized for what they are intended to do, they will not generally give you the fastest possible code for any given CUDA application.

We'll discuss a few such libraries in this section.

### 5.16.1 CUBLAS

CUDA includes some parallel linear algebra routines callable from straight C code. In other words, you can get the benefit of GPU in linear algebra contexts without directly programming in CUDA.

[^33]
### 5.16.1.1 Example: Row Sums Once Again

Below is an example RowSumsCB.c, the matrix row sums example again, this time using CUBLAS. We can find the vector of row sums of the matrix A by post-multiplying A by a column vector of all 1 s .

I compiled the code by typing

```
gcc -g -I/usr/local/cuda/include -L/usr/local/cuda/lib RowSumsCB.c -lcublas -lcudart
```

You should modify for your own CUDA locations accordingly. Users who merely wish to use CUBLAS will find the above more convenient, but if you are mixing CUDA and CUBLAS, you would use nvcc:

```
nvcc -g -G RowSumsCB.c -lcublas
```

Here is the code:

```
#include <stdio.h>
#include <cublas.h> // required include
int main(int argc, char **argv)
{
    int n = atoi(argv[1]); // number of matrix rows/cols
        float *hm, // host matrix
            *hrs, // host rowsums vector
            *ones, // 1s vector for multiply
            *dm, // device matrix
            *drs; // device rowsums vector
        // allocate space on host
        hm = (float *) malloc(n*n*sizeof(float));
        hrs = (float *) malloc(n*sizeof(float));
        ones =(float *) malloc(n*sizeof(float));
        // as a test, fill hm with consecutive integers, but in column-major
        // order for CUBLAS; also put 1s in ones
        int i,j;
        float t = 0.0;
        for (i = 0; i < n; i++) {
            ones[i] = 1.0;
            for (j = 0; j < n; j++)
            hm[j*n+i] = t++;
    }
    cublasInit(); // required init
    // set up space on the device
    cublasAlloc(n*n, sizeof(float),(void**)&dm);
    cublasAlloc(n,sizeof(float),(void**)&drs);
```

```
    // copy data from host to device
    cublasSetMatrix(n,n, sizeof(float),hm,n,dm,n);
    cublasSetVector(n, sizeof(float),ones, 1, drs,1);
    // matrix times vector ("mv")
    cublasSgemv('n',n,n,1.0,dm,n, drs,1,0.0, drs,1);
    // copy result back to host
    cublasGetVector(n, sizeof(float), drs,1,hrs,1);
    // check results
    if (n<20) for (i = 0; i < n; i++) printf("%f\n",hrs[i]);
    // clean up on device (should call free() on host too)
    cublasFree (dm);
    cublasFree(drs);
    cublasShutdown();
```

\}

As noted in the comments, CUBLAS assumes FORTRAN-style, i.e. column-major order, for matrices.

Now that you know the basic format of CUDA calls, the CUBLAS versions will look similar. In the call
cublasAlloc ( $n * n$, sizeof (float), (void**) \&dm);
for instance, we are allocating space on the device for an $\mathrm{n} \times \mathrm{n}$ matrix of floats.
The call
cublasSetMatrix(n,n,sizeof(float), hm, n, dm, $n$ );
is slightly more complicated. Here we are saying that we are copying $\mathbf{h m}$, an n x n matrix of floats on the host, to $\mathbf{d m}$ on the host. The $\mathbf{n}$ arguments in the last and third-to-last positions again say that the two matrices each have $\mathbf{n}$ dimensioned rows. This seems redundant, but this is needed in cases of matrix tiling, where the number of rows of a tile would be less than the number of rows of the matrix as a whole.

The 1s in the call
cublasSetVector(n, sizeof(float), ones, $1, \mathrm{drs}, 1$ );
are needed for similar reasons. We are saying that in our source vector ones, for example, the elements of interest are spaced 1 elements apart, i.e. they are contiguous. But if we wanted our vector to be some row in a matrix with, say, 500 rows, the elements of any particular row of interest would be spaced 500 elements apart, again keeping in mind that column-major order is assumed.

The actual matrix multiplication is done here:

```
cublasSgemv('n',n,n,1.0,dm,n,drs,1,0.0,drs,1);
```

The "mv" in "cublasSgemv" stands for "matrix times vector." Here the call says: no ('n'), we do not want the matrix to be transposed; the matrix has $\mathbf{n}$ rows and $\mathbf{n}$ columns; we wish the matrix to be multiplied by 1.0 (if 0 , the multiplication is not actually performed, which we could have here); the matrix is at $\mathbf{d m}$; the number of dimensioned rows of the matrix is $\mathbf{n}$; the vector is at drs; the elements of the vector are spaced 1 word apart; we wish the vector to not be multiplied by a scalar (see note above); the resulting vector will be stored at drs, 1 word apart.

Further information is available in the CUBLAS manual.

### 5.16.2 Thrust

The Thrust library is usable not only with CUDA but also to general OpenMP code! So I've put my coverage of Thrust in a separate chapter, Chapter 6.

### 5.16.3 CUDPP

CUDPP is similar to Thrust (though CUDPP was developed earlier) in terms of operations offered. It is perhaps less flexible than Thrust, but is easier to learn and is said to be faster.
(No examples yet, as the author did not have access to a CUDPP system yet.)

### 5.16.4 CUFFT

CUFFT does for the Fast Fourier Transform what CUBLAS does for linear algebra, i.e. it provides CUDA-optimized FFT routines.

### 5.17 Other CUDA Examples in This Book

There are additional CUDA examples in later sections of this book. These include $\sqrt{4}^{4}$

- Prof. Richard Edgar's matrix-multiply code, optimized for use of shared memory, Section 11.3.2.2.

[^34]- Odd/even transposition sort, Section 12.3.3, showing a typical CUDA pattern for iterative algorithms.
- Gaussian elimination for linear systems, Section 11.5.1.


## Chapter 6

## Introduction to Thrust Programming

In the spirit of CUBLAS (Section 5.16.1) and other packages, the CUDA people have brought in another package to ease CUDA programming, Thrust. It uses the C++STL library as a model, and Thrust is indeed a C++ template library. It includes various data manipulation routines, such as for sorting and prefix scan operations.

### 6.1 Compiling Thrust Code

Thrust allows the programmer a choice of back ends, i.e. platforms on which the executable code will run. In addition to the CUDA back end, for running on the GPU, one can also choose OpenMP as the back end. The latter choice allows the high-level expressive power of Thrust to be used on multicore machines. A third choice is Intel's TBB language.

### 6.1.1 Compiling to CUDA

If your CUDA version is at least 4.0, then Thrust is included, which will be assumed here. In that case, you compile Thrust code with nvcc, no special link commands needed.

### 6.1.2 Compiling to OpenMP

You can use Thrust to generate OpenMP code. The Thrust "include" files work without having a GPU. Here for instance is how you would compile the first example program below ${ }^{1}$

[^35]```
g++ -g -O2 -o unqcount unqcount.cpp -fopenmp -lgomp \
    -DTHRUST_DEVICE_BACKEND=THRUST_DEVICE_BACKEND_OMP \
    -I/usr/home/matloff/Tmp/tmp1
```

I had no CUDA-capable GPU on this machine, but put the Thrust include directory tree in /usr/home/matloff/Tmp/tmp1, and then compiled as with any other include file.

The result is real OpenMP code. ${ }^{2}$ Everywhere you set up a Thrust vector, you'll be using OpenMP, i.e. the threads set up by Thrust will be OpenMP threads on the CPU rather than CUDA threads on the GPU 3 You set the number of threads as you do with any OpenMP program, e.g. with the environment variable OMP_NUM_THREADS.

### 6.2 Example: Counting the Number of Unique Values in an Array

As our first example, suppose we wish to determine the number of distinct values in an integer array. The following code may not be too efficient, but as an introduction to Thrust fundamental building blocks, we'll take the following approach:
(a) sort the array
(b) compare the array to a shifted version of itself, so that changes from one distinct element to another can be detected, producing an array of 1 s (change) and 0 s (no change)
(c) count the number of 1 s

Here's the code:

```
// various Thrust includes
#include <thrust/host_vector.h>
#include <thrust/device_vector.h>
#include <thrust/generate.h>
#include <thrust/sort.h>
#include <thrust/copy.h>
#include <thrust/count.h>
#include <cstdlib>
int rand16() // generate random integers mod 16
{ return rand() % 16; }
// C++ functor, to be called from thrust::transform(); compares
// corresponding elements of the arrays x and y, yielding 0 when they
```

[^36]```
// match, 1 when they don't
struct finddiff
{
    __device__ int operator()(const int& x, const int&y)
    { return x = y?0:1; }
};
int main(void)
{
    // generate test data, 1000 random numbers, on the host, int type
    thrust::host_vector <int> hv(1000);
    thrust::generate(hv.begin(), hv.end(), rand16);
    // copy data to the device, creating a vector there
    thrust::device_vector <int> dv = hv;
    // sort data on the device
    thrust::sort(dv.begin(), dv.end());
    // create device vector to hold differences, with length 1 less than
    // dv's length
    thrust::device_vector <int> diffs(dv.size()-1);
    // find the diffs; note that the syntax is finddiff(), not finddiff
    thrust::transform(dv.begin(), dv.end() - 1,
        dv.begin()+1, diffs.begin(), finddiff());
    // count the 1s, by removing 0s and checking new length
    // (or could use thrust::count())
    int ndiffs = thrust::reduce(diffs.begin(), diffs.end(), (int) 0,
        thrust:: plus<int > ());
    printf("# distinct: %d\n",ndiffs +1);
    // we've achieved our goal, but let's do a little more
    // transfer data back to host
    thrust::copy(dv.begin(), dv.end(), hv.begin());
    printf("the sorted array:\n");
    for (int i = 0; i < 1000; i++) printf("%d\n",hv[i]);
    return 0;
}
```

After generating some random data on a host array hv, we copy it to the device, creating a vector $\mathbf{d v}$ there. This code is certainly much simpler to write than slogging through calls to cudaMalloc() and cudaMemcpy ()!

The heart of the code is the call to thrust::transform(), which is used to implement step (b) in our outline above. It performs a "map" operation as in functional programming, taking one or two
arrays (the latter is the case here) as input, and outputting an array of the same size.
This example, as is typical in Thrust code, defines a functor. Well, what is a functor? This is a C++ mechanism to produce a callable function, laregly similar in goal to using a pointer to a function. In the context above, we are turning a C++ struct into a callable function, and we can do so with classes too. Since structs and classes can have member variables, we can store needed data in them, and that is what distinguishes functors from function pointers.

The transform function does elementwise operation-it calls the functor on each corresponding pair of elements from the two input arguments (0th element with 0th element, 1st with 1st, etc.), placing the results in the output array. So we must thus design our functor to do the map operation. In this case, we want to compare successive elements of our array (after sorting it), so we must find a way to do this through some element-by-element operation. The solution is to do elementwise comparison of the array and its shifted version. The call is

```
thrust::transform(dv.begin(), dv.end() - 1,
    dv.begin()+1,diffs.begin(), finddiff());
```

Note the parentheses in "finddiff()." This is basically a constructor, creating an instance of a finddiff object and returning a pointer to it. By contrast, in the code

```
thrust::generate(hv.begin(), hv.end(), rand16);
```

rand16() is an ordinary function, not a functor, so we just write its name here, thus passing a pointer to the function.

In the code

```
_-device_- int operator()(const int& x, const int&y)
{ return x = y?0:1; }
```

the C++ keyword operator says we are defining a function, which in this case has two int inputs and an int output. We stated earlier that functors are callable structs, and this is what gets called.

Thrust vectors have built-in member functions begin() and end(), that specify the start and the place 1 element past the end of the array. Note that we didn't actually create our shifted array in

```
thrust::transform(dv.begin(), dv.end() - 1,
    dv.begin()+1, diffs.begin(), finddiff());
```

Instead, we specified "the array beginning 1 element past the start of dv."
The "places" returned by calling begin() and end() above are formally called iterators, and work in a manner similar to pointers. Note again that end() returns a pointer to the location just after the last element of the array. The pointers are of type thrust::device_vector<int>::iterator here, with similar expressions for cases other than int type.

The transform function, in this case the comparison operation, will be done in parallel, on the GPU or other backend, as was the sorting. All that's left is to count the 1 s . We want to do that in parallel too, and Thrust provides another functional programming operation, reduction (as in OpenMP). We specify Thrust's built-in addition function (we could have defined our own if it were a more complex situation) as the operation, and 0 as the initial value:

```
int ndiffs = thrust::reduce(diffs.begin(), diffs.end(), (int) 0, thrust:: plus<int >());
```

We also could have used Thrust's thrust::count() function for further convenience.
Below is a shorter version of our unique-values-counter program, using thrust::unique(). Note that that function only removes consecutive duplicates, so the preliminary sort is still needed.

```
// various Thrust includes
#include <thrust/host_vector.h>
#include <thrust/device_vector.h>
#include <thrust/generate.h>
#include <thrust/sort.h>
#include <thrust/copy.h>
#include <thrust/unique.h>
#include <cstdlib>
int rand16()
{ return rand() % 16; }
int main(void)
{
    thrust::host_vector<int> hv(1000);
    thrust::generate(hv.begin(), hv.end(), rand16);
    thrust::device_vector<int> dv = hv;
    thrust::sort(dv.begin(), dv.end());
    thrust::device_vector<int >::iterator newend =
        thrust:: unique(dv.begin(), dv.end ());
    printf("# distinct: %d\n",newend - dv.begin());
    return 0;
}
```

Note the line

```
thrust::device_vector<int >::iterator newend =
    thrust:: unique(dv.begin(),dv.end ());
```

The unique() function returns an iterator pointing to (one element past) the end of the result of applying the unique-ifying operation. We can then "subtract" iterator values to get our desired count:

```
printf("# distinct: %d\n",newend - dv.begin());
```


### 6.3 Example: A Plain-C Wrapper for Thrust sort()

We may wish to wrap utility Thrust code in a function callable from a purely C/C++ program. The code below does that for the Thrust sort function.

```
// definitely needed
extern "C" void tsort(int *x, int *nx);
#include <thrust/device_vector.h>
#include <thrust/sort.h>
// nx set up as pointer so can call from R
void tsort(int *x, int *nx)
{ int n = *nx;
    // set up device vector and copy x to it
    thrust:: device_vector <int> dx (x, x+n);
    // sort, then copy back to x
    thrust:: sort(dx.begin(), dx.end());
    thrust::copy(dx.begin(), dx.end(), x);
}
```

To compile in the CUDA case, run nvcc -c and then run gcc (or whatever) as usual, making sure to link with the CUDA library. For instance,

```
nvcc -c SortForC.cu
gcc Main.c S*o -L/usr/local/cuda/lib -lcudart
```

Here's an example:

```
// TestSort.cpp: interface to Thrust sort from non-CUDA callers
#include <stdio.h>
extern "C" void tsort(int *x, int *nx);
// test
int main()
{ int x[5] = {12,13,5,8,88};
    int n=5,*nx; nx = &n;
    int i;
    tsort(x,nx);
    for (i = 0; i < 5; i++) printf("%d\n",x[i]);
}
```

Compile and run for the OpenMP case:

```
g++ -g -mcmodel=medium -fopenmp -lgomp \
    -DTHRUST_DEVICE_BACKEND=THRUST_DEVICE_BACKEND_OMP \
    -I/home/matloff/Thrust SortForC.cpp -o sort.o -c
```

```
g++ -g -mcmodel=medium TestSort.cpp sort.o -fopenmp -lgomp
% a.out
5
8
12
13
88
```

And for the CUDA case:

```
nvcc SortForC.cu -c -o sfc.o
gcc -g TestSort.c sfc.o -L/usr/local/cuda/lib -lcudart
% a.out
5
8
12
13
88
```


### 6.4 Example: Calculating Percentiles in an Array

One of the most useful types of Thrust operations is that provided by conditional functions. For instance, copy_if() acts as a filter, copying from an array only those elements that satisfy a predicate. In the example below, for instance, we can copy every third element of an array, or every eighth etc.

```
// illustration of copy_if()
// find every k-th element in given array, going from smallest to
// largest; k obtained from command line and fed into ismultk() functor
// these are the the ik/n * 100 percentiles, i = 1, 2, ...
#include <stdio.h>
#include <thrust/device_vector.h>
#include <thrust/sort.h>
#include <thrust/sequence.h>
#include <thrust/remove.h> // for copy_if() but not copy_if.h
// functor
struct ismultk {
    const int increm; // k in above comments
    // get k from call
    ismultk(int _increm): increm(_increm) {}
    __device__
    bool operator()(const int i)
```

```
    { return i != 0 && (i % increm) = 0;
    }
};
// test
int main(int argc, char **argv)
{ int x[15] = {6,12,5,13,3,5,4,5,8,88,1,11,9,22,168};
    int n=15;
    thrust:: device_vector <int> dx (x, x+n);
    thrust:: sort(dx.begin(),dx.end());
    thrust::device_vector <int> seq(n);
    thrust::sequence(seq.begin(), seq.end (),0);
    thrust::device_vector<int> out(n);
    int incr = atoi(argv[1]); // k
    // for each i in seq, call ismultk() on this i, and if get a true
    // result, put dx[i] into out
    thrust::device_vector<int >::iterator newend =
        thrust::copy_if(dx.begin(), dx.end(),seq.begin(),out.begin(),
            ismultk(incr));
    thrust::copy(out.begin(), newend,
        std::ostream_iterator<int>(std::cout, " "));
    std::cout << "\n";
}
```

The sequence() function simply generates an array consisting of $0,1,2, \ldots, \mathrm{n}-1$.
Our functor here is a little more advanced than the one we saw earlier. It now has an argument, which is incr in the case of our call,

```
thrust:: copy_if(dx.begin(), dx.end(), seq,out,ismultk(incr));
```

That is then used to set a member variable in the struct:

```
const int increm; // k in above comments
ismultk(int _increm): increm(_increm) {}
```

This is in a sense the second, though nonexplicit, argument to our calls to ismultk(). For example, in our call,

```
thrust:: copy_if(hx.begin(),hx.end(), seq,out,ismultk(incr));
```

the function designated by operator within the ismultk struct will be called individually on each element in $\mathbf{h x}$, each one playing role of $\mathbf{i}$ in

```
bool operator()(const int i)
{ return i != 0 && (i % increm) == 0;
}
```

Since this code references increm, the value incr in our call above is used as well. The variable increm acts as a "global" variable to all the actions of the operator.

### 6.5 Mixing Thrust and CUDA Code

In order to mix Thrust and CUDA code, Thrust has the function thrust::raw_pointer_cast() to convert from a Thrust device pointer type to a CUDA device pointer type, and has thrust::device_ptr to convert in the other direction.

In our example in Section 6.6, we convert from Thrust to an ordinary address on the device:

```
int *wd;
wd = thrust::raw_pointer_cast(&w[0]);
{`.if (i != 0 && (i % increm) = 0) wd[i] = 2* wd[i];
```

In the other direction, say we start with a CUDA pointer, and want to use it in Thrust. We might have something like

```
int *dz;
cudaMalloc(&dz,100*sizeof(int));
thrust::device_ptr<int> tz(dz);
int k = thrust::reduce(tz,tz+100, (int) 0, thrust::plus<int >());
```


### 6.6 Example: Doubling Every ${ }^{\text {th }}$ Element of an Array

Let's adapt the code from the last section in order to illustrate another technique.
Suppose instead of copying every $\mathrm{k}^{\text {th }}$ element of an array (after this first one), we wish to merely double each such element. There are various ways we could do this, but here we'll use an approach that shows another way we can use functors.

```
// double every k-th element in given array; k obtained from command
// line
#include <stdio.h>
#include <thrust/device_vector.h>
#include <thrust/sequence.h>
#include <thrust/remove.h> // for copy_if()
// functor
struct ismultk
{
    const int increm; // k in above comments
```

```
    const thrust:: device_vector<int >::iterator w; // "pointer" to our array
    int *wd;
    // get "pointer," k
    ismultk(thrust:: device_vector <int >::iterator _w, int _increm):
        w(_w), increm(_increm) {
            wd = thrust::raw_pointer_cast(&w[0]);
        }
    __device__
    bool operator()(const int i) // bool is phony, but void doesn't work
    { if (i != 0 && (i % increm) =0) wd[i] = 2* wd[i];
    }
};
// test
int main(int argc, char **argv)
{ // test case:
    int x[15] = {6,12,5,13,3,5,4,5,8,88,1,11,9,22,168};
    int n=15;
    thrust:: device_vector <int> dx (x,x+n);
    thrust::device_vector <int> seq(n);
    thrust::sequence(seq.begin(), seq.end (),0);
    thrust:: device_vector<int> out(n);
    int incr = atoi(argv[1]); // k
    // for each i in seq, call ismultk() on this i, and if get a true
    // result, put 0 in dx[i]
    thrust:: copy_if(dx.begin(), dx.end(),seq.begin(),out.begin(),
        ismultk(dx.begin(),incr));
    // did it work?
    thrust::copy(dx.begin(), dx.end(),
        std::ostream_iterator <int>(std::cout, " "));
    std::cout << "\n";
}
```

Our functor here is quite different from before.
First, one of the functor's arguments is an iterator, rather than a simple type like int. This is really just like passing an array pointer to an ordinary C function.

Second, we've converted that iterator to a simple device array

```
const thrust::device_vector<int >::iterator w; // "pointer" to our array
int *wd;
...
    wd = thrust::raw_pointer_cast(&w[0]);
```

Our call to copy_if() doesn't actually do any copying. We are exploiting the "if" in "copy if," not the "copy."

### 6.7 Scatter and Gather Operations

These basically act as permuters; see the comments in the following small examples.

## scatter:

```
// illustration of thrust::scatter(); permutes an array according to a
// map array
#include <stdio.h>
#include <thrust/device_vector.h>
#include <thrust/scatter.h>
int main()
{ int x[5] = {12,13,5,8,88};
    int n=5;
    thrust:: device_vector <int> dx (x, x+n);
    // allocate map vector
    thrust:: device_vector <int> dm(n);
    // allocate vector for output of gather
    thrust::device_vector<int> hdst(n);
    // example map
    int m[5] = {3,2,4,1,0};
    thrust::copy (m,m+n,dm.begin());
    thrust:: scatter(dx.begin(),dx.end(),dm.begin(), ddst.begin());
    // the original x[0] should now be at position 3, the original x[1]
    // now at position 2, etc., i.e. 88,8,13,12;,5 check it:
    thrust::copy(ddst.begin(), ddst.end(),
        std::ostream_iterator<int>(std::cout," "));
    std::cout << "\n";
}
```


## gather():

```
// illustrations of thrust::gather(); permutes an array according to a
// map array
#include <stdio.h>
#include <thrust/device_vector.h>
#include <thrust/gather.h>
int main()
{ int x[5] = {12,13,5,8,88};
    int n=5;
    thrust:: device_vector <int> dx (x, x+n);
    // allocate map vector
    thrust::device_vector <int> dm(n);
    // allocate vector for output of gather
    thrust::device_vector <int> ddst(n);
    // example map
```

```
    int m[5] = {3,2,4,1,0};
    thrust::copy(m,m+n,dm.begin());
    thrust::gather(dm. begin(),dm.end(),dx.begin(), ddst.begin());
    // the original x[3] should now be at position 0, the original x[2]
    // now at position 1, etc., i.e. 8,5,88,13,12; check it:
    thrust:: copy(ddst.begin(), ddst.end(),
        std::ostream_iterator <int>(std::cout, " "));
    std::cout << "\n";
```

\}

### 6.7.1 Example: Matrix Transpose

Here's an example of scatter(), applying it to transpose a matrix:

```
// matrix transpose, using scatter()
// similar to (though less efficient than) code included in the examples
// in the Thrust package
// matrices assumed stored in one dimension, row-major order
#include <stdio.h>
#include <thrust/device_vector.h>
#include <thrust/scatter.h>
#include <thrust/sequence.h>
struct transidx {
    const int nr; // number of rows in input
    const int nc; // number of columns in input
    // set nr, nc
    __host__ __device__
    transidx(int _nr, int _nc): nr(_nr), nc(_nc) {};
    // element i in input should map to which element in output?
    __host__ __device__
    int operator()(const int i)
    { int r = i / nc; int c = i % nc; // row r, col c in input
        // that will be row c and col r in output, which has nr cols
        return c * nr + r;
    }
};
int main()
{ int mat[6] ={ // test data
            5, 12, 13,
            3, 4, 5};
    int nrow=2,ncol=3,n=nrow * ncol;
    thrust:: device_vector <int> dmat(mat,mat+n);
    // allocate map vector
```

```
    thrust::device_vector<int> dmap(n);
    // allocate vector for output of gather
    thrust::device_vector<int> ddst(n);
    // construct map; element r of input matrix goes to s of output
    thrust::device_vector<int> seq(n);
    thrust::sequence (seq.begin(),seq.end());
    thrust::transform(seq.begin(), seq.end(),dmap. begin(), transidx(nrow, ncol));
    thrust::scatter(dmat.begin(),dmat.end(),dmap.begin(),ddst.begin());
    // ddst should now hold the transposed matrix, 5,3,12,4,13,5; check it:
    thrust::copy(ddst.begin(), ddst.end(), std::ostream_iterator<int>(std::cout, " "));
    std::cout << "\n";
```

\}

The idea is to determine, for each index in the original matrix, the index for that element in the transposed matrix. Not much new here in terms of Thrust, just more complexity.
It should be mentioned that the performance of this algorithm with a GPU backend would likely be better if matrix tiling were used (Section 11.2).

### 6.8 Advanced ("Fancy") Iterators

Since each Thrust call invokes considerable overhead, Thrust offers some special iterators to reduce memory access time and memory space requirements. Here are a few:

- Counting iterators: These play the same role as thrust::sequence(), but without actually setting up an array, thus avoiding the memory issues.
- Transform iterators: If your code first calls thrust:transform() and then makes another Thrust call on the result, you can combine them, which the Thrust people call fusion.
- Zip iterators: These essentially "zip" together two arrays (picture two halves of a zipper lining up parallel to each other as you zip up a coat). This is often useful when one needs to retain information on the position of an element within its array.
- Discard iterators: Sometimes we call transform() but don't need its output. Discard iterators then act in a manner similar to / dev/null.


### 6.8.1 Example: Matrix Transpose Again

Let's re-do the example of Section 6.7.1, this time using fusion.
// matrices assumed stored in one dimension, row-major order

```
#include <stdio.h>
#include <thrust/device_vector.h>
#include <thrust/scatter.h>
#include <thrust/sequence.h>
#include <thrust/iterator/transform_iterator.h>
struct transidx : public thrust::unary_function<int,int>
{
    const int nr; // number of rows in input
    const int nc; // number of columns in input
    // set nr, nc
    --host_- --device.-
    transidx(int _nr, int _nc): nr(_nr), nc(_nc) {};
    // element i in input should map to which element in output?
    _-host_- _-device_-
    int operator()(int i)
    { int r = i / nc; int c = i % nc; // row r, col c in input
        // that will be row c and col r in output, which has nr cols
        return c * nr + r;
    }
};
int main()
{ int mat[6] = {
            5, 12, 13,
            3, 4, 5};
    int nrow=2,ncol=3,n=nrow*ncol;
    thrust::device_vector <int> dmat(mat,mat+n);
    // allocate map vector
    thrust::device_vector<int> dmap(n);
    // allocate vector for output of gather
    thrust::device_vector<int> ddst(n);
    // construct map; element r of input matrix goes to s of output
    thrust:: device_vector<int> seq(n);
    thrust::sequence (seq.begin(), seq.end());
    thrust::scatter(
        dmat.begin(),dmat.end(),
        thrust::make_transform_iterator(seq. begin(), transidx(nrow, ncol)),
        ddst.begin());
    thrust::copy(ddst.begin(), ddst.end(),
        std::ostream_iterator<int>(std::cout, " "));
    std::cout << "\n";
}
```

The key new code here is:

```
thrust::scatter(
    dmat.begin(), dmat.end(),
    thrust::make_transform_iterator(seq.begin(), transidx (nrow, ncol)),
    ddst.begin());
```

Fusion requires a special type of iterator, whose type is horrendous to write. So, Thrust provides the make_transform_iterator() function, which we call to produce the special iterator needed, and then put the result directly into the second phase of our fusion, in this case into scatter().

Essentially our use of make_transform_iterator() is telling Thrust, "Don't apply transidx() to seq yet. Instead, perform that operation as you go along, and feed each result of transidx() directly into scatter ()." That word direct is the salient one here; it means we save n memory reads and $n$ memory writes $4^{4}$ Moreover, we save the overhead of the kernel call, if our backend is CUDA.

Note that we also had to be a little bit more elaborate with data typing issues, writing the first line of our struct declaration as

```
struct transidx : public thrust:: unary_function<int,int>
```

It won't work without this!
It would be nice to be able to use a counting iterator in the above code, but apparently the compiler encounters problems with determining where the end of the counting sequence is. There is similar code in the examples directory that comes with Thrust, and that one uses gather() instead of scatter(). Since the former specifies a beginning and an end for the map array, counting interators work fine.

### 6.9 A Timing Comparison

Let's look at matrix transpose one more time. First, we'll use the method, shown in earlier sections, of passing a device vector iterator to a functor. For variety, let's use Thrust's for_each() function. The following will be known as Code 1:

```
// matrix transpose, for_each version
#include <stdio.h>
#include <thrust/device_vector.h>
// functor; holds iterators for the input and output matrices, and each
// invocation of the function copies from one element from the former to
// the latter
struct copyelt2xp
{
    int nrow;
    int ncol;
    const thrust::device_vector<int >::iterator m; // input matrix
    const thrust::device_vector<int >::iterator mxp; // output matrix
    int *m1,*mxp1;
```

[^37]```
    copyelt 2xp(thrust:: device_vector<int >::iterator _m,
                thrust:: device_vector<int > : iteerator _mxp,
                int _nr, int _nc):
        m(_m), mxp(_mxp), nrow(_nr), ncol(_nc) {
            m1 = thrust::raw_pointer_cast (&m[0]);
            mxp1 = thrust:: raw_pointer_cast (&mxp [0]);
        }
    __device__
    void operator ()(const int i)
    // copies the i-th element of the input matrix to the output matrix
    { // elt i in input is row r, col c there
        int r = i / ncol; int c = i % ncol;
        // that elt will be row c and col r in output, which has nrow
        // cols, so copy as follows
        mxp1[c*nrow +r ] = m1[r*ncol+c];
    }
};
// transpose nr x nc inmat to outmat
void transp(int *inmat, int *outmat, int nr, int nc)
{
    thrust:: device_vector <int> dmat(inmat, inmat+nr*nc);
    // make space for the transpose
    thrust::device_vector <int> dxp(nr*nc);
    thrust:: counting_iterator <int> seqb (0);
    thrust:: counting_iterator <int> seqe = seqb + nr*nc;
    // for each i in seq, copy the matrix elt to its spot in the
    // transpose
    thrust:: for_each(seqb, seqe,
        copyelt 2xp(dmat.begin(), dxp.begin(), nr, nc ));
    thrust : : copy (dxp.begin(), dxp.end (), outmat );
}
int rand16() // generate random integers mod 16
{ return rand() % 16; }
// test code: cmd line args are matrix size, then row, col of elt to be
// checked
int main(int argc, char **argv)
{ int nr = atoi(argv[1]); int nc = nr;
    int *mat = (int *) malloc(nr*nc*sizeof(int));
    int *matxp = (int *) malloc(nr*nc*sizeof(int));
    thrust::generate (mat, mat+nr*nc, rand16);
    int checkrow = atoi(argv [2]);
    int checkcol= atoi(argv[3]);
    printf("%d\n",mat[checkrow *nc+checkcol]);
    transp(mat, matxp,nr,nc);
    printf("%d\n",matxp[checkcol*nc+checkrow]);
}
```

The for_each() function does what the name implies: It calls a function/functor for each element in a sequence, doing so in a parallel manner. Note that this also obviates our earlier need to use a discard iterator.

For comparison, we'll use the matrix transpose code that is included in Thrust's examples/ file, to be referred to as Code 2:

```
// matrix transpose, from the Thrust package examples
#include <thrust/host_vector.h>
#include <thrust/device_vector.h>
#include <thrust/functional.h>
#include <thrust/gather.h>
#include <thrust/scan.h>
#include <thrust/iterator/counting_iterator.h>
#include <thrust/iterator/transform_iterator.h>
#include <iostream>
#include <iomanip>
#include <stdio.h>
// convert a linear index to a linear index in the transpose
struct transpose_index : public thrust:: unary_function<size_t,size_t>
{
    size_t m, n;
    __host__ _-device__
    transpose_index(size_t _m, size_t _n) : m(_m), n(_n) {}
    __host__ __device__
    size_t operator()(size_t linear_index)
    {
        size_t i = linear_index / n;
        size_t j = linear_index % n;
        return m * j + i;
    }
};
// convert a linear index to a row index
struct row_index : public thrust:: unary_function<size_t,size_t>
{
    size_t n;
    __host__ _-device__
    row_index(size_t _n) : n(_n) {}
    __host__ _-device__
    size_t operator()(size_t i)
    {
        return i / n;
```

```
    }
};
// transpose an M-by-N array
template <typename T>
void transpose(size_t m, size_t n, thrust:: device_vector<T>& src, thrust:: device_vector<T>& dst
{
    thrust::counting_iterator<size_t> indices(0);
    thrust::gather
        (thrust::make_transform_iterator(indices, transpose_index(n, m)),
            thrust::make_transform_iterator(indices, transpose_index(n, m)) + dst.size(),
            src.begin(),
            dst.begin());
}
void transp(int *inmat, int *outmat, int nr, int nc)
{
    thrust::device_vector<int> dmat(inmat,inmat+nr*nc);
    // make space for the transpose
    thrust::device_vector<int> dxp(nr*nc);
    transpose(nr,nc,dmat,dxp);
    thrust::copy(dxp.begin(),dxp.end(),outmat);
}
int rand16() // generate random integers mod 16
{ return rand() % 16; }
// test code: cmd line args are matrix size, then row, col of elt to be
// checked
int main(int argc, char **argv)
{ int nr= atoi(argv[1]); int nc= nr;
    int *mat = (int *) malloc(nr*nc*sizeof(int));
    int *matxp = (int *) malloc(nr*nc*sizeof(int));
    thrust::generate(mat,mat+nr*nc, rand16);
    int checkrow = atoi(argv[2]);
    int checkcol = atoi(argv[3]);
    printf("%d\n", mat[checkrow*nc+checkcol]);
    transp(mat, matxp,nr,nc);
    printf("%d\n",matxp[checkcol*nc+checkrow]);
}
```

This approach is more efficient than ours in Section 6.7.1 making use of gather() instead of scatter(). It also takes advantage of fusion etc.

Code 1 is a lot easier to program than Code 2, but is it efficient? It turns out, though, that-good news!-the simpler code, i.e. Code 1, is actually a little faster than Code 2 in the case of a CUDA backend, and a lot faster in the OpenMP case.

Here we ran on CUDA backends, on a $10000 \times 10000$ matrix:

| device | Code 1 | Code 2 |
| ---: | ---: | ---: |
| GeForce 9800 GTX | 3.67 | 3.75 |
| Tesla C2050 | 3.43 | 3.50 |

What about OpenMP? Here are some timing runs on a multicore machine (many more cores than the 16 we tried), using an input matrix of size $6000 \times 6000$ :

| \# threads | Code 1 | Code 2 |
| ---: | ---: | ---: |
| 2 | 9.57 | 23.01 |
| 4 | 5.17 | 10.62 |
| 8 | 3.01 | 7.42 |
| 16 | 1.99 | 3.35 |

### 6.10 Example: Transforming an Adjacency Matrix

Here is a Thrust approach to the example of Sections 4.13 and 5.11. To review, here is the problem:
Say we have a graph with adjacency matrix

$$
\left(\begin{array}{llll}
0 & 1 & 0 & 0  \tag{6.1}\\
1 & 0 & 0 & 1 \\
0 & 1 & 0 & 1 \\
1 & 1 & 1 & 0
\end{array}\right)
$$

with row and column numbering starting at 0 , not 1 . We'd like to transform this to a two-column matrix that displays the links, in this case

$$
\left(\begin{array}{ll}
0 & 1  \tag{6.2}\\
1 & 0 \\
1 & 3 \\
2 & 1 \\
2 & 3 \\
3 & 0 \\
3 & 1 \\
3 & 2
\end{array}\right)
$$

For instance, there is a 1 on the far right, second row of the above matrix, meaning that in the graph there is an edge from vertex 1 to vertex 3 . This results in the row (1,3) in the transformed matrix seen above.

Here's Thrust code to do this:

```
// transgraph problem, using Thrust
#include <stdio.h>
#include <thrust/device_vector.h>
#include <thrust/transform.h>
#include <thrust/remove.h>
#include <thrust/iterator/discard_iterator.h>
// forms one row of the output matrix
struct makerow {
    const thrust:: device_vector <int >::iterator outmat;
    int *om;
    const int nc; // number of columns
    makerow(thrust:: device_vector<int >::iterator _outmat,int _nc) :
        outmat(_outmat), nc(_nc) { om = thrust::raw_pointer_cast(&outmat[0]); }
    __device_-
    // the j-th 1 is in position i of the orig matrix
    bool operator()(const int i, const int j)
    { om[2*j] = i / nc;
        om[2*j+1]= i % nc;
    }
};
int main(int argc, char **argv)
{ int x[12]={
            0,1,1,0,
            1,0,0,1,
        1,1,0,0};
    int nr=3,nc=4, nrc = nr*nc, i;
    thrust:: device_vector <int> dx(x,x+nrc);
    thrust::device_vector <int> ones(x,x+nrc);
    thrust:: counting_iterator <int> seqb (0);
    thrust::counting_iterator <int> seqe = seqb + nrc;
    // get 1-D indices of the 1s
    thrust:: device_vector <int >::iterator newend =
        thrust:: copy_if(seqb, seqe, dx.begin(),ones.begin(),
            thrust::identity<int>());
    int n1s = newend - ones.begin();
    thrust::device_vector<int> newmat (2*n1s);
    thrust::device_vector<int> out(n1s);
    thrust::counting_iterator <int> seq2b(0);
    thrust::transform(ones.begin(), newend,seq2b,
        thrust::make_discard_iterator(), makerow(newmat. begin(), nc));
    thrust:: copy(newmat.begin(), newmat.end(),
        std::ostream_iterator <int >(std::cout, " "));
    std::cout << "\n";
}
```

One new feature here is the use of counting iterators. First, we create two of them in the code

```
thrust:: counting_iterator < int> seqb (0);
thrust:: counting_iterator <int > seqe = seqb + nrc;
```

Here seqb (virtually) points to the 0 in $0,1,2, \ldots$ Actually no array is set up, but references to seqb will act as if there is an array there. The counting iterator seqb starts at nrc, but its role here is simply to demarcate the end of the (virtual) array.

Now, how does the code work? The call to copy_if() has the goal of indentifying where in $\mathbf{d x}$ the 1 s are located. This is accomplished by calling Thrust's identity() function, which just does $f(x)$ $=\mathrm{x}$, which is enough, as it will return either 1 or 0 , the latter interpreted as True. In other words, the values between seqb and seqe will be copied whenever the corresponding values in dx are 1 s . The copied values are then placed into our array ones, which will now tell us where in $\mathbf{d x}$ the 1 s are. Each such value, recall, will correspond to one row of our output matrix. The construction of the latter action is done by calling transform():

```
thrust::transform(ones.begin(), newend, seq2b,
    thrust::make_discard_iterator(), makerow(newmat.begin(), nc ));
```

The construction of the output matrix, newmat, is actually done as a side effect of calling makerow(). For this reason, we've set our third parameter to thrust::make_discard_iterator(). Since we never use the output from transform() itself, and it thus would be wasteful-of both memory space and memory bandwidth - to store that output in a real array. Hence we use a discard array instead.

Our algorithm consists of two stages-first finding the locations of the 1 s , and then calculating the output matrix. Could we combine the two stages? Possibly, but there are difficulties to deal with.

The biggest problem is that we don't know the size of the output matrix in advance; counting the 1 s separately gives us that information. Without that, we'd either have to make the output matrix too large initially and then shrink it, or continually expand it as we go through the computation. The latter would probably result in a major slowdown, as memory allocation takes time.

### 6.11 Prefix Scan

Thrust includes functions for prefix scan (see Chapter 10):

```
// illustration of parallel prefix sum
#include <stdio.h>
#include <thrust/device_vector.h>
#include <thrust/scan.h>
```

```
int main(int argc, char **argv)
{ int x[7] = {6,12,5,13,3,5,4};
    int n=7,i;
    thrust:: device_vector <int> hx (x,x+n);
    // in-place scan; default operation is +
    thrust::inclusive_scan(hx.begin(), hx.end(), hx.begin());
    thrust::copy(hx.begin(), hx.end(),
        std::ostream_iterator<int>(std::cout, " "));
    std::cout << "\n";
}
```


### 6.12 More on Use of Thrust for a CUDA Backend

### 6.12.1 Synchronicity

Thrust calls are in fact CUDA kernel calls, and thus entail some latency. Other than the transform()family functions, the calls are all synchronous.

### 6.13 Error Messages

A message like

```
terminate called after throwing an instance of 'std:: bad_alloc'
    what(): std:: bad_alloc
```

may mean that Thrust wasn't able to allocate your large array on the GPU.
Also, beware of the following. Consider the code

```
thrust::device_vector <int> seq(n);
thrust::copy_if(hx.begin(), hx.end(), seq,out,ismultk(hx.begin(), incr));
```

We forgot the .begin() for seq! If seq had been a non-Thrust array, declared as

```
int seq[n];
```

it would have been fine, but not for a Thrust array.
Unfortunately, the compiler gives us a very long megillah as an error message, a highly uninformative one. Keep this in mind if you get a 30 -line compiler error.

The same thing happens if we forget to state the proper "include" files.

### 6.14 Other Examples of Thrust Code in This Book

- An application of Thrust's prefix-scan functionality is presented in Section 10.5


## Chapter 7

## Message Passing Systems

Message passing systems are probably the most common platforms for parallel processing today.

### 7.1 Overview

Traditionally, shared-memory hardware has been extremely expensive, with a typical system costing hundreds of thousands of dollars. Accordingly, the main users were for very large corporations or government agencies, with the machines being used for heavy-duty server applications, such as for large databases and World Wide Web sites. The conventional wisdom is that these applications require the efficiency that good shared-memory hardware can provide.

But the huge expense of shared-memory machines led to a quest for high-performance messagepassing alternatives, first in hypercubes and then in networks of workstations (NOWs).

The situation changed radically around 2005, when "shared-memory hardware for the masses" became available in dual-core commodity PCs. Chips of higher core multiplicity are commercially available, with a decline of price being inevitable. Ordinary users will soon be able to afford shared-memory machines featuring dozens of processors.

Yet the message-passing paradigm continues to thrive. Many people believe it is more amenable to writing really fast code, and the the advent of cloud computing has given message-passing a big boost. In addition, many of the world's very fastest systems (see www.top500.org for the latest list) are in fact of the message-passing type.

In this chapter, we take a closer look at this approach to parallel processing.

### 7.2 A Historical Example: Hypercubes

A popular class of parallel machines in the 1980s and early 90s was that of hypercubes. Intel sold them, for example, as did a subsidiary of Oracle, nCube. A hypercube would consist of some number of ordinary Intel processors, with each processor having some memory and serial I/O hardward for connection to its "neighbor" processors.

Hypercubes proved to be too expensive for the type of performance they could achieve, and the market was small anyway. Thus they are not common today, but they are still important, both for historical reasons (in the computer field, old techniques are often recycled decades later), and because the algorithms developed for them have become quite popular for use on general machines. In this section we will discuss architecture, algorithms and software for such machines.

### 7.2.1 Definitions

A hypercube of dimension d consists of $D=2^{d}$ processing elements (PEs), i.e. processormemory pairs, with fast serial I/O connections between neighboring PEs. We refer to such a cube as a d-cube.

The PEs in a d-cube will have numbers 0 through D-1. Let $\left(c_{d-1}, \ldots, c_{0}\right)$ be the base- 2 representation of a PE's number. The PE has fast point-to-point links to d other PEs, which we will call its neighbors. Its ith neighbor has number $\left(c_{d-1}, \ldots, 1-c_{i-1}, \ldots, c_{0}\right) \cdot{ }^{1}$

For example, consider a hypercube having $\mathrm{D}=16$, i.e. $\mathrm{d}=4$. The PE numbered 1011, for instance, would have four neighbors, 0011, 1111, 1001 and 1010.


It is sometimes helpful to build up a cube from the lower-dimensional cases. To build a (d+1)-

[^38]dimensional cube from two d-dimensional cubes, just follow this recipe:
(a) Take a d-dimensional cube and duplicate it. Call these two cubes subcube 0 and subcube 1 .
(b) For each pair of same-numbered PEs in the two subcubes, add a binary digit 0 to the front of the number for the PE in subcube 0 , and add a 1 in the case of subcube 1. Add a link between them.

The following figure shows how a 4-cube can be constructed in this way from two 3-cubes:


Given a PE of number $\left(c_{d-1}, \ldots, c_{0}\right)$ in a d-cube, we will discuss the i-cube to which this PE belongs, meaning all PEs whose first d-i digits match this PE's 2 Of all these PEs, the one whose last i digits are all 0 s is called the root of this i-cube.

For the 4-cube and PE 1011 mentioned above, for instance, the 2-cube to which that PE belongs consists of $1000,1001,1010$ and 1011-i.e. all PEs whose first two digits are 10 - and the root is 1000.

Given a PE, we can split the i-cube to which it belongs into two (i-1)-subcubes, one consisting of those PEs whose digit i-1 is 0 (to be called subcube 0), and the other consisting of those PEs whose digit i-1 is 1 (to be called subcube 1). Each given PE in subcube 0 has as its partner the PE in subcube 1 whose digits match those of the given PE, except for digit i-1.

[^39]To illustrate this, again consider the 4 -cube and the PE 1011. As an example, let us look at how the 3 -cube it belongs to will split into two 2-cubes. The 3 -cube to which 1011 belongs consists of $1000,1001,1010,1011,1100,1101,1110$ and 1111. This 3 -cube can be split into two 2-cubes, one being $1000,1001,1010$ and 1011 , and the other being $1100,1101,1110$ and 1111 . Then PE 1000 is partners with PE 1100, PE 1001 is partners with PE 1101, and so on.

Each link between two PEs is a dedicated connection, much preferable to the shared link we have when we run, say, MPI, on a collection of workstations on an Ethernet. On the other hand, if one PE needs to communicate with a non-neighbor PE, multiple links (as many as d of them) will need to be traversed. Thus the nature of the communications costs here is much different than for a network of workstations, and this must be borne in mind when developing programs.

### 7.3 Networks of Workstations (NOWs)

The idea here is simple: Take a bunch of commodity PCs and network them for use as parallel processing systems. They are of course individual machines, capable of the usual uniprocessor, nonparallel applications, but by networking them together and using message-passing software environments such as MPI, we can form very powerful parallel systems.

The networking does result in a significant loss of performance, but the price/performance ratio in NOW can be much superior in many applications to that of shared-memory or hypercube hardware of comparable number of CPUs.

### 7.3.1 The Network Is Literally the Weakest Link

Still, one factor which can be key to the success of a NOW is to use a fast network, both in terms of hardware and network protocol. Ordinary Ethernet and TCP/IP are fine for the applications envisioned by the original designers of the Internet, e.g. e-mail and file transfer, but they are slow in the NOW context.

A popular network for a NOW today is Infiniband (IB) (www.infinibandta.org). It features low latency, about 1.0-3.0 microseconds, high bandwidth, about 1.0-2.0 gigaBytes per second), and uses a low amount of the CPU's cycles, around 5-10\%.

The basic building block of IB is a switch, with many inputs and outputs, similar in concept to $\Omega$-net. You can build arbitrarily large and complex topologies from these switches.

A central point is that IB, as with other high-performance networks designed for NOWs, uses RDMA (Remote Direct Memory Access) read/write, which eliminates the extra copying of data between the application program's address space to that of the operating system.

IB has high performance and scalable ${ }^{3}$ implementations of distributed locks, semaphores, collective communication operations. An atomic operation takes about 3-5 microseconds.

IB implements true multicast, i.e. the simultaneous sending of messages to many nodes. Note carefully that even though MPI has its MPI_Bcast() function, it will send things out one at a time unless your network hardware is capable of multicast, and the MPI implementation you use is configured specifically for that hardware.

For information on network protocols, e.g. for example www.rdmaconsortium.org A research paper evaluating a tuned implementation of MPI on IB is available at nowlab.cse.ohio-state. edu/publications/journal-papers/2004/liuj-ijpp04.pdf.

### 7.3.2 Other Issues

Increasingly today, the workstations themselves are multiprocessor machines, so a NOW really is a hybrid arrangement. They can be programmed either purely in a message-passing manner-e.g. running eight MPI processes on four dual-core machines - or in a mixed way, with a shared-memory approach being used within a workstation but message-passing used between them.

NOWs have become so popular that there are now "recipes" on how to build them for the specific purpose of parallel processing. The term Beowulf come to mean a NOW, usually with a fast network connecting them, used for parallel processing. The term $N O W$ itself is no longer in use, replaced by cluster. Software packages such as ROCKS (http://www.rocksclusters.org/ wordpress/) have been developed to make it easy to set up and administer such systems.

### 7.4 Scatter/Gather Operations

Writing message-passing code is a lot of work, as the programmer must explicitly arrange for transfer of data. Contrast that, for instance, to shared-memory machines, in which cache coherency transactions will cause data transfers, but which are not arranged by the programmer and not even seen by him/her.

In order to make coding on message-passing machines easier, higher-level systems have been devised. These basically operate in the scatter/gather paradigm, in which a "manager" node sends out chunks of work to the other nodes, serving as "workers," and then collects and assembles the results sent back the workers.

MPI includes scatter/gather operations in its wide offering of functions, and they are used in many

[^40]MPI applications. R's snow package, which will be discussed in Section ??, is based entirely on scatter/gather, as is MapReduce, to be discussed below.

## Chapter 8

## Introduction to MPI

MPI is the de facto standard for message-passing software.

### 8.1 Overview

### 8.1.1 History

Though (small) shared-memory machines have come down radically in price, to the point at which a dual-core PC is now commonplace in the home, historically shared-memory machines were available only to the "very rich"-large banks, national research labs and so on. This led to interest in message-passing machines.

The first "affordable" message-machine type was the Hypercube, developed by a physics professor at Cal Tech. It consisted of a number of processing elements (PEs) connected by fast serial I/O cards. This was in the range of university departmental research labs. It was later commercialized by Intel and NCube.

Later, the notion of networks of workstations (NOWs) became popular. Here the PEs were entirely independent PCs, connected via a standard network. This was refined a bit, by the use of more suitable network hardware and protocols, with the new term being clusters.

All of this necessitated the development of standardized software tools based on a message-passing paradigm. The first popular such tool was Parallel Virtual Machine (PVM). It still has its adherents today, but has largely been supplanted by the Message Passing Interface (MPI).

MPI itself later became MPI 2. Our document here is intended mainly for the original.

### 8.1.2 Structure and Execution

MPI is merely a set of Application Programmer Interfaces (APIs), called from user programs written in C, C++ and other languages. It has many implementations, with some being open source and generic, while others are proprietary and fine-tuned for specific commercial hardware.

Suppose we have written an MPI program $\mathbf{x}$, and will run it on four machines in a cluster. Each machine will be running its own copy of $\mathbf{x}$. Official MPI terminology refers to this as four processes. Now that multicore machines are commonplace, one might indeed run two or more cooperating MPI processes - where now we use the term processes in the real OS sense - on the same multicore machine. In this document, we will tend to refer to the various MPI processes as nodes, with an eye to the cluster setting.

Though the nodes are all running the same program, they will likely be working on different parts of the program's data. This is called the Single Program Multiple Data (SPMD) model. This is the typical approach, but there could be different programs running on different nodes. Most of the APIs involve a node sending information to, or receiving information from, other nodes.

### 8.1.3 Implementations

Two of the most popular implementations of MPI are MPICH and LAM. MPICH offers more tailoring to various networks and other platforms, while LAM runs on networks. Introductions to MPICH and LAM can be found, for example, at http://heather.cs.ucdavis.edu/~matloff/ MPI/NotesMPICH.NM.html and http://heather.cs.ucdavis.edu/~matloff/MPI/NotesLAM.NM. html, respectively.

LAM is no longer being developed, and has been replaced by Open MPI (not to be confused with OpenMP). Personally, I still prefer the simplicity of LAM. It is still being maintained.

Note carefully: If your machine has more than one MPI implementation, make absolutely sure one is not interfering with the other. Make sure all execution and library paths all include one and only one implementation at a time.

### 8.1.4 Performance Issues

Mere usage of a parallel language on a parallel platform does not guarantee a performance improvement over a serial version of your program. The central issue here is the overhead involved in internode communication.

Infiniband, one of the fastest cluster networks commercially available, has a latency of about 1.03.0 microseconds, meaning that it takes the first bit of a packet that long to get from one node on
an Infiniband switch to another. Comparing that to the nanosecond time scale of CPU speeds, one can see that the communications overhead can destroy a program's performance. And Ethernet is quite a bit slower than Infiniband.

Latency is quite different from bandwidth, which is the number of bits sent per second. Say the latency is 1.0 microsecond and the bandwidth is 1 gigabit, i.e. 1000000000 bits per second or 1000 bits per microsecond. Say the message is 2000 bits long. Then the first bit of the message arrives after 1 microsecond, and the last bit arrives after an additional 2 microseconds. In other words, the message is does not arrive fully at the destination until 3 microseconds after it is sent.

In the same setting, say bandwidth is 10 gigabits. Now the message would need 1.2 seconds to arrive fully, in spite of a 10 -fold increase in bandwidth. So latency is a major problem even if the bandwidth is high.

For this reason, the MPI applications that run well on networks tend to be of the "embarrassingly parallel" type, with very little communication between the processes.

Of course, if your platform is a shared-memory multiprocessor (especially a multicore one, where communication between cores is particularly fast) and you are running all your MPI processes on that machine, the problem is less severe. In fact, some implementations of MPI communicate directly through shared memory in that case, rather than using the TCP/IP or other network protocol.

### 8.2 Review of Earlier Example

Though the presentation in this chapter is self-contained, you may wish to look first at the somewhat simpler example in Section 1.3.3.2, a pipelined prime number finder.

### 8.3 Example: Dijkstra Algorithm

### 8.3.1 The Algorithm

The code implements the Dijkstra algorithm for finding the shortest paths in an undirected graph. Pseudocode for the algorithm is

```
Done = {0}
NonDone = {1,2,\ldots.,N-1}
for J = 1 to N-1 Dist[J] = infinity`
Dist[0] = 0
for Step = 1 to N-1
    find J such that Dist[J] is min among all J in NonDone
```

```
transfer J from NonDone to Done
NewDone = J
for K = 1 to N-1
    if K is in NonDone
        Dist[K] = min(Dist[K],Dist[NewDone]+G[NewDone,K])
```

At each iteration, the algorithm finds the closest vertex J to 0 among all those not yet processed, and then updates the list of minimum distances to each vertex from 0 by considering paths that go through J. Two obvious potential candidate part of the algorithm for parallelization are the "find J" and "for K" lines, and the above OpenMP code takes this approach.

### 8.3.2 The MPI Code

```
// Dijkstra.c
// MPI example program: Dijkstra shortest-path finder in a
// bidirectional graph; finds the shortest path from vertex O to all
// others
// command line arguments: nv print dbg
// where: nv is the size of the graph; print is 1 if graph and min
// distances are to be printed out, 0 otherwise; and dbg is 1 or 0, 1
// for debug
// node O will both participate in the computation and serve as a
// "manager"
#include <stdio.h>
#include <mpi.h>
#define MYMIN_MSG O
#define OVRLMIN_MSG 1
#define COLLECT_MSG 2
// global variables (but of course not shared across nodes)
int nv, // number of vertices
    *notdone, // vertices not checked yet
    nnodes, // number of MPI nodes in the computation
    chunk, // number of vertices handled by each node
    startv,endv, // start, end vertices for this node
    me, // my node number
    dbg;
unsigned largeint, // max possible unsigned int
    mymin[2], // mymin[0] is min for my chunk,
                // mymin[1] is vertex which achieves that min
        othermin[2], // othermin[0] is min over the other chunks
                        // (used by node 0 only)
                            // othermin[1] is vertex which achieves that min
        overallmin[2], // overallmin[0] is current min over all nodes,
                // overallmin[1] is vertex which achieves that min
        *ohd, // 1-hop distances between vertices; "ohd[i][j]" is
```

```
        // ohd[i*nv+j]
    *mind; // min distances found so far
double T1,T2; // start and finish times
void init(int ac, char **av)
{ int i,j,tmp; unsigned u;
    nv = atoi(av[1]);
    dbg = atoi(av[3]);
    MPI_Init(&ac,&av);
    MPI_Comm_size(MPI_COMM_WORLD,&nnodes);
    MPI_Comm_rank(MPI_COMM_WORLD,&me);
    chunk = nv/nnodes;
    startv = me * chunk;
    endv = startv + chunk - 1;
    u = -1;
    largeint = u >> 1;
    ohd = malloc(nv*nv*sizeof(int));
    mind = malloc(nv*sizeof(int));
    notdone = malloc(nv*sizeof(int));
    // random graph
    // note that this will be generated at all nodes; could generate just
    // at node 0 and then send to others, but faster this way
    srand(9999);
    for (i = 0; i < nv; i++)
            for (j = i; j < nv; j++) {
            if (j == i) ohd[i*nv+i] = 0;
            else {
                    ohd[nv*i+j] = rand() % 20;
                    ohd[nv*j+i] = ohd[nv*i+j];
            }
        }
    for (i = 0; i < nv; i++) {
        notdone[i] = 1;
        mind[i] = largeint;
    }
    mind[0] = 0;
    while (dbg) ; // stalling so can attach debugger
}
// finds closest to 0 among notdone, among startv through endv
void findmymin()
{ int i;
    mymin[0] = largeint;
    for (i = startv; i <= endv; i++)
        if (notdone[i] && mind[i] < mymin[0]) {
            mymin[0] = mind[i];
            mymin[1] = i;
        }
}
void findoverallmin()
{ int i;
    MPI_Status status; // describes result of MPI_Recv() call
    // nodes other than 0 report their mins to node 0, which receives
    // them and updates its value for the global min
    if (me > 0)
        MPI_Send(mymin,2,MPI_INT,0,MYMIN_MSG,MPI_COMM_WORLD);
```

```
    else {
        // check my own first
        overallmin[0] = mymin[0];
        overallmin[1] = mymin[1];
        // check the others
        for (i = 1; i < nnodes; i++) {
            MPI_Recv(othermin,2,MPI_INT,i,MYMIN_MSG,MPI_COMM_WORLD,&status);
            if (othermin[0] < overallmin[0]) {
                overallmin[0] = othermin[0];
                    overallmin[1] = othermin[1];
            }
        }
    }
}
void updatemymind() // update my mind segment
{ // for each i in [startv,endv], ask whether a shorter path to i
    // exists, through mv
    int i, mv = overallmin[1];
    unsigned md = overallmin[0];
    for (i = startv; i <= endv; i++)
        if (md + ohd[mv*nv+i] < mind[i])
            mind[i] = md + ohd[mv*nv+i];
}
void disseminateoverallmin()
{ int i;
    MPI_Status status;
    if (me == 0)
        for (i = 1; i < nnodes; i++)
            MPI_Send(overallmin,2,MPI_INT,i,OVRLMIN_MSG,MPI_COMM_WORLD);
    else
        MPI_Recv(overallmin,2,MPI_INT,0,OVRLMIN_MSG,MPI_COMM_WORLD,&status);
}
void updateallmind() // collects all the mind segments at node 0
{ int i;
    MPI_Status status;
    if (me > 0)
        MPI_Send(mind+startv, chunk,MPI_INT, 0, COLLECT_MSG,MPI_COMM_WORLD);
    else
        for (i = 1; i < nnodes; i++)
            MPI_Recv(mind+i*chunk, chunk,MPI_INT,i,COLLECT_MSG,MPI_COMM_WORLD,
                &status);
}
void printmind() // partly for debugging (call from GDB)
{ int i;
    printf("minimum distances:\n");
    for (i = 1; i < nv; i++)
        printf("%u\n",mind[i]);
}
void dowork()
{ int step, // index for loop of nv steps
        i;
    if (me == 0) T1 = MPI_Wtime();
    for (step = 0; step < nv; step++) {
```

```
        findmymin();
        findoverallmin();
        disseminateoverallmin();
        // mark new vertex as done
        notdone[overallmin[1]] = 0;
        updatemymind(startv,endv);
    }
    updateallmind();
    T2 = MPI_Wtime();
}
int main(int ac, char **av)
{ int i,j,print;
    init(ac,av);
    dowork();
    print = atoi(av[2]);
    if (print && me == 0) {
        printf("graph weights:\n");
        for (i = 0; i < nv; i++) {
            for (j = 0; j < nv; j++)
                printf("%u ",ohd[nv*i+j]);
            printf("\n");
        }
        printmind();
    }
    if (me == 0) printf("time at node 0: %f\n",(float)(T2-T1));
    MPI_Finalize();
}
```

The various MPI functions will be explained in the next section.

### 8.3.3 Introduction to MPI APIs

### 8.3.3.1 MPI_Init() and MPI_Finalize()

These are required for starting and ending execution of an MPI program. Their actions may be implementation-dependent. For instance, if our platform is an Ethernet-based cluster, MPI_Init() will probably set up the TCP/IP sockets via which the various nodes communicate with each other. On an Infiniband-based cluster, connections in the special Infiniband network protocol will be established. On a shared-memory multiprocessor, an implementation of MPI that is tailored to that platform would take very different actions.

### 8.3.3.2 MPI_Comm_size() and MPI_Comm_rank()

In our function init() above, note the calls

MPI_Comm_size(MPI_COMM_WORLD,\&nnodes) ;
MPI_Comm_rank(MPI_COMM_WORLD, \&me) ;
The first call determines how many nodes are participating in our computation, placing the result in our variable nnodes. Here MPI_COMM_WORLD is our node group, termed a communicator in MPI parlance. MPI allows the programmer to subdivide the nodes into groups, to facilitate performance and clarity of code. Note that for some operations, such as barriers, the only way to apply the operation to a proper subset of all nodes is to form a group. The totality of all groups is denoted by MPI_COMM_WORLD. In our program here, we are not subdividing into groups.

The second call determines this node's ID number, called its rank, within its group. As mentioned earlier, even though the nodes are all running the same program, they are typically working on different parts of the program's data. So, the program needs to be able to sense which node it is running on, so as to access the appropriate data. Here we record that information in our variable me.

### 8.3.3.3 MPI_Send()

To see how MPI's basic send function works, consider our line above,

MPI_Send(mymin, 2, MPI_INT, 0, MYMIN_MSG, MPI_COMM_WORLD);

Let's look at the arguments:
mymin: We are sending a set of bytes. This argument states the address at which these bytes begin.
2, MPI_INT: This says that our set of bytes to be sent consists of 2 objects of type MPI_INT. That means 8 bytes on 32 -bit machines, so why not just collapse these two arguments to one, namely the number 8? Why did the designers of MPI bother to define data types? The answer is that we want to be able to run MPI on a heterogeneous set of machines, with MPI serving as the "broker" between them in case different architectures among those machines handle data differently.
First of all, there is the issue of endianness. Intel machines, for instance, are little-endian, which means that the least significant byte of a memory word has the smallest address among bytes of the word. Sun SPARC chips, on the other hand, are big-endian, with the opposite storage scheme. If our set of nodes included machines of both types, straight transmission of sequences of 8 bytes might mean that some of the machines literally receive the data backwards!

Secondly, these days 64 -bit machines are becoming more and more common. Again, if our set of nodes were to include both 32 -bit and 64 -bit words, some major problems would occur if no conversion were done.
$\mathbf{0}$ : We are sending to node 0 .
MYMIN_MSG: This is the message type, programmer-defined in our line
\#define MYMIN_MSG 0

Receive calls, described in the next section, can ask to receive only messages of a certain type.
COMM_WORLD: This is the node group to which the message is to be sent. Above, where we said we are sending to node 0 , we technically should say we are sending to node 0 within the group MPI_COMM_WORLD.

### 8.3.3.4 MPI_Recv()

Let's now look at the arguments for a basic receive:

MPI_Recv(othermin, 2,MPI_INT,i,MYMIN_MSG,MPI_COMM_WORLD,\&status);
othermin: The received message is to be placed at our location othermin.
2,MPI_INT: Two objects of MPI_INT type are to be received.
i: Receive only messages of from node i. If we did not care what node we received a message from, we could specify the value MPI_ANY_SOURCE.

MYMIN_MSG: Receive only messages of type MYMIN_MSG. If we did not care what type of message we received, we would specify the value MPI_ANY_TAG.

OMM_WORLD: Group name.
status: Recall our line

MPI_Status status; // describes result of MPI_Recv() call

The type is an MPI struct containing information about the received message. Its primary fields of interest are MPI_SOURCE, which contains the identity of the sending node, and MPI_TAG, which contains the message type. These would be useful if the receive had been done with MPI_ANY_SOURCE or MPI_ANY_TAG; the status argument would then tell us which node sent the message and what type the message was.

### 8.4 Example: Removing 0s from an Array

```
#include <mpi.h>
#include <stdlib.h>
#define MAX_N 100000
#define MAXNPROCS 100
#define DATA_MSG 0
#define NEWDATAMSG 1
int nnodes, // number of MPI processes
    n, // size of original array
    me, // my MPI ID
    has0s[MAXN], // original data
    no0s[MAXN], // 0-free data
    nno0s; // number of non-0 elements
int debug;
init(int argc, char **argv)
{
    int i;
    MPI_Init(&argc,&argv );
    MPI_Comm_size (MPLCOMM_WORLD,& nnodes );
    MPI_Comm_rank (MPL_COMM_WORLD,&me);
    n = atoi(argv[1]);
    if (me=0) {
            for (i = 0; i < n; i++)
                has0s[i] = rand() % 4;
    } else {
            debug = atoi(argv[2]);
            while (debug) ;
    }
}
void managernode()
{
    MPI_Status status;
    int i;
    int lenchunk;
    lenchunk = n / (nnodes -1); // assumed divides evenly
    for (i = 1; i < nnodes; i++) {
            MPI_Send(has0s+(i-1)*lenchunk, lenchunk,
                MPI_INT, i ,DATA_MSG,MPLCOMM_WORLD );
    }
    int k = 0;
    for (i = 1; i < nnodes; i ++) {
            MPI_Recv(no0s+k,MAXN,
                    MPI_INT, i ,NEWDATAMSG,MPLCOMMLWORLD,& status );
```

```
        MPI_Get_count(&status,MPI_INT,&lenchunk);
        k += lenchunk;
    }
    nno0s = k;
}
void remov0s(int *oldx, int n, int *newx, int *nnewx)
{ int i, count = 0;
    for (i = 0; i < n; i++)
        if (oldx[i] != 0) newx[count++] = oldx[i];
    *nnewx = count;
}
void workernode()
{
    int lenchunk;
    MPI_Status status;
    MPI_Recv(has0s ,MAX_N,
        MPI_INT,0 ,DATA_MSG,MPLCOMMLWORLD,& status );
    MPI_Get_count(&status,MPI_INT,&lenchunk );
    remov0s(has0s,lenchunk, no0s,&nno0s);
    MPI_Send (no0s, nno0s,
            MPI_INT, ,0 ,NEWDATAMSG,MPLCOMMLWORLD );
}
int main(int argc,char **argv)
{
    int i;
    init(argc, argv);
    if (me = 0 && n < 25) {
        for (i = 0; i < n; i++) printf("%d ",has0s[i]);
        printf("\n");
    }
    if (me = 0) managernode();
    else workernode();
    if (me=0 && n < 25) {
        for (i = 0; i < n; i++) printf("%d ",no0s[i]);
        printf("\n");
    }
    MPI_Finalize();
}
```


### 8.5 Debugging MPI Code

If you are using GDB - either directly, or via an IDE such as Eclipse or Netbeans - the trick with MPI is to attach GDB to your running MPI processes.

Set up code like that we've seen in our examples here:

1 while (dbg) ;
This deliberately sets up an infinite loop of $\mathbf{d b g}$ is nonzero, for reasons to be discussed below.
For instance, suppose I'm running an MPI program a.out, on machines A, B and C. I would start the processes as usual, and have three terminal windows open. I'd $\log$ in to machine $A$, find the process number for a.out, using for example a command like ps ax on Unix-family systems, then attach GDB to that process. Say the process number is 88888 . I'd attach by running the command

```
% gdb a.out }8888
```

That would start GDB, in the midst of my already-running process, thus stuck in the infinite loop seen above. I hit ctrl-c to interrupt it, which gives me the GDB prompt, (gdb). I then type

```
(gdb) set var dbg = 0
```

which means when I next hit the command in GDB, the program will proceed, not stuck in the loop anymore. But first I set my breakpoints.

### 8.6 Collective Communications

MPI features a number of collective communication capabilities, a number of which are used in the following refinement of our Dijkstra program:

### 8.6.1 Example: Refined Dijkstra Code

```
// Dijkstra.coll1.c
// MPI example program: Dijkstra shortest-path finder in a
// bidirectional graph; finds the shortest path from vertex O to all
// others; this version uses collective communication
// command line arguments: nv print dbg
// where: nv is the size of the graph; print is 1 if graph and min
// distances are to be printed out, O otherwise; and dbg is 1 or 0, 1
// for debug
// node O will both participate in the computation and serve as a
// "manager"
#include <stdio.h>
#include <mpi.h>
// global variables (but of course not shared across nodes)
int nv, // number of vertices
```

```
    *notdone, // vertices not checked yet
    nnodes, // number of MPI nodes in the computation
    chunk, // number of vertices handled by each node
    startv,endv, // start, end vertices for this node
    me, // my node number
    dbg;
unsigned largeint, // max possible unsigned int
            mymin[2], // mymin[0] is min for my chunk,
            // mymin[1] is vertex which achieves that min
    overallmin[2], // overallmin[0] is current min over all nodes,
                                    // overallmin[1] is vertex which achieves that min
            *ohd, // 1-hop distances between vertices; "ohd[i][j]" is
            // ohd[i*nv+j]
            *mind; // min distances found so far
double T1,T2; // start and finish times
void init(int ac, char **av)
{ int i,j,tmp; unsigned u;
    nv = atoi(av[1]);
    dbg = atoi(av[3]);
    MPI_Init(&ac,&av);
    MPI_Comm_size(MPI_COMM_WORLD,&nnodes);
    MPI_Comm_rank(MPI_COMM_WORLD,&me);
    chunk = nv/nnodes;
    startv = me * chunk;
    endv = startv + chunk - 1;
    u = -1;
    largeint = u >> 1;
    ohd = malloc(nv*nv*sizeof(int));
    mind = malloc(nv*sizeof(int));
    notdone = malloc(nv*sizeof(int));
    // random graph
    // note that this will be generated at all nodes; could generate just
    // at node 0 and then send to others, but faster this way
    srand(9999);
    for (i = 0; i < nv; i++)
        for (j = i; j < nv; j++) {
            if (j == i) ohd[i*nv+i] = 0;
            else {
                    ohd[nv*i+j] = rand() % 20;
                    ohd[nv*j+i] = ohd[nv*i+j];
            }
        }
    for (i = 0; i < nv; i++) {
        notdone[i] = 1;
        mind[i] = largeint;
    }
    mind[0] = 0;
    while (dbg) ; // stalling so can attach debugger
}
// finds closest to O among notdone, among startv through endv
void findmymin()
{ int i;
    mymin[0] = largeint;
    for (i = startv; i <= endv; i++)
        if (notdone[i] && mind[i] < mymin[0]) {
```

```
        mymin[0] = mind[i];
        mymin[1] = i;
    }
}
void updatemymind() // update my mind segment
{ // for each i in [startv,endv], ask whether a shorter path to i
    // exists, through mv
    int i, mv = overallmin[1];
    unsigned md = overallmin[0];
    for (i = startv; i <= endv; i++)
        if (md + ohd[mv*nv+i] < mind[i])
            mind[i] = md + ohd[mv*nv+i];
}
void printmind() // partly for debugging (call from GDB)
{ int i;
    printf("minimum distances:\n");
    for (i = 1; i < nv; i++)
        printf("%u\n",mind[i]);
}
void dowork()
{ int step, // index for loop of nv steps
            i;
    if (me == 0) T1 = MPI_Wtime();
    for (step = 0; step < nv; step++) {
        findmymin();
        MPI_Reduce(mymin,overallmin,1,MPI_2INT,MPI_MINLOC, 0,MPI_COMM_WORLD);
        MPI_Bcast(overallmin,1,MPI_2INT,0,MPI_COMM_WORLD);
        // mark new vertex as done
        notdone[overallmin[1]] = 0;
        updatemymind(startv,endv);
    }
    // now need to collect all the mind values from other nodes to node 0
    MPI_Gather(mind+startv,chunk,MPI_INT,mind,chunk,MPI_INT,0,MPI_COMM_WORLD);
    T2 = MPI_Wtime();
}
int main(int ac, char **av)
{ int i,j,print;
    init(ac,av);
    dowork();
    print = atoi(av[2]);
    if (print && me == 0) {
            printf("graph weights:\n");
            for (i = 0; i < nv; i++) {
                for (j = 0; j < nv; j++)
                    printf("%u ",ohd[nv*i+j]);
            printf("\n");
        }
        printmind();
    }
    if (me == 0) printf("time at node 0: %f\n",(float)(T2-T1));
    MPI_Finalize();
}
```

The new calls will be explained in the next section.

### 8.6.2 MPI_Bcast()

In our original Dijkstra example, we had a loop
for (i = 1; i < nnodes; i++) MPI_Send(overallmin, 2,MPI_INT,i,OVRLMIN_MSG,MPI_COMM_WORLD) ;
in which node 0 sends to all other nodes. We can replace this by

MPI_Bcast (overallmin, 2,MPI_INT, 0,MPI_COMM_WORLD);

In English, this call would say,

At this point all nodes participate in a broadcast operation, in which node 0 sends 2 objects of type MPI_INT to each node (including itself). The source of the data will be located at address overallmin at node 0 , and the other nodes will receive the data at a location of that name.

Note my word "participate" above. The name of the function is "broadcast," which makes it sound like only node 0 executes this line of code, which is not the case; all the nodes in the group (in this case that means all nodes in our entire computation) execute this line. The only difference is the action; most nodes participate by receiving, while node 0 participates by sending.

Actually, this call to MPI_Bcast() is doing more than replacing the loop, since the latter had been part of an if-then-else that checked whether the given process had rank 0 or not.

Why might this be preferable to using an explicit loop? First, it would obviously be much clearer. That makes the program easier to write, easier to debug, and easier for others (and ourselves, later) to read.

But even more importantly, using the broadcast may improve performance. We may, for instance, be using an implementation of MPI which is tailored to the platform on which we are running MPI. If for instance we are running on a network designed for parallel computing, such as Myrinet or Infiniband, an optimized broadcast may achieve a much higher performance level than would simply a loop with individual send calls. On a shared-memory multiprocessor system, special machine instructions specific to that platform's architecture can be exploited, as for instance IBM has done for its shared-memory machines. Even on an ordinary Ethernet, one could exploit Ethernet's own broadcast mechanism, as had been done for PVM, a system like MPI (G. Davies and N.

Matloff, Network-Specific Performance Enhancements for PVM, Proceedings of the Fourth IEEE International Symposium on High-Performance Distributed Computing, 1995, 205-210; N. Matloff, Analysis of a Programmed Backoff Method for Parallel Processing on Ethernets, in Network-Based Parallel Computing).

### 8.6.3 MPI_Reduce()/MPI_Allreduce()

Look at our call

MPI_Reduce(mymin, overallmin, 1,MPI_2INT, MPI_MINLOC, $\left.0, M P I \_C O M M \_W O R L D\right) ;$
above. In English, this would say,

At this point all nodes in this group participate in a "reduce" operation. The type of reduce operation is MPI_MINLOC, which means that the minimum value among the nodes will be computed, and the index attaining that minimum will be recorded as well. Each node contributes a value to be checked, and an associated index, from a location mymin in their programs; the type of the pair is MPI_2INT. The overall min value/index will be computed by combining all of these values at node 0 , where they will be placed at a location overallmin.

MPI also includes a function MPI_Allreduce(), which does the same operation, except that instead of just depositing the result at one node, it does so at all nodes. So for instance our code above,

```
MPI_Reduce(mymin,overallmin,1,MPI_2INT,MPI_MINLOC,0,MPI_COMM_WORLD);
MPI_Bcast(overallmin,1,MPI_2INT,0,MPI_COMM_WORLD);
```

could be replaced by

```
MPI_Allreduce(mymin, overallmin,1,MPI_2INT,MPI_MINLOC,MPI_COMM_WORLD);
```

Again, these can be optimized for particular platforms.
Here is a table of MPI reduce operations:

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| ---: | ---: |
| MPI_MAX | max |
| MPI_MIN | min |
| MPI_SUM | sum |
| MPI_PROD | product |
| MPI_LAND | wordwise boolean and |
| MPI_LOR | wordwise boolean or |
| MPI_LXOR | wordwise exclusive or |
| MPI_BAND | bitwise boolean and |
| MPI_BOR | bitwise boolean or |
| MPI_BXOR | bitwise exclusive or |
| MPI_MAXLOC | max value and location |
| MPI_MINLOC | min value and location |

MPI_MAX
MPI_MIN

MPI_PROD
MPI_LAND
MPI_LOR
MPI_LXOR
MPI_BAND
MPI_BOR
MPI_MAXLOC MPI_MINLOC
max
min
sum
product
wordwise boolean and wordwise boolean or wordwise exclusive or bitwise boolean and bitwise boolean or bitwise exclusive or min value and location

### 8.6.4 MPI_Gather()/MPI_Allgather()

A classical approach to parallel computation is to first break the data for the application into chunks, then have each node work on its chunk, and then gather all the processed chunks together at some node. The MPI function MPI_Gather() does this.

In our program above, look at the line

```
MPI_Gather(mind+startv, chunk,MPI_INT,mind, chunk,MPI_INT, 0,MPI_COMM_WORLD);
```

In English, this says,
At this point all nodes participate in a gather operation, in which each node (including Node 0) contributes chunk number of MPI integers, from a location mind+startv in that node's program. Node 0 then receives chunk items sent from each node, stringing everything together in node order and depositing it all at mind in the program running at Node 0.
(Yes, the fifth argument is redundant with the second; same for the thrid and sixth.)
There is also MPI_Allgather(), which places the result at all nodes, not just one. Its call form is the same as MPI_Gather(), but with one fewer argument (since the identity of "the" gathering node is no longer meaningful):

```
int MPI_Allgather(srcbuf, srccount, srctype, destbuf, destcount, desttype, communicator)
```


### 8.6.5 The MPI_Scatter()

This is the opposite of MPI_Gather(), i.e. it breaks long data into chunks which it parcels out to individual nodes. For example, in the code in the next section, the call

MPI_Scatter (oh, lenchunk, MPI_INT, ohchunk, lenchunk, MPI_INT, 0, MPLCOMMLWORLD) ;
means

Node 0 will break up the array oh of type MPI_INT into chunks of length lenchunk, sending the $\mathrm{i}^{\text {th }}$ chunk to Node i, where lenchunk items will be deposited at ohchunk.

### 8.6.6 Example: Count the Number of Edges in a Directed Graph

Below is MPI code to count the number of edges in a directed graph. ("Directed" means that a link from i to j does not necessarily imply one from j to i .)

In the context here, me is the node's rank; $\mathbf{n v}$ is the number of vertices; $\mathbf{o h}$ is the one-hop distance matrix; and nnodes is the number of MPI processes. At the beginning only the process of rank 0 has a copy of oh, but it sends that matrix out in chunks to the other nodes, each of which stores its chunk in an array ohchunk.

```
lenchunk = nv / nnodes;
MPI_Scatter(oh, lenchunk, MPI_INT, ohchunk, lenchunk, MPI_INT, 0,
    MPLCOMMLWORLD);
mycount = 0;
for (i = 0; i < nv*nv/nnodes)
    if (ohchunk[i] != 0) mycount++;
MPI_Reduce(&mycount,&numedge ,1,MPI_INT,MPI_SUM, 0 ,MPLCOMMLWORLD);
if (me = 0) printf("there are %d edges\n",numedge);
```


### 8.6.7 Example: Cumulative Sums

Here we find cumulative sums. For instance, if the original array is (3,1,2,0,3,0,1,2), then it is changed to $(3,4,6,6,9,9,10,12)$. (This topic is pursued in depth in Chapter 10.)

```
// finds cumulative sums in the array x
#include <mpi.h>
#include <stdlib.h>
#define MAX_N 10000000
#define MAXNODES 10
int nnodes, // number of MPI processes
    n, // size of x
    me, // MPI rank of this node
    // full data for node 0, part for the rest
```

```
    x [MAXN],
    csums[MAXN], // cumulative sums for this node
    maxvals[MAXNODES]; // the max values at the various nodes
int debug;
init(int argc, char **argv)
{
    int i;
    MPI_Init(&argc,&argv );
    MPI_Comm_size (MPLCOMM_WORLD,& nnodes );
    MPI_Comm_rank (MPLCOMM_WORLD,&me);
    n = atoi(argv[1]);
    // test data
    if (me=0) {
        for (i = 0; i < n; i++)
            x[i] = rand() % 32;
    }
    debug = atoi(argv[2]);
    while (debug) ;
}
void cumulsums()
{
    MPI_Status status;
    int i, lenchunk,sum, node;
    lenchunk = n / nnodes; // assumed to divide evenly
    // note that node 0 will participate in the computation too
    MPI_Scatter(x,lenchunk,MPI_INT, x, lenchunk,MPI_INT,
        0,MPLCOMM_WORLD);
    sum = 0;
    for (i = 0; i < lenchunk; i++) {
        csums[i] = sum + x[i];
        sum += x[i];
    }
    MPI_Gather(&csums[lenchunk - 1], 1,MPI_INT ,
        maxvals,1,MPI_INT,0,MPLCOMM_WORLD);
    MPI_Bcast (maxvals, nnodes,MPI_INT ,0 ,MPLCOMM_WORLD);
    if (me > 0) {
        sum = 0;
        for (node = 0; node < me; node++) {
            sum += maxvals[node];
        }
        for (i = 0; i < lenchunk; i++)
            csums[i] += sum;
    }
    MPI_Gather (csums, lenchunk, MPI_INT, csums, lenchunk, MPI_INT,
        0,MPLCOMMLWORLD);
}
```

```
int main(int argc,char **argv)
{
    int i;
    init(argc, argv);
    if (me = 0 && n < 25) {
        for (i = 0; i < n; i++) printf("%d ",x[i]);
        printf("\n");
    }
    cumulsums ();
    if (me=0 && n < 25) {
            for (i = 0; i < n; i++) printf("%d ",csums[i]);
            printf("\n");
    }
    MPI_Finalize();
}
```


### 8.6.8 Example: an MPI Solution to the Mutual Outlinks Problem

Consider the example of Section 2.4.3. We have a network graph of some kind, such as Web links. For any two vertices, say any two Web sites, we might be interested in mutual outlinks, i.e. outbound links that are common to two Web sites.

The MPI code below finds the mean number of mutual outlinks, among all pairs of vertices in a graph.

```
// MPI solution to the mutual outlinks problem
// adjacency matrix m is global at each node, broadcast from node 0
// assumes m is nxn, and number of nodes is < n
// for each node i, check all possible pairing nodes j > i; the various
// nodes work on values of i in a Round Robin fashion, with node k
// handling all i for which i mod nnodes = k
#include <mpi.h>
#include <stdlib.h>
#define MAXLENGTH 10000000
int nnodes, // number of MPI processes
    n, // size of x
    me, // MPI rank of this node
    m[MAXLENGTH], // adjacency matrix
    grandtot; // grand total of all counts of mutuality
// get adjacency matrix, in this case just by simulation
void getm()
```

```
{ int i;
    for (i = 0; i < n*n; i++)
        m[i] = rand() % 2;
}
init(int argc, char **argv)
{
    int i;
    MPI_Init(&argc,&argv);
    MPI_Comm_size (MPLCOMM_WORLD,& nnodes );
    MPI_Comm_rank (MPLCOMM_WORLD,&me);
    n = atoi(argv[1]);
    if (me=0) {
        getm(); // get the data (app-specific)
    }
}
void mutlinks()
{
    int i,j,k,tot;
    MPI_Bcast(m, n*n,MPI_INT, 0 ,MPLCOMMLWORLD);
    tot = 0;
    for (i = me; i < n-1; i += nnodes) {
        for (j = i+1; j < n; j++) {
            for (k = 0; k < n; k++)
                tot +=m[twod2oned(n,i,k)] * m[twod2oned (n,j,k)];
        }
    }
    MPI_Reduce(&tot,&grandtot,1,MPI_INT,MPI_SUM,0 ,MPLCOMMLWORLD);
}
// convert 2-D subscript to 1-D
int twod2oned(n,i,j)
{ return n * i + j; }
int main(int argc,char **argv)
{ int i,j;
    init(argc, argv);
    if (me= 0 && n < 5) { // check test input
        for (i = 0; i < n; i++) {
            for (j = 0; j < n; j++) printf("%d ",m[twod2oned(n,i,j )]);
            printf("\n");
        }
    }
    mutlinks();
    if (me= 0) printf("%f\n",((float) grandtot)/(n*(n-1)/2));
    MPI_Finalize();
}
```


### 8.6.9 The MPI_Barrier()

This implements a barrier for a given communicator. The name of the communicator is the sole argument for the function.

Explicit barriers are less common in message-passing programs than in the shared-memory world.

### 8.6.10 Creating Communicators

Again, a communicator is a subset (either proper or improper) of all of our nodes. MPI includes a number of functions for use in creating communicators. Some set up a virtual "topology" among the nodes.

For instance, many physics problems consist of solving differential equations in two- or threedimensional space, via approximation on a grid of points. In two dimensions, groups may consists of rows in the grid.

Here's how we might divide an MPI run into two groups (assumes an even number of MPI processes to begin with):

```
MPI_Comm_size(MPI_COMM_WORLD,&nnodes);
MPI_Comm_rank(MPI_COMM_WORLD,&me);
...
// declare variables to bind to groups
MPI_Group worldgroup, subgroup;
// declare variable to bind to a communicator
MPI_Comm subcomm;
int i,startrank,nn2 = nnodes/2;
int *subranks = malloc(nn2*sizeof(int));
if (me < nn2) start = 0;
else start = nn2;
for (i = 0; i < nn2; i++)
    subranks[i] = i + start;
// bind the world to a group variable
MPI_Comm_group(MPI_COMM_WORLD, &worldgroup);
// take worldgroup the nn2 ranks in "subranks" and form group
// "subgroup" from them
MPI_Group_incl(worldgroup, nn2, subranks, subgroup);
// create a communicator for that new group
MPI_Comm_create(MPI_COMM_WORLD, subgroup, subcomm);
// get my rank in this new group
MPI_Group_rank (subgroup, &subme);
```

You would then use subcomm instead of MPI_COMM_WORLD whenever you wish to, say, broadcast, only to that group.

### 8.7 Buffering, Synchrony and Related Issues

As noted several times so far, interprocess communication in parallel systems can be quite expensive in terms of time delay. In this section we will consider some issues which can be extremely important in this regard.

### 8.7.1 Buffering, Etc.

To understand this point, first consider situations in which MPI is running on some network, under the TCP/IP protocol. Say an MPI program at node A is sending to one at node B.

It is extremely import to keep in mind the levels of abstraction here. The OS's TCP/IP stack is running at the Session, Transport and Network layers of the network. MPI-meaning the MPI internals-is running above the TCP/IP stack, in the Application layers at A and B. And the MPI user-written application could be considered to be running at a "Super-application" layer, since it calls the MPI internals. (From here on, we will refer to the MPI internals as simply "MPI.")

MPI at node A will have set up a TCP/IP socket to B during the user program's call to MPI_Init(). The other end of the socket will be a corresponding one at B. This setting up of this socket pair as establishing a connection between A and B. When node A calls MPI_Send(), MPI will write to the socket, and the TCP/IP stack will transmit that data to the TCP/IP socket at B. The TCP/IP stack at B will then send whatever bytes come in to MPI at B.

Now, it is important to keep in mind that in TCP/IP the totality of bytes sent by A to B during lifetime of the connection is considered one long message. So for instance if the MPI program at A calls MPI_Send() five times, the MPI internals will write to the socket five times, but the bytes from those five messages will not be perceived by the TCP/IP stack at B as five messages, but rather as just one long message (in fact, only part of one long message, since more may be yet to come).

MPI at B continually reads that "long message" and breaks it back into MPI messages, keeping them ready for calls to MPI_Recv() from the MPI application program at B. Note carefully that phrase, keeping them ready; it refers to the fact that the order in which the MPI application program requests those messages may be different from the order in which they arrive.

On the other hand, looking again at the TCP/IP level, even though all the bytes sent are considered one long message, it will physically be sent out in pieces. These pieces don't correspond to the pieces written to the socket, i.e. the MPI messages. Rather, the breaking into pieces is done for the purpose of flow control, meaning that the TCP/IP stack at A will not send data to the one at B if the OS at B has no room for it. The buffer space the OS at B has set up for receiving data is limited. As A is sending to B , the TCP layer at B is telling its counterpart at A when A is allowed to send more data.

Think of what happens the MPI application at B calls MPI_Recv(), requesting to receive from A, with a certain tag T. Say the first argument is named $\mathbf{x}$, i.e. the data to be received is to be deposited at $\mathbf{x}$. If MPI sees that it already has a message of tag $T$, it will have its MPI_Recv() function return the message to the caller, i.e. to the MPI application at B. If no such message has arrived yet, MPI won't return to the caller yet, and thus the caller blocks.

MPI_Send() can block too. If the platform and MPI implementation is that of the TCP/IP network context described above, then the send call will return when its call to the OS' write() (or equivalent, depending on OS) returns, but that could be delayed if the OS' buffer space is full. On the other hand, another implementation could require a positive response from B before allowing the send call to return.

Note that buffering slows everything down. In our TCP scenario above, MPI_Recv() at B must copy messages from the OS' buffer space to the MPI application program's program variables, e.g. $\mathbf{x}$ above. This is definitely a blow to performance. That in fact is why networks developed specially for parallel processing typically include mechanisms to avoid the copying. Infiniband, for example, has a Remote Direct Memory Access capability, meaning that A can write directly to $\mathbf{x}$ at B. Of course, if our implementation uses synchronous communication, with A's send call not returning until A gets a response from B, we must wait even longer.

Technically, the MPI standard states that MPI_Send( $\mathrm{x}, \ldots$. ) will return only when it is safe for the application program to write over the array which it is using to store its message, i.e. $\mathbf{x}$. As we have seen, there are various ways to implement this, with performance implications. Similarly, MPI_Recv( $\mathbf{y}, \ldots$ ) will return only when it is safe to read $\mathbf{y}$.

### 8.7.2 Safety

With synchronous communication, deadlock is a real risk. Say A wants to send two messages to B , of types U and V , but that B wants to receive V first. Then A won't even get to send V , because in preparing to send U it must wait for a notice from B that B wants to read U -a notice which will never come, because B sends such a notice for V first. This would not occur if the communication were asynchronous.

But beyond formal deadlock, programs can fail in other ways, even with buffering, as buffer space is always by nature finite. A program can fail if it runs out of buffer space, either at the sender or the receiver. See www.llnl.gov/computing/tutorials/mpi_performance/samples/unsafe.c for an example of a test program which demonstrates this on a certain platform, by deliberating overwhelming the buffers at the receiver.

In MPI terminology, asynchronous communication is considered unsafe. The program may run fine on most systems, as most systems are buffered, but fail on some systems. Of course, as long as you know your program won't be run in nonbuffered settings, it's fine, and since there is potentially
such a performance penalty for doing things synchronously, most people are willing to go ahead with their "unsafe" code.

### 8.7.3 Living Dangerously

If one is sure that there will be no problems of buffer overflow and so on, one can use variant send and receive calls provided by MPI, such as MPI_Isend() and MPI_Irecv(). The key difference between them and MPI_Send() and MPI_Recv() is that they return immediately, and thus are termed nonblocking. Your code can go on and do other things, not having to wait.

This does mean that at A you cannot touch the data you are sending until you determine that it has either been buffered somewhere or has reached $\mathbf{x}$ at B. Similarly, at B you can't use the data at $\mathbf{x}$ until you determine that it has arrived. Such determinations can be made via MPI_Wait(). In other words, you can do your send or receive, then perform some other computations for a while, and then call MPI_Wait() to determine whether you can go on. Or you can call MPI_Probe() to ask whether the operation has completed yet.

### 8.7.4 Safe Exchange Operations

In many applications A and B are swapping data, so both are sending and both are receiving. This too can lead to deadlock. An obvious solution would be, for instance, to have the lower-rank node send first and the higher-rank node receive first.

But a more convenient, safer and possibly faster alternative would be to use MPI's MPI_Sendrecv() function. Its prototype is

```
intMPI_Sendrecv_replace(void* buf, int count, MPI_Datatype datatype,
    int dest, int sendtag, int source, int recvtag, MPI_Comm comm,
    MPI_Status *status)
```

Note that the sent and received messages can be of different lengths and can use different tags.

### 8.8 Use of MPI from Other Languages

MPI is a vehicle for parallelizing $\mathrm{C} / \mathrm{C}++$, but some clever people have extended the concept to other languages, such as the cases of Python and R that we treat in Chapters ?? and ??.

### 8.9 Other MPI Examples in This Book

- The pipelined prime number finder in Chapter 1 .
- Bucket sort with sampling, in Section 12.5 .


## Chapter 9

## Cloud Computing

In cloud computing, the idea is that a large corporation that has many computers could sell time on them, for example to make profitable use of excess capacity. The typical customer would have occasional need for large-scale computing - and often large-scale data storage. The customer would submit a program to the cloud computing vendor, who would run it in parallel on the vendor's many machines (unseen, thus forming the "cloud"), then return the output to the customer.

Google, Yahoo! and Amazon, among others, have recently gotten into the cloud computing business. Moreover, universities, businesses, research labs and so on are setting up their own small clouds, typically on clusters (a bunch of computers on the same local network, possibly with central controlling software for job management).

The paradigm that has become standard in the cloud today is MapReduce, developed by Google. In rough form, the approach is as follows. Various nodes server as mappers, and others serve as reducers ${ }^{\text {१ }}$

The terms map and reduce are in the functional programming sense. In the case of reduce, the idea is similar to reduction oeprations we've seen earlier in this book, such as the reduction clause in OpenMP and MPI_Reduce() for MPI. So, reducers in Hadoop perform operations such as summation, finding minima or maxima, and so on.

In this chapter we give a very brief introduction to Hadoop, today's open-source application of choice of MapReduce.

[^41]
### 9.1 Platforms and Modes

In terms of platforms, Hadoop is basically a Linux product. Quoting from the Hadoop Quick Start, http://hadoop.apache.org/common/docs/r0.20.2/quickstart.html\#Supported+Platforms:

Supported Platforms:

- GNU/Linux is supported as a development and production platform. Hadoop has been demonstrated on GNU/Linux clusters with 2000 nodes.
- Win32 is supported as a development platform. Distributed operation has not been well tested on Win32, so it is not supported as a production platform.

Hadoop runs in one of three modes, of varying degrees of parallelism:

- standalone mode: Single mapper, single reducer, mainly useful for testing.
- pseudo-distributed mode: Single node, but multiple mapper and reducer threads.
- fully-distributed mode: Multiple nodes, multiple mappers and reducers.


### 9.2 Overview of Operations

Here is a sumary of how a Hadoop application runs:

- Divide the input data into chunks of records. (In many cases, this is already the case, as a very large data file might be distributed across many machines.)
- Send the chunks to the mappers.
- Each mapper does some transformation (a "map," in functional programming terms) to each record in its chunks, and sends the output records to Hadoop.
- Hadoop collects the transformed records, splits them into chunks, sorts them, and then sends the chunks to the reducers.
- Each reducer does some summary operation (functional programming "reduce"), producing output records.
- Hadoop collects all those output records and concatenates them, producing the final output.


### 9.3 Role of Keys

The sorting operation, called the shuffle phase, is based on a key defined by the programmer. The key defines groups. If for instance we wish to find the total number of men and women in a certain debate, the key would be gender. The reducer would do addition, in this case adding 1 s , one 1 for each person, but keeping a separate sum for each gender.

During the shuffle stage, Hadoop sends all records for a given key, e.g. all men, to one reducer. In other words, records for a given key will never be split across multiple reducers. (Keep in mind, though, that typically a reducer will have the records for many keys.)

### 9.4 Hadoop Streaming

Actually Hadoop is really written for Java or C++ applications. However, Hadoop can work with programs in any language under Hadoop's Streaming option, by reading from STDIN and writing to STDOUT, in text, line-oriented form in both cases. In other words, any executable program, be it Java, C/C++, Python, R, shell scripts or whatever, can run in Hadoop in streaming mode.

Everything is text-file based. Mappers input lines of text, and output lines of text. Reducers input lines of text, and output lines of text. The final output is lines of text.

Streaming mode may be less efficient, but it is simple to develop programs in it, and efficient enough in many applications. Here we present streaming mode.

So, STDIN and STDOUT are key in this mode, and as mentioned earlier, input and output are done in terms of lines of text. One additional requirement, though, is that the line format for both mappers and reducers must be

```
key \t value
```

where $\backslash t$ is the TAB character.
The usage of text format does cause some slowdown in numeric programs, for the conversion of strings to numbers and vice versa, but again, Hadoop is not designed for maximum efficiency.

### 9.5 Example: Word Count

The typical introductory example is word count in a group of text files. One wishes to determine what words are in the files, and how many times each word appears. Let's simplify that a bit, so that we simply want a count of the number of words in the files, not an individual count for each word.

The initial input is the lines of the files (combined internally by Hadoop into one superfile). The mapper program breaks a line into words, and emits (key,value) pairs in the form of ( 0,1 ). Our key here, 0 , is arbitrary and meaningless, but we need to have one.

In the reducer stage, all those (key,value) pairs get sorted by the Hadoop internals (which has no effect in this case), and then fed into the reducers. Since there is only one key, 0 , only one reducer will actually be involved. The latter adds up all its input values, i.e. all the 1s, yielding a grand total number of words in all the files.

Here's the pseudocode:

## mapper:

```
for each line in STDIN
    break line into words, placed in wordarray
    for each word in wordarray
            # we have found 1 word
            print '0', '1' to STDOUT
```

reducer:

```
count = 0
for each line in STDIN
    split line into (key,value) # i.e. (0,1) here
    count += value # i.e. add 1 to count
print count
```

In terms of the key 0 , the final output tells us how many words there were of type 0 . Since we arbitrarily considered all words to be of type 0 , the final output is simply an overall word count.

### 9.6 Example: Maximum Air Temperature by Year

A common Hadoop example on the Web involves data with the format for each line

```
year month day high_temperature air_quality
```

It is desired to find the maximum air temperature for each year.
The code would be very similar to the outline in the word count example above, but now we have a real key - the year. So, in the end, the final output will be a listing of maximum temperature by year.

Our mapper here will simply be a text processor, extracting year and temperature from each line of text. (This is a common paradigm for Hadoop applications.) The reducer will then do the maximum-finding work.

Here is the pseudocode:

## mapper:

```
for each line in STDIN
    extract year and temperature
    print year, temperature to STDOUT
```

We have to be a bit more careful in the case of the reducer. Remember, though no year will be split across reducers, each reducer will likely receive the data for more than one year. It needs to find and output the maximum temperature for each of those years..$^{2}$

Since Hadoop internals sort the output of the mappers by key, our reducer code can expect a bunch of records for one year, then a bunch for another year and so on. So, as the reducer goes through its input line by line, it needs to detect when one bunch ends and the next begins. When such an event occurs, it outputs the max temp for the bunch that just ended.

Here is the pseudocode:

## reducer:

```
currentyear = NULL
currentmax = "-infinity"
for each line in STDIN
    split line into year, temperature
    if year = currentyear: # still in the current bunch
            currentmax = max(currentmax,temperature)
        else: # encountered a new bunch
            # print summary for previous bunch
            if currentyear not NULL:
                print currentyear, currentmax
        # start our bookkeeping for the new bunch
        currentyear = year
        currentmax = temperature
print currentyear, currentmax
```


### 9.7 Role of Disk Files

Hadoop has its own file system, HDFS, which is built on top of the native OS' file system of the machines.

Very large files are possible, in some cases spanning more than one disk/machine. Indeed, this is the typical goal of Hadoop-to easily parallelize operations on a very large database. Files are

[^42]typically gigabytes or terabytes in size. Moreover, there may be thousands of clusters, and millions of files.

This raises serious reliability issues. Thus HDFS is replicated, with each HDFS block existing in at least 3 copies, i.e. on at least 3 separate disks.

Disk files play a major role in Hadoop programs:

- Input is from a file in the HDFS system.
- The output of the mappers goes to temporary files in the native OS' file system.
- Final output is to a file in the HDFS system. As noted earlier, that file may be distributed across several disks/machines.

Note that by having the input and output files in HDFS, we minimize communications costs in shipping the data. The slogan used is "Moving computation is cheaper than moving data."

### 9.8 The Hadoop Shell

The HDFS can be accessed via a set of Unix-like commands. For example,
1 hadoop fs $-m k d i r$ somedir
creates a file somedir in my HDFS (invisible to me). Then
1 hadoop fs -put gy* somedir
copies all my files whose names begin with "gy" to somedir, and
1 hadoop fs -1 s somedir
lists the file names in the directory somedir.
See http://hadoop.apache.org/common/ for a list of available commands.

### 9.9 Running Hadoop

You run the above word count example something like this, say on the UCD CSIF machines.
Say my input data is in the directory indata on my HDFS, and I want to write the output to a new directory outdata. Say I've placed the mapper and reducer programs in my home directory (non-HDFS). I could then run

```
\$ hadoop jar \}
    /usr/local/hadoop -0.20 .2 / contrib/streaming/hadoop \(-0.20 .2-\) streaming.jar \(\backslash\)
    -input indata -output outdata \}
    -mapper mapper.py -reducer reducer.py \}
    -file /home/matloff/mapper.py \}
    -file /home/matloff/reducer.py
```

This tells Hadoop to run a Java .jar file, which in our case here contains the code to run streamingmode Hadoop, with the specified input and output data locations, and with the specified mapper and reducer functions. The -file flag indicates the locations of those functions (not needed if they are in my shell search path).

I could then run

```
1 \text { hadoop fs -ls outdata}
```

to see what files were produced, say part_00000, and then type
1 hadoop fs -cat outdata/part_00000
to view the results.
Note that the .py files must be executable, both in terms of file permissions and in terms of invoking the Python interpreter, the latter done by including

```
#!/usr/bin/env python
```

as the first line in the two Python files.

### 9.10 Example: Transforming an Adjacency Graph

Yet another rendition of the app in Section 4.13, but this time with a bit of problem, which will illustrate a limitation of Hadoop.

To review:
Say we have a graph with adjacency matrix

$$
\left(\begin{array}{llll}
0 & 1 & 0 & 0  \tag{9.1}\\
1 & 0 & 0 & 1 \\
0 & 1 & 0 & 1 \\
1 & 1 & 1 & 0
\end{array}\right)
$$

with row and column numbering starting at 0 , not 1 . We'd like to transform this to a two-column
matrix that displays the links, in this case

$$
\left(\begin{array}{ll}
0 & 1  \tag{9.2}\\
1 & 0 \\
1 & 3 \\
2 & 1 \\
2 & 3 \\
3 & 0 \\
3 & 1 \\
3 & 2
\end{array}\right)
$$

Suppose further that we require this listing to be in lexicographical order, sorted on source vertex and then on destination vertex.

At first, this seems right up Hadoop's alley. After all, Hadoop does sorting for us within groups automatically, and we could set up one group per row of the matrix, in other words make the row number the key.

We will actually do this below, but there is a fundamental problem: Hadoop's simple elegance hinges on there being an independence between the lines in the input file. We should be able to process them one line at a time, independently of other lines.

The problem with this is that we will have many mappers, each reading only some rows of the adjacency matrix. Then for any given row, the mapper handling that row doesn't know what row number this row had in the original matrix. So we have no key!

The solution is to add a column to the matrix, containing the original row numbers. The matrix above, for instance, would become

$$
\left(\begin{array}{lllll}
0 & 0 & 1 & 0 & 0  \tag{9.3}\\
1 & 1 & 0 & 0 & 1 \\
2 & 0 & 1 & 0 & 1 \\
3 & 1 & 1 & 1 & 0
\end{array}\right)
$$

Adding this column may be difficult to do, if the matrix is very large and already distributed over many machines. Assuming we do this, though, here is the mapper code (real code this time, not pseudocode) $3^{3}$.

```
#!/usr/bin/env python
# map/reduce pair inputs a graph adjacency matrix and outputs a list of
# links; if say row 3, column 8 of the input is 1, then there will be a
```

[^43]```
# row (3,8) in the final output
import sys
for line in sys.stdin:
        tks = line.split() # get tokens
        srcnode = tks[0]
        links = tks[1:]
        for dstnode in range(len(links)):
            if links[dstnode] =
                toprint = '%s\t%s, % (srcnode, dstnode)
            print toprint
```

And the reducer code:

```
#!/usr/bin/env python
import sys
for line in sys.stdin:
    line = line.strip()
    print line # could remove the \t
```

Note that the row number, needed for other reasons, is also serving as our Hadoop key variable.
Recall that in the word count and yearly temperature examples above, the reducer did the main work, with the mappers playing only a supporting role. In this case here, it's the opposite, with the reducers doing little more than printing what they're given. However, keep in mind that Hadoop itself did a lot of the work, with its shuffle phase, which produced the sorting that we required in the output.

Here's the same code in R:

## mapper:

```
#!/usr/bin/env Rscript
# map/reduce pair inputs a graph adjacency matrix and outputs a list of
# links; if say row 3, column 8 of the input is 1, then there will be a
# row (3,8) in the final output
con <- file("stdin", open = "r")
mapin <- readLines(con) # better not to read all at once, but keep simple
for (line in mapin) {
    tks <- strsplit(line,split=" ")
    tks <- tks[[1]]
    srcnode <- tks[1]
    links <- tks[-1]
    for (dstnode in 1:length(links)) {
            if (links[dstnode] = '1')
```

```
        cat(srcnode,"\t",dstnode,"\n",sep="")
    }
}
```

reducer:

```
#!/usr/bin/env Rscript
    con <- file("stdin", open = "r")
    mapin <- readLines(con)
    for (line in mapin) {
        line <- strsplit(line, split="\t") # remove \t
        line <- line[[1]]
        cat(line,"\n")
}
```


### 9.11 Example: Identifying Outliers

In any large data set, there are various errors, say 3 -year-olds who are listed as 7 feet tall. One way to try to track these down is to comb the data for outliers, which are data points (rows in the data set) that are far from the others. These may not be erroneous, but they are "suspicious," and we want to flag them for closer inspection.

In this simple version, we will define an outlier point to be one for which at least one of its variables is in the upper p proportion in its group. Say for example p is 0.02 , and our groups are male adults and female adults, with our data variables being height and weight. Then if the height for some man were in the upper $2 \%$ of all men in the data set, we'd flag him as an outlier; we'd do the same for weight. Note that he'd be selected if either his height or his weight were in the top $2 \%$ for that variable among men in the data set. Of course, we could also look at the bottom $2 \%$, or those whose Euclidean distance as vectors are in the most distant $2 \%$ from the centroid of the data in a group, etc.

We'll define groups in terms of combinations of variables. These might be, say, Asian male lawyers, female Kentucky natives registered as Democrats, etc.

## mapper:

```
#!/usr/bin/env Rscript
# map/reduce pair inputs a data matrix, forms groups according to
# combinations of specified integer variables, and then outputs the
# indices of outliers in each group
# any observation that has at least one variable in the top p% of its
# group is considered an outlier
```

```
9
# the first column of the input matrix is the observation number,
# starting at 0
# the group number for an observation is the linear index in
# lexicographical terms
# in Hadoop command line, use -mapper "olmap.R '0.4 2 3 6 16'" to
# specify parameters:
# p (given as a decimal number, e.g. 0.02); d, the number
# of data variables; g, the number of grouping variables; and finally g
# numbers which are the upper bounds for the last g-1 group variables
# (the lower bounds are always assumed to be 0)
init <- function() {
    ca <- commandArgs(trailingOnly=T)
    pars <- ca[1]
    pars <- strsplit(pars, split=" ")[[1]]
    pars <- pars[-1]
    ndv <<- as.integer(pars[1])
    ngv <<- as.integer (pars[2])
    # a few position variables, used in findgrp() below
    grpstart <<- 2
    grpend <<- grpstart + ngv - 1
    datastart <<- 2+ngv
    dataend <<- 1 + ngv + ndv
    # get upper bounds, and their reverse-cumulative products
    ubds <<- as.integer(pars[3:(1+ngv)])
    ubdsprod <<- vector(length=ngv-1)
    for (i in 1:(ngv-1))
        ubdsprod[i] <<- prod(ubds[i:(ngv-1)])
}
# converts vector of group variables to group number
findgrp <- function(grpvars) {
    sum <- 0
    for (i in 1:(ngv-1)) {
        m<- grpvars[i]
        sum <- sum + m * ubdsprod[i]
    }
    return(sum+grpvars[ngv])
}
# test
init()
con <- file("stdin", open = "r")
mapin <- readLines(con) # better not to read all at once
for (line in mapin) {
    tks <- strsplit(line, split=" ")
```

```
    tks <- tks[[1]]
    rownum <- tks[1]
    grpvars <- tks[grpstart:grpend]
    grpvars <- as.integer(grpvars)
    grpnum <- findgrp(grpvars)
    cat(grpnum,"\t",rownum," ",tks[datastart:dataend],"\n")
}
```


## reducer:

```
#!/usr/bin/env Rscript
# see comments in olmap.R
# in Hadoop command line, use -reducer "olred.R '0.4 2 3 6 16'" or
# similar
# R's quantile() too complicated
quantl <- function(x,q) {
    return(sort(x)[ceiling(length(x)*q)])
}
init <- function() {
    ca <- commandArgs(trailingOnly=T)
    pars <- ca[1]
    pars <- strsplit(pars, split=" ")[[1]]
    # pars <- c("0.4","2","3","6","16") # for little test
    # pars <- c("0.1","2","2","3") # for big test
    p <<- as.double(pars[1])
    ndv <<- as.integer(pars[2])
}
emitoutliers <- function(datamat) {
    # find the upper p quantile for each variable (skip row number)
    toohigh <- apply(datamat[, -(1:2), drop=F],2,quantl,1-p)
    for (i in 1:nrow(datamat)) {
            if (any(datamat[i, -(1:2)] >= toohigh))
                cat(datamat [i,],"\n")
    }
}
# test
init()
con <- file("stdin", open = "r")
mapin <- readLines(con)
oldgrpnum <- -1
nmapin <- length(mapin)
for (i in 1:nmapin) {
    line <- mapin[i]
    tks <- strsplit(line,split="\t")[[1]]
```

```
    grpnum <- as.integer(tks[1])
    tmp <- strsplit(tks[2], split=" ")[[1]]
    tmp <- tmp[tmp != ""] # delete empty element
    row <- as.numeric (tmp)
    if (oldgrpnum =-1) {
        datamat <- matrix(c(grpnum, row), nrow=1)
        oldgrpnum <- grpnum
    } else if (grpnum = oldgrpnum) {
    datamat <- rbind(datamat, c(grpnum,row))
    if (i = nmapin) emitoutliers(datamat)
} else { # new group
    emitoutliers(datamat)
    datamat <- matrix(c(grpnum, row), nrow=1)
    oldgrpnum <- grpnum
}
```

\}

### 9.12 Debugging Hadoop Streaming Programs

One advantage of the streaming approach is that mapper and reducer programs can be debugged via normal tools, since those programs can be run on their own, without Hadoop, simply by using the Unix/Linux pipe capability.

This all depends on the fact that Hadoop essentially does the following Unix shell computation:

```
cat inputfile | mapperprog | sort -n | reducerprog
```

(Omit the $\mathbf{- n}$ if the key is a string.)
The mapper alone works like

```
cat inputfile mapperprog
```

You thus can use whatever debugging tool you favor, to debug the mapper and reducer code separately.

Note, though, that the above pipe is not quite the same as Hadoop. the pipe doesn't break up the data, and there may be subtle problems arising as a result. But overall, the above approach provides a quick and easy first attempt at debugging.

The userlogs subdirectory of your Hadoop logs directory contains files that may be helpful, such as stderr.

### 9.13 It's a Lot More Than Just Programming

The real challenge in Hadoop is often not the programming, but rather the minimization of overhead. This involves things like tuning the file system, the number of mappers and reducers, and so on. These topics are beyond the scope of this book.

## Chapter 10

## The Parallel Prefix Problem

An operation that arises in a variety of parallel algorithms is that of prefix (or scan). In its abstract form, it inputs a sequence of objects $\left(x_{0}, \ldots, x_{n-1}\right)$, and outputs $\left(s_{0}, \ldots, s_{n-1}\right)$, where

$$
\begin{gather*}
s_{0}=x_{0} \\
s_{1}=x_{0} \otimes x_{1}  \tag{10.1}\\
\ldots, \\
s_{n-1}=x_{0} \otimes x_{1} \otimes \ldots \otimes x_{n-1}
\end{gather*}
$$

where $\otimes$ is some associative operator.
That's pretty abstract. The most concrete example would be that in which $\otimes$ is + and the objects are numbers. The scan of $(12,5,13)$ would then be $(12,12+5,12+5+13)=(12,17,30)$.

This is called an inclusive scan, in which $x_{i}$ is included in $s_{i}$. The exclusive version of the above example would be $(0,12,17)$.

Prefix scan has become a popular tool in the parallel processing community, applicable in a surprising variety of situations. Various examples will arise in succeeding chapters, but we'll present one in the next section in order to illustrate the versatility of the prefix approach.

### 10.1 Example: Permutations

Say we have the vector $(12,5,13,8,88)$. Applying the permutation $(2,0)$ would say the old element 0 becomes element 2 , the old element 2 becomes element 0 , and all the rest stay the same. The result would be $(13,5,12,8,88)$. If we then applied the permutation $(1,2,4)$, it would mean that element 1
goes to position 2,2 goes to 4 , and 4 goes to 1 , with everything else staying put. Our new vector would then be ( $13,88,5,8,12$ ).

This too can be cast in matrix terms, by representing any permutation as a matrix multiplication. We just apply the permutation to the identity matrix I, and then postmultiply the (row) vector by the matrix. For instance, the matrix corresponding to the permutation $(1,2,4)$ is

$$
\left(\begin{array}{lllll}
0 & 0 & 1 & 0 & 0  \tag{10.2}\\
0 & 1 & 0 & 0 & 0 \\
1 & 0 & 0 & 0 & 0 \\
0 & 0 & 0 & 1 & 0 \\
0 & 0 & 0 & 0 & 1
\end{array}\right)
$$

so applying $(1,2,4)$ to $(12,5,13,8,88)$ above can be done as

$$
(12,5,13,8,88)\left(\begin{array}{ccccc}
0 & 0 & 1 & 0 & 0  \tag{10.3}\\
0 & 1 & 0 & 0 & 0 \\
1 & 0 & 0 & 0 & 0 \\
0 & 0 & 0 & 1 & 0 \\
0 & 0 & 0 & 0 & 1
\end{array}\right)=(13,5,12,8,88)
$$

So in terms of 10.1, $x_{0}$ would be the identity matrix, $x_{i}$ for $\mathrm{i}>0$ would be the $\mathrm{i}^{\text {th }}$ permutation matrix, and $\otimes$ would be matrix multiplication.

Note, however, that although we've couched the problem in terms of matrix multiplication, these are sparse matrices, i.e. have many 0 s. Thus a general parallel matrix-multiply routine may not be efficient, and special parallel methods for sparse matrices should be used (Section 11.7).

Note that the above example shows that in finding a scan,

- the elements might be nonscalars
- the associative operator need not be commutative


### 10.2 General Strategies for Parallel Scan Computation

For the time being, we'll assume we have n threads, i.e. one for each datum. Clearly this condition will often not hold, so we'll extend things later.

We'll describe what is known as a data parallel solution to the prefix problem.

Here's the basic idea, say for $\mathrm{n}=8$ :
Step 1:

$$
\begin{align*}
& x_{1} \leftarrow x_{0}+x_{1}  \tag{10.4}\\
& x_{2} \leftarrow x_{1}+x_{2}  \tag{10.5}\\
& x_{3} \leftarrow x_{2}+x_{3}  \tag{10.6}\\
& x_{4} \leftarrow x_{3}+x_{4}  \tag{10.7}\\
& x_{5} \leftarrow x_{4}+x_{5}  \tag{10.8}\\
& x_{6} \leftarrow x_{5}+x_{6}  \tag{10.9}\\
& x_{7} \leftarrow x_{6}+x_{7} \tag{10.10}
\end{align*}
$$

## Step 2:

$$
\begin{align*}
& x_{2} \leftarrow x_{0}+x_{2}  \tag{10.11}\\
& x_{3} \leftarrow x_{1}+x_{3}  \tag{10.12}\\
& x_{4} \leftarrow x_{2}+x_{4}  \tag{10.13}\\
& x_{5} \leftarrow x_{3}+x_{5}  \tag{10.14}\\
& x_{6} \leftarrow x_{4}+x_{6}  \tag{10.15}\\
& x_{7} \leftarrow x_{5}+x_{7} \tag{10.16}
\end{align*}
$$

## Step 3:

$$
\begin{align*}
& x_{4} \leftarrow x_{0}+x_{4}  \tag{10.17}\\
& x_{5}  \tag{10.18}\\
& \leftarrow x_{1}+x_{5}  \tag{10.19}\\
& x_{6} \leftarrow x_{2}+x_{6}  \tag{10.20}\\
& x_{7} \leftarrow x_{3}+x_{7}
\end{align*}
$$

In Step 1, we look at elements that are 1 apart, then Step 2 considers the ones that are 2 apart, then 4 for Step 3.

Why does this work? Well, consider how the contents of $x_{7}$ evolve over time. Let $a_{i}$ be the original $x_{i}, \mathrm{i}=0,1, \ldots, \mathrm{n}-1$. Then here is $x_{7}$ after the various steps:

| step | contents |
| ---: | ---: |
| 1 | $a_{6}+a_{7}$ |
| 2 | $a_{4}+a_{5}+a_{6}+a_{7}$ |
| 3 | $a_{0}+a_{1}+a_{2}+a_{3}+a_{4}+a_{5}+a_{6}+a_{7}$ |

Similarly, after Step 3, the contents of $x_{7}$ will be $a_{0}+a_{1}+a_{2}+a_{3}+a_{4}+a_{5}+a_{6}$ (check it!). So, in the end, the locations of $x_{i}$ will indeed contain the prefix sums.

For general n, the routing is as follows. At Step i, each $x_{j}$ is routed both to itself and to $x_{j+2^{i-1}}$, for $j>=2^{i-1}$. (Some threads, more in each successive step, are idle.)

There will be $\log _{2} n$ steps, or if n is not a power of 2 , case the number of steps is $\left\lfloor\log _{2} n\right\rfloor$.
Note two important points:

- The location $x_{i}$ appears both as an input and an output in the assignment operations above. In our implementation, we need to take care that the location is not written to before its value is read. One way to do this is to set up an auxiliary array $y_{i}$. In odd-numbered steps, the $y_{i}$ are written to with the $x_{i}$ as inputs, and vice versa for the even-numbered steps.
- As noted above, as time goes on, more and more threads are idle. Thus load balancing is poor.
- Synchronization at each step incurs overhead in a multicore/multiprocessr setting. (Worse for GPU if multiple blocks are used).

Now, what if $n$ is greater than p , our number of threads? Let Ti denote thread i. The standard approach is that taken in Section 5.10.

```
break the array into p blocks
parallel for i = 0,\ldots,p-1
    Ti does scan of block i, resulting in Si
form new array G of rightmost elements of each Si
do parallel scan of G
parallel for i = 1,\ldots,p-1
    Ti adds Gi to each element of block i
```

For example, say we have the array

2252685031117102910
and three threads. We break the data into three sections,

| 225268 | 503111 | 792910 |
| :--- | :--- | :--- |

and then apply a scan to each section:

| 2275361 | 50535465 | 7164555 |
| :--- | :--- | :--- |

But we still don't have the scan of the array overall. That 50 , for instance, should be $61+50=$ 111 and the 53 should be $61+53=114$. In other words, 61 must be added to that second section, $(50,53,54,65)$, and $61+65=126$ must be added to the third section, $(7,16,45,55)$. This then is the last step, yielding

| 2275361 | 111114115126 | 133142171181 |
| :--- | :--- | :--- |

Another possible approach would be make n "fake" threads FTj . Each Ti plays the role of n/p of the FTj . The FTj then do the parallel scan as at the beginning of this section. Key point: Whenever a Ti becomes idle, it is assigned to help other Tk.

### 10.3 Implementations

The MPI standard actually includes built-in parallel prefix functions, MPI_Scan(). A number of choices are offered for $\otimes$, such as maximum, minimum, sum, product etc.

The Thrust library for CUDA or OpenMP includes functions thrust::inclusive_scan() and thrust::exclusive_sca
The CUDPP (CUDA Data Parallel Primitives Library) package contains CUDA functions for sorting and other operations, many of which are based on parallel scan. See http://gpgpu. org/developer/cudpp for the library code, and a detailed analysis of optimizing parallel prefix in a GPU context in the book GPU Gems 3, available either in bookstores or free online at http://developer.nvidia.com/object/gpu_gems_home.html.

### 10.4 Example: Parallel Prefix, Run-Length Decoding in OpenMP

Here an OpenMP implementation of the approach described at the end of Section 10.2, for addition:

```
#include <omp.h>
// calculates prefix sums sequentially on u, in-place, where u is an
// m-element array
void seqprfsum(int *u,int m)
{ int i, s=u[0];
    for (i = 1; i < m; i++) {
        u[i] += s;
        s}=\textrm{u}[\textrm{i}]
    }
}
// OMP example, calculating prefix sums in parallel on the n-element
```

```
// array x, in-place; for simplicity, assume that n is divisible by the
// number of threads; z is for intermediate storage, an array with length
// equal to the number of threads; }x\mathrm{ and z point to global arrays
void parprfsum(int *x, int n, int *z)
{
    #pragma omp parallel
    { int i,j,me=omp_get_thread_num(),
                nth = omp_get_num_threads(),
                chunksize = n / nth,
                start = me * chunksize;
        seqprfsum(&x[start],chunksize);
        #pragma omp barrier
        #pragma omp single
        {
        for (i = 0; i < nth-1; i++)
            z[i] = x[(i+1)*chunksize - 1];
        seqprfsum(z,nth-1);
        }
        if (me > 0) {
            for (j = start; j < start + chunksize; j++) {
                x[j] += z[me - 1];
            }
        }
    }
}
```

Here is an example of use: A method for compressing data is to store only repeat counts in runs, where the latter means a set of consecutive, identical values. For instance, the sequence $2,2,2,0,0,5,0,0$ would be compressed to $3,2,2,0,1,5,2,0$, meaning that the data consist of first three 2 s , then two 0 s , then one 5 , and finally two 0 s . Note that the compressed version consists of alternating run counts and run values, respectively 2 and 0 at the end of the above example.

To solve this in OpenMP, we'll first call the above functions to decide where to place the runs in our overall output.

```
void uncomprle(int *x,int nx,int *tmp,int *y,int * ny)
{
    int i, nx2 = nx/2;
    int z[MAXTHREADS];
    for (i}=0; i < nx2; i++) tmp[i+1]=x[2*i]
    parprfsum(tmp+1,nx2+1,z);
    tmp[0] = 0;
    #pragma omp parallel
    { int j,k;
        int me=omp_get_thread_num();
        #pragma omp for
        for (j = 0; j < nx2; j++) {
            // where to start the j-th run?
            int start = tmp[j];
```

```
        // what value is in the run?
        int val = x[2*j+1];
        // how long is the run?
        int nrun = x[2*j];
        for (k = 0; k < nrun; k++)
            y[start+k] = val;
        }
    }
    *ny = tmp[nx2];
}
```


### 10.5 Example: Run-Length Decompression in Thrust

Here's how we could do the first part of the operation above, i.e. determining where to place the runs in our overall output, in Thrust:

```
#include <stdio.h>
#include <thrust/device_vector.h>
#include <thrust/scan.h>
#include <thrust/sequence.h>
#include <thrust/remove.h>
struct iseven {
    bool operator()(const int i)
    { return (i % 2) = 0;
    }
};
int main()
{ int i;
    int x[12] = {2,3,1,9,3,5,2,6,2,88,1,12};
    int nx = 12;
    thrust::device_vector<int> out(nx);
    thrust::device_vector <int> seq(nx);
    thrust::sequence(seq.begin(), seq.end (),0);
    thrust::device_vector <int> dx(x, x+nx);
    thrust:: device_vector <int > :: iterator newend =
            thrust:: copy_if(dx.begin(), dx.end(), seq.begin(), out.begin(), iseven());
    thrust::inclusive_scan(out.begin(),out.end(),out.begin());
    // "out" should be 2, 2+1 = 3,2+1+3=6,\ldots
    thrust::copy(out.begin(), newend,
            std::ostream_iterator<int>(std::cout, " "));
    std::cout << "\n";
}
```


## Chapter 11

## Introduction to Parallel Matrix Operations

## 11.1 "We're Not in Physicsland Anymore, Toto"

In the early days parallel processing was mostly used in physics problems. Typical problems of interest would be grid computations such as the heat equation, matrix multiplication, matrix inversion (or equivalent operations) and so on. These matrices are not those little $3 x 3$ toys you worked with in your linear algebra class. In parallel processing applications of matrix algebra, our matrices can have thousands of rows and columns, or even larger.

The range of applications of parallel processing is of course far broader today, such as image processing, social networks and data mining. Google employs a number of linear algebra experts, and they deal with matrices with literally millions of rows or columns.

We assume for now that the matrices are dense, meaning that most of their entries are nonzero. This is in contrast to sparse matrices, with many zeros. Clearly we would use differents type of algorithms for sparse matrices than for dense ones. We'll cover sparse matrices a bit in Section 11.7

### 11.2 Partitioned Matrices

Parallel processing of course relies on finding a way to partition the work to be done. In the matrix algorithm case, this is often done by dividing a matrix into blocks (often called tiles these days).

For example, let

$$
A=\left(\begin{array}{ccc}
1 & 5 & 12  \tag{11.1}\\
0 & 3 & 6 \\
4 & 8 & 2
\end{array}\right)
$$

and

$$
B=\left(\begin{array}{ccc}
0 & 2 & 5  \tag{11.2}\\
0 & 9 & 10 \\
1 & 1 & 2
\end{array}\right),
$$

so that

$$
C=A B=\left(\begin{array}{ccc}
12 & 59 & 79  \tag{11.3}\\
6 & 33 & 42 \\
2 & 82 & 104
\end{array}\right) .
$$

We could partition A as

$$
A=\left(\begin{array}{ll}
A_{00} & A_{01}  \tag{11.4}\\
A_{10} & A_{11}
\end{array}\right)
$$

where

$$
\begin{gather*}
A_{00}=\left(\begin{array}{ll}
1 & 5 \\
0 & 3
\end{array}\right),  \tag{11.5}\\
A_{01}=\binom{12}{6},  \tag{11.6}\\
A_{10}=\left(\begin{array}{ll}
4 & 8
\end{array}\right) \tag{11.7}
\end{gather*}
$$

and

$$
\begin{equation*}
A_{11}=(2) . \tag{11.8}
\end{equation*}
$$

Similarly we would partition B and C into blocks of a compatible size to A,

$$
B=\left(\begin{array}{ll}
B_{00} & B_{01}  \tag{11.9}\\
B_{10} & B_{11}
\end{array}\right)
$$

and

$$
C=\left(\begin{array}{ll}
C_{00} & C_{01}  \tag{11.10}\\
C_{10} & C_{11}
\end{array}\right),
$$

so that for example

$$
B_{10}=\left(\begin{array}{ll}
1 & 1 \tag{11.11}
\end{array}\right) .
$$

The key point is that multiplication still works if we pretend that those submatrices are numbers! For example, pretending like that would give the relation

$$
\begin{equation*}
C_{00}=A_{00} B_{00}+A_{01} B_{10}, \tag{11.12}
\end{equation*}
$$

which the reader should verify really is correct as matrices, i.e. the computation on the right side really does yield a matrix equal to $C_{00}$.

### 11.3 Parallel Matrix Multiplication

Since so many parallel matrix algorithms rely on matrix multiplication, a core issue is how to parallelize that operation.

Let's suppose for the sake of simplicity that each of the matrices to be multiplied is of dimensions n x n. Let p denote the number of "processes," such as shared-memory threads or message-passing nodes.

### 11.3.1 Message-Passing Case

For concreteness here and in other sections below on message passing, assume we are using MPI.
The obvious plan of attack here is to break the matrices into blocks, and then assign different blocks to different MPI nodes. Assume that $\sqrt{p}$ evenly divides n, and partition each matrix into
submatrices of size $n / \sqrt{p} \times n / \sqrt{p}$. In other words, each matrix will be divided into m rows and m columns of blocks, where $m=n / \sqrt{p}$.
One of the conditions assumed here is that the matrices A and B are stored in a distributed manner across the nodes. This situation could arise for several reasons:

- The application is such that it is natural for each node to possess only part of A and B.
- One node, say node 0 , originally contains all of A and B, but in order to conserve communication time, it sends each node only parts of those matrices.
- The entire matrix would not fit in the available memory at the individual nodes.

As you'll see, the algorithms then have the nodes passing blocks among themselves.

### 11.3.1.1 Fox's Algorithm

Consider the node that has the responsibility of calculating block $(\mathrm{i}, \mathrm{j})$ of the product C , which it calculates as

$$
\begin{equation*}
A_{i 0} B_{0 j}+A_{i 1} B_{1 j}+\ldots+A_{i i} B_{i j}+\ldots+A_{i, m-1} B_{m-1, j} \tag{11.13}
\end{equation*}
$$

Rearrange this with $A_{i i}$ first:

$$
\begin{equation*}
A_{i i} B_{i j}+A_{i, i+1} B_{, i+1 j}+\ldots+A_{i, m-1} B_{m-1, j}+A_{i 0} B_{0 j}+A_{i 1} B_{1 j}+\ldots+A_{i, i-1} B_{i-1, j} \tag{11.14}
\end{equation*}
$$

Written more compactly, this is

$$
\begin{equation*}
\sum_{k=0}^{m-1} A_{i,(i+k) \bmod m} B_{(i+k) \bmod m, j} \tag{11.15}
\end{equation*}
$$

In other words, start with the $A_{i i}$ term, then go across row i of A, wrapping back up to the left end when you reach the right end. The order of summation in this rearrangement will be the actual order of computation. It's similar for $B$, in column j .

The algorithm is then as follows. The node which is handling the computation of $C_{i j}$ does this (in parallel with the other nodes which are working with their own values of $i$ and $j$ ):

```
iup \(=i+1 \bmod m\);
idown \(=\mathrm{i}-1 \bmod \mathrm{~m}\);
for \((\mathrm{k}=0 ; \mathrm{k}<\mathrm{m} ; \mathrm{k}++\) ) \{
    \(\mathrm{km}=(\mathrm{i}+\mathrm{k}) \bmod \mathrm{m}\);
    broadcast \((\mathrm{A}[\mathrm{i}, \mathrm{km}])\) to all nodes handling row i of C ;
    \(\mathrm{C}[\mathrm{i}, \mathrm{j}]=\mathrm{C}[\mathrm{i}, \mathrm{j}]+\mathrm{A}[\mathrm{i}, \mathrm{km}] * \mathrm{~B}[\mathrm{~km}, \mathrm{j}]\)
    send \(B[k m, j]\) to the node handling \(C[i d o w n, j]\)
    receive new \(B[k m+1 \bmod m, j]\) from the node handling \(C[i u p, j]\)
\}
```

The main idea is to have the various computational nodes repeatedly exchange submatrices with each other, timed so that a node receives the submatrix it needs for its computation "just in time."

This is Fox's algorithm. Cannon's algorithm is similar, except that it does cyclical rotation in both rows and columns, compared to Fox's rotation only in columns but broadcast within rows.

The algorithm can be adapted in the obvious way to nonsquare matrices, etc.

### 11.3.1.2 Performance Issues

Note that in MPI we would probably want to implement this algorithm using communicators. For example, this would make broadcasting within a block row more convenient and efficient.

Note too that there is a lot of opportunity here to overlap computation and communication, which is the best way to solve the communication problem. For instance, we can do the broadcast above at the same time as we do the computation.

Obviously this algorithm is best suited to settings in which we have PEs in a mesh topology. This includes hypercubes, though one needs to be a little more careful about communications costs there.

### 11.3.2 Shared-Memory Case

### 11.3.2.1 Example: Matrix Multiply in OpenMP

Since a matrix multiplication in serial form consists of nested loops, a natural way to parallelize the operation in OpenMP is through the for pragma, e.g.

```
#pragma omp parallel for
for (i = 0; i < ncolsa; i++)
    for (j = 0; i < nrowsb; j++) {
        sum = 0;
        for (k=0; i < ncolsa; i++)
            sum +=a[i][k]*b[k][j];
    }
```

This would parallelize the outer loop, and we could do so at deeper nesting levels if profitable.

### 11.3.2.2 Example: Matrix Multiply in CUDA

Given that CUDA tends to work better if we use a large number of threads, a natural choice is for each thread to compute one element of the product, like this:

```
_-global_- void matmul(float *ma,float *mb, float *mc,int nrowsa,
    int ncolsa, int ncolsb, float *total)
{ int k,i,j; float sum;
    // find i,j according to thread and block ID
    sum = 0;
    for (k = 0; k < ncolsa; k++)
        sum += a[i*ncolsa+k]*b[k*ncols+j];
    *total = sum;
}
```

This should produce a good speedup. But we can do even better, much much better.
The CUBLAS package includes very finely-tuned algorithms for matrix multiplication. The CUBLAS source code is not public, though, so in order to get an idea of how such tuning might be done, let's look at Prof. Richard Edgar's algorithm, which makes use of shared memory. (Actually, this may be what CUBLAS uses.)

```
__global__ void MultiplyOptimise(const float *A, const float *B, float *C) {
    // Extract block and thread numbers
    int bx = blockIdx.x; int by = blockIdx.y;
    int tx = threadIdx.x; int ty = threadIdx.y;
    // Index of first A sub-matrix processed by this block
    int aBegin = dc_wA * BLOCK_SIZE * by;
    // Index of last A sub-matrix
    int aEnd = aBegin + dc_wA - 1;
    // Stepsize of A sub-matrices
    int aStep = BLOCK_SIZE;
    // Index of first B sub-matrix
    // processed by this block
    int bBegin = BLOCK_SIZE * bx;
    // Stepsize for B sub-matrices
    int bStep = BLOCK_SIZE * dc_wB;
    // Accumulator for this thread
    float Csub = 0;
    for(int a = aBegin, b=bBegin; a <= aEnd; a += aStep, b+= bStep) {
        // Shared memory for sub-matrices
        __shared__ float As[BLOCK_SIZE][BLOCK_SIZE];
        __shared__ float Bs[BLOCK_SIZE][BLOCK_SIZE];
        // Load matrices from global memory into shared memory
        // Each thread loads one element of each sub-matrix
```

```
        As[ty][tx] = A[a + (dc_wA * ty) + tx];
        Bs[ty][tx] = B[b + (dc_wB * ty) + tx];
        // Synchronise to make sure load is complete
        _-syncthreads();
        // Perform multiplication on sub-matrices
        // Each thread computes one element of the C sub-matrix
        for( int k = 0; k < BLOCK_SIZE; k++ ) {
        Csub += As[ty][k] * Bs[k][tx];
        }
        // Synchronise again
        _-syncthreads();
    }
    // Write the C sub-matrix back to global memory
    // Each thread writes one element
    int c = (dc_wB * BLOCK_SIZE * by) + (BLOCK_SIZE*bx);
    C[c + (dc_wB*ty) + tx] = Csub;
```

\}

Here are the relevant portions of the calling code, including defined constants giving the number of columns ("width") of the multiplier matrix and the number of rows ("height") of the multiplicand:

```
#define BLOCK_SIZE 16
__constant__ int dc_wA;
__constant_- int dc_wB;
..
// Sizes must be multiples of BLOCK_SIZE
dim3 threads(BLOCK_SIZE,BLOCK_SIZE);
dim3 grid(wB/BLOCK_SIZE,hA/BLOCK_SIZE);
MultiplySimple<<<<grid,threads >>>(d_A, d_B, d_C);
...
```

(Note the alternative way to configure threads, using the functions threads() and $\operatorname{grid}()$. .)
Here the the term "block" in the defined value BLOCK_SIZE refers both to blocks of threads and the partitioning of matrices. In other words, a thread block consists of 256 threads, to be thought of as a $16 \times 16$ "array" of threads, and each matrix is partitioned into submatrices of size $16 \times 16$.

In addition, in terms of grid configuration, there is again a one-to-one correspondence between thread blocks and submatrices. Each submatrix of the product matrix C will correspond to, and will be computed by, one block in the grid.

We are computing the matrix product $\mathrm{C}=\mathrm{AB}$. Denote the elements of A by $a_{i j}$ for the element in row i , column j , and do the same for B and C . Row-major storage is used.

Each thread will compute one element of C, i.e. one $c_{i j}$. It will do so in the usual way, by multiplying column j of B by row i of A . However, the key issue is how this is done in concert with the other threads, and the timing of what portions of A and B are in shared memory at various times.

Concerning the latter, note the code

```
for (int \(\mathrm{a}=\mathrm{aBegin}, \mathrm{b}=\mathrm{bBegin} ; \mathrm{a}<=\mathrm{aEnd} ; \mathrm{a}+=\mathrm{aStep}, \mathrm{b}+=\mathrm{bStep})\) \{
    // Shared memory for sub-matrices
    _-shared_- float As[BLOCK_SIZE][BLOCK_SIZE];
    _-shared_- float Bs[BLOCK_SIZE][BLOCK_SIZE];
    // Load matrices from global memory into shared memory
    // Each thread loads one element of each sub-matrix
    \(\operatorname{As}[\mathrm{ty}][\mathrm{tx}]=\mathrm{A}\left[\mathrm{a}+\left(\mathrm{dc} \_\mathrm{wA} * \mathrm{ty}\right)+\mathrm{tx}\right] ;\)
    \(\operatorname{Bs}[\mathrm{ty}][\mathrm{tx}]=\mathrm{B}[\mathrm{b}+(\mathrm{dc}-\mathrm{wB} * \mathrm{ty})+\mathrm{tx}] ;\)
```

Here we loop across a row of submatrices of A, and a column of submatrices of B, calculating one submatrix of C. In each iteration of the loop, we bring into shared memory a new submatrix of A and a new one of B. Note how even this copying from device global memory to device shared memory is shared among the threads.

As an example, suppose

$$
A=\left(\begin{array}{cccccc}
1 & 2 & 3 & 4 & 5 & 6  \tag{11.16}\\
7 & 8 & 9 & 10 & 11 & 12
\end{array}\right)
$$

and

$$
B=\left(\begin{array}{cccc}
1 & 2 & 3 & 4  \tag{11.17}\\
5 & 6 & 7 & 8 \\
9 & 10 & 11 & 12 \\
13 & 14 & 15 & 16 \\
17 & 18 & 19 & 20 \\
21 & 22 & 23 & 24
\end{array}\right)
$$

Further suppose that BLOCK_SIZE is 2. That's too small for good efficiency-giving only four threads per block rather than 256-but it's good for the purposes of illustration.

Let's see what happens when we compute $C_{00}$, the 2 x 2 submatrix of C's upper-left corner. Due to the fact that partitioned matrices multiply "just like numbers," we have

$$
\begin{align*}
C_{00} & =A_{00} B_{00}+A_{01} B_{10}+A_{02} B_{20}  \tag{11.18}\\
& =\left(\begin{array}{ll}
1 & 2 \\
7 & 8
\end{array}\right)\left(\begin{array}{ll}
1 & 2 \\
5 & 6
\end{array}\right)+\ldots \tag{11.19}
\end{align*}
$$

Now, all this will be handled by thread block number ( 0,0 ), i.e. the block whose X and Y "coordinates" are both 0 . In the first iteration of the loop, $A_{11}$ and $B_{11}$ are copied to shared memory for that block, then in the next iteration, $A_{12}$ and $B_{21}$ are brought in, and so on.

Consider what is happening with thread number $(1,0)$ within that block. Remember, its ultimate goal is to compute $c_{21}$ (adjusting for the fact that in math, matrix subscripts start at 1 ). In the first iteration, this thread is computing

$$
\left(\begin{array}{ll}
1 & 2 \tag{11.20}
\end{array}\right)\binom{1}{5}=11
$$

It saves that 11 in its running total Csub, eventually writing it to the corresponding element of C :

```
int c = (dc_wB * BLOCK_SIZE * by) + (BLOCK_SIZE*bx);
C[c + (dc_wB*ty) + tx ] = Csub;
```

Professor Edgar found that use of shared device memory resulted a huge improvement, extending the original speedup of 20 X to 500 X !

### 11.4 Finding Powers of Matrices

In some applications, we are interested not just in multiplying two matrices, but rather in multiplying a matrix by itself, many times.

### 11.4.1 Example: Graph Connectedness

Let n denote the number of vertices in the graph. Define the graph's adjacency matrix A to be the nxn matrix whose element ( $\mathrm{i}, \mathrm{j}$ ) is equal to 1 if there is an edge connecting vertices i an j (i.e. $i$ and $j$ are "adjacent"), and 0 otherwise. The corresponding reachability matrix $R$ has its ( $\mathrm{i}, \mathrm{j}$ ) element equal to 1 if there is some path from i to j , and 0 otherwise.
One can prove that

$$
\begin{equation*}
R=b\left[(I+A)^{n-1}\right], \tag{11.21}
\end{equation*}
$$

where I is the identity matrix and the function $b()$ (' $b$ ' for "boolean") is applied elementwise to its matrix argument, replacing each nonzero element by 1 while leaving the elements which are 0 unchanged. The graph is connected if and only if all elements of $R$ are 1 s .

So, the original graph connectivity problem reduces to a matrix problem.

### 11.4.2 Example: Fibonacci Numbers

The basic problem is well known: Find the Fibonacci numbers $f_{n}$, where

$$
\begin{equation*}
f_{0}=f_{1}=1 \tag{11.22}
\end{equation*}
$$

and

$$
\begin{equation*}
f_{n}=f_{n-1}+f_{n-2}, n>1 \tag{11.23}
\end{equation*}
$$

The point is that (11.23) can be couched in matrix terms as

$$
\begin{equation*}
\binom{f_{n+1}}{f_{n}}=A\binom{f_{n}}{f_{n-1}} \tag{11.24}
\end{equation*}
$$

where

$$
A=\left(\begin{array}{ll}
1 & 1  \tag{11.25}\\
1 & 0
\end{array}\right)
$$

Given the initial conditions 11.22 and 11.24 , we have

$$
\begin{equation*}
\binom{f_{n+1}}{f_{n}}=A^{n-1}\binom{1}{1} \tag{11.26}
\end{equation*}
$$

In other words, our problem reduces to one of finding the powers $A, A^{2}, \ldots, A^{n-1}$.

### 11.4.3 Example: Matrix Inversion

Many applications make use of $A^{-1}$ for an $\mathrm{n} \times \mathrm{n}$ square matrix A. In many cases, it is not computed directly, but here we address methods for direct computation.

We could use the methods of Section 11.5 to find matrix inverses, but there is alos a power series method.

Recall that for numbers x that are smaller than 1 in absolute value,

$$
\begin{equation*}
\frac{1}{1-x}=1+x+x^{2}+\ldots \tag{11.27}
\end{equation*}
$$

In algebraic terms, this would be that for an n x n matrix C ,

$$
\begin{equation*}
(I-C)^{-1}=I+C+C^{2}+\ldots \tag{11.28}
\end{equation*}
$$

This can be shown to converge if

$$
\begin{equation*}
\max _{i, j}\left|c_{i j}\right|<1 \tag{11.29}
\end{equation*}
$$

To invert our matrix A, then, we can set $\mathrm{C}=\mathrm{I}-\mathrm{A}$, giving us

$$
\begin{equation*}
A^{-1}=(I-C)^{-1}=I+C+C^{2}+\ldots=I+(I-A)+(I-A)^{2}+\ldots \tag{11.30}
\end{equation*}
$$

To meet the convergence condition, we could set $\tilde{A}=d A$, where d is small enough so that 11.29) holds for $I-\tilde{A}$. This will be possible, if all the elements of A are nonnegative. We then find the inverse of dA , and in the end multiply by d to get the inverse of A .

### 11.4.4 Parallel Computation

Finding a power such as $A^{2}$ could be viewed as a special case of the matrix multiplication AB with $\mathrm{A}=\mathrm{B}$. There are some small improvements that we could make in our algorithm in the previous section for this case, but also there are other approaches that could yield much better dividends.

Suppose for instance we need to find $A^{32}$, as in the graph theory example above. We could apply the above algorithm 31 times. But a much faster approach would be to first calculate $A^{2}$, then square that result to get $A^{4}$, then square it to get $A^{8}$ and so on. That would get us $A^{32}$ by applying a matrix multiplication algorithm only five times, instead of 31 .

### 11.5 Solving Systems of Linear Equations

Suppose we have a system of equations

$$
\begin{equation*}
a_{i 0} x_{0}+\ldots+a_{i, n-1} x_{n-1}=b_{i}, i=0,1, \ldots, n-1, \tag{11.31}
\end{equation*}
$$

where the $x_{i}$ are the unknowns to be solved for.

As you know, this system can be represented compactly as

$$
\begin{equation*}
A x=b, \tag{11.32}
\end{equation*}
$$

where A is nx n and x and b is nx 1 .

### 11.5.1 Gaussian Elimination

Form the $\mathrm{nx}(\mathrm{n}+1)$ matrix $\mathrm{C}=(\mathrm{A} \mid \mathrm{b})$ by appending the column vector b to the right of A . (It may be advantageous to add padding on the right of b .)

Then we work on the rows of C , with the pseudocode for the sequential case in the most basic version being

```
for ii = 0 to n-1
    divide row ii by c[i][i]
    for r = 0 to n-1, r != i
        replace row r by row r - c[r][ii] times row ii
```

In the divide operation in the above pseudocode, $c_{i i}$ might be 0 , or close to 0 . In that case, a pivoting operation is performed (not shown in the pseudocode): that row is first swapped with another one further down.

This transforms C to reduced row echelon form, in which A is now the identity matrix I and b is now our solution vector x .

A variation is to transform only to row echelon form. This means that $C$ ends up in upper triangular form, with all the elements $c_{i j}$ with $i>j$ being 0 , and with all diagonal elements being equal to 1 . Here is the pseudocode:

```
for ii = 0 to n-1
    divide row ii by c[i][i]
    for r = ii+1 to n-1 // vacuous if r = n-1
        replace row r by row r - c[r][ii] times row ii
```

This corresponds to a new set of equations,

$$
\begin{aligned}
c_{00} x_{0}+c_{11} x_{1}+c_{22} x_{2}+\ldots+c_{0, n-1} x_{n-1} & =b_{0} \\
c_{11} x_{1}+c_{22} x_{2}+\ldots+c_{1, n-1} x_{n-1} & =b_{1} \\
c_{22} x_{2}+\ldots+c_{2, n-1} x_{n-1} & =b_{2} \\
\ldots & \\
c_{n-1, n-1} x_{n-1} & =b_{n-1}
\end{aligned}
$$

We then find the $x_{i}$ via back substitution:

```
\(x[n-1]=b[n-1] / c[n-1, n-1]\)
for \(\mathrm{i}=\mathrm{n}-2\) downto 0
    \(x[i]=(b[i]-c[i][n-1] * x[n-1]-\ldots-c[i][i+1] * x[i+1]) / c[i][i]\)
```


### 11.5.2 Example: Gaussian Elimination in CUDA

Here's CUDA code for the reduced row echelon form version, suitable for a not-extremely-large matrix:

```
// linear index for matrix element at row i, column j, in an m-column
// matrix
_-device_- int onedim(int i, int j,int m) {return i*m+j;}
// replace u by c* u; vector of length m
__device__ void cvec(float *u, int m, float c)
{ for (int i = 0; i<m; i++) u[i] = c* u[i]; }
// multiply the vector u of length m by the constant c (not changing u)
// and add the result to v
-_device_- void vplscu(float *u, float *v, int m, float c)
{ for (int i = 0; i < m; i++) v[i] += c*u[i]; }
// copy the vector u of length m to v
_-device_- void cpuv(float *u, float *v, int m)
{ for (int i = 0; i < m; i++) v[i] = u[i]; }
// solve matrix equation }\textrm{Ax}=\textrm{b}; straight Gaussian elimination, n
// pivoting etc.; the matrix ab is (A|b), n rows; ab is destroyed, with
// x placed in the last column; one block, with thread i handling row i
__global__ void gauss(float *ab, int n)
{ int i,n1=n+1,abii,abme;
    extern --shared_- float iirow[];
    int me = threadIdx.x;
    for (i = 0; i < n; i++) {
        if (i = me) {
            abii = onedim(i,i,n1);
            cvec(&ab[abii],n1-i,1/ab[abii]);
            cpuv(&ab[abii], iirow,n1-i);
            }
            -_syncthreads();
            if (i != me) {
                abme = onedim(me,i,n1);
                vplscu(iirow,&ab[abme],n1-i,-ab[abme]);
            }
            --syncthreads();
        }
}
```

Here we have one thread for each row, and are using just one block, so as to avoid interblock synchronization problems and to easily use shared memory. Concerning the latter, note that since the pivot row, iirow, is read many times, it makes sense to put it in shared memory.

Needless to say, the restriction to one block is quite significant. With a 512 -thread limit per block, this would limit us to $512 \times 512$ matrices. But it's even worse than that-if shared memory is only 4 K in size, in single precision that would mean something like $30 \times 30$ matrices! We could go to multiple blocks, at the cost of incurring synchronization delays coming from repeated kernel calls.

In a row echelon version of the code, we could have dynamic assignment of rows to threads, but still would eventually have load balancing issues.

### 11.5.3 The Jacobi Algorithm

One can rewrite (11.31) as

$$
\begin{equation*}
x_{i}=\frac{1}{a_{i i}}\left[b_{i}-\left(a_{i 0} x_{0}+\ldots+a_{i, i-1} x_{i-1}+a_{i, i+1} x_{i+1}+\ldots+a_{i, n-1} x_{n-1}\right)\right], i=0,1, \ldots, n-1 \tag{11.33}
\end{equation*}
$$

This suggests a natural iterative algorithm for solving the equations. We start with our guess being, say, $x_{i}=b_{i}$ for all i. At our $\mathrm{k}^{\text {th }}$ iteration, we find our $(\mathrm{k}+1)^{\text {st }}$ guess by plugging in our $\mathrm{k}^{\text {th }}$ guess into the right-hand side of 11.33). We keep iterating until the difference between successive guesses is small enough to indicate convergence.

This algorithm is guaranteed to converge if each diagonal element of A is larger in absolute value than the sum of the absolute values of the other elements in its row.

Parallelization of this algorithm is easy: Just assign each process to handle a section of $x=$ $\left(x_{0}, x_{1}, \ldots, x_{n-1}\right)$. Note that this means that each process must make sure that all other processes get the new value of its section after every iteration.

Note too that in matrix terms 11.33) can be expressed as

$$
\begin{equation*}
x^{(k+1)}=D^{-1}\left(b-O x^{(k)}\right) \tag{11.34}
\end{equation*}
$$

where D is the diagonal matrix consisting of the diagonal elements of A (so its inverse is just the diagonal matrix consisting of the reciprocals of those elements), O is the square matrix obtained by replacing A's diagonal elements by 0 s , and $x^{(i)}$ is our guess for x in the $\mathrm{i}^{\text {th }}$ iteration. This reduces the problem to one of matrix multiplication, and thus we can parallelize the Jacobi algorithm by utilizing a method for doing parallel matrix multiplication.

### 11.5.4 Example: OpenMP Implementation of the Jacobi Algorithm

OpenMP code for Jacobi is straightforward:

```
#include <omp.h>
// partitions s..e into nc chunks, placing the ith in first and last (i
// = 0,\ldots,nc-1)
void chunker(int s, int e, int nc, int i, int *first, int *last)
{ int chunksize = (e-s+1) / nc;
    *first = s + i * chunksize;
    if (i < nc-1) *last = *first + chunksize - 1;
    else *last = e;
}
// returns the "dot product" of vectors u and v
float innerprod(float *u, float *v, int n)
{ float sum = 0.0; int i;
    for (i = 0; i < n; i++)
            sum += u[i] * v[i];
    return sum;
}
// solves AX = Y, A nxn; stops iteration when total change is < n*eps
void jacobi(float *a, float *x, float *y, int n, float eps)
{
    float *oldx = malloc(n*sizeof(float));
    float se;
    #pragma omp parallel
    { int i;
            int thn = omp_get_thread_num();
            int nth = omp_get_num_threads();
            int first, last;
            chunker(0,n-1,nth,thn,&first,&last);
            for (i = first; i <= last; i++) oldx[i] = x[i] = 1.0;
            float tmp;
            while (1) {
            for (i = first; i <= last; i++) {
                    tmp = innerprod(&a[n*i],oldx,n);
                    tmp -= a[n*i+i] * oldx[i];
                    x[i] = (y[i] - tmp) / a[n*i+i];
            }
            #pragma omp barrier
            #pragma omp for reduction(+:se)
            for (i = first; i <= last; i++)
                    se += abs(x[i]-oldx[i]);
            #pragma omp barrier
            if (se < n*eps) break;
            for (i = first; i <= last; i++)
                    oldx[i] = x[i];
```

```
        }
    }
}
```

Note the use of the OpenMP reduction clause.

### 11.5.5 Example: R/gputools Implementation of Jacobi

Here's the R code, using gputools:

```
library(gputools)
jcb <- function(a,b,eps) {
    n <- length(b)
    d <- diag(a) # a vector, not a matrix
    tmp <- diag(d) # a matrix, not a vector
    o <- a - diag(d)
    di <- 1/d
    x <- b # initial guess, could be better
    repeat {
        oldx <- x
        tmp <- gpuMatMult(o,x)
        tmp <- b - tmp
        x <- di * tmp # elementwise multiplication
        if (sum(abs(x-oldx))< n * eps) return(x)
    }
}
```


### 11.6 Eigenvalues and Eigenvectors

With the popularity of document search (Web search, text mining etc.), eigenanalysis has become much more broadly used. Given the size of the problems, again parallel computation is needed. This can become quite involved, with many complicated methods having been developed.

### 11.6.1 The Power Method

One of the simplest methods is the power method. Consider an nxn matrix A, with eigenvalues $\lambda_{1}, \ldots, \lambda_{n}$, where the labeling is such that $\left|\lambda_{1}\right| \geq\left|\lambda_{2}\right| \geq \ldots \geq\left|\lambda_{n}\right|$. We'll assume here that $A$ is a symmetric matrix, which it is for instance in statistical applications (Section 14.4). That implies that the eigenvalues of A are real, and that the eigenvectors are orthogonal to each other.

Start with some nonzero vector x , and define the $\mathrm{k}^{\text {th }}$ iterate by

$$
\begin{equation*}
x^{(k)}=\frac{A^{k} x}{\left\|A^{k} x\right\|} \tag{11.35}
\end{equation*}
$$

Under mild conditions, $x^{(k)}$ converges to an eigenvector $v_{1}$ corresponding to $\lambda_{1}$. Moreover, the quantities $\left(A x^{(k)}\right)^{\prime} x^{(k)}$ converge to $\lambda_{1}$.

This method is reportedly used in Google's PageRank algorithm, which is only concerned with the largest eigenvalue/eigenvector. But what if you want more?

Consider now the matrix

$$
\begin{equation*}
B=A-\lambda_{1} v_{1} v_{1}^{\prime} \tag{11.36}
\end{equation*}
$$

where we've now scaled $v_{1}$ to have length 1 .
Then

$$
\begin{align*}
B v_{1} & =A v_{1}-\lambda_{1} v 1\left(v_{1}^{\prime} v_{1}\right)  \tag{11.37}\\
& =\lambda_{1} v_{1}-\lambda_{1} v 1(1)  \tag{11.38}\\
& =0 \tag{11.39}
\end{align*}
$$

and for $i>0$,

$$
\begin{align*}
B v_{i} & =A v_{i}-\lambda_{1} v 1\left(v_{1}^{\prime} v_{i}\right)  \tag{11.40}\\
& =\lambda_{i} v_{i}-\lambda_{1} v 1(0)  \tag{11.41}\\
& =\lambda_{i} v_{i} \tag{11.42}
\end{align*}
$$

In other words, the eigenvalues of B are $\lambda_{2}, \ldots, \lambda_{n}, 0$. So we can now apply the same procedure to B to get $\lambda_{2}$ and $v_{2}$, and iterate for the rest.

### 11.6.2 Parallel Computation

To use the power method in parallel, note that this is again a situation in which we wish to compute powers of matrices. However, there is also scaling involved, as seen in (11.35). We may wish to try the "log method" of Section 11.4, with scaling done occasionally.

The CULA library for CUDA, mentioned earlier, includes routines for finding the singular value decomposition of a matrix, thus providing the eigenvectors 1 The R package gputools has an interface to the SVD routine in CULA.

### 11.7 Sparse Matrices

As mentioned earlier, in many parallel processing applications of linear algebra, the matrices can be huge, even having millions of rows or columns. However, in many such cases, most of the matrix consists of 0 s . In an effort to save memory, one can store such matrices in compressed form, storing only the nonzero elements.

Sparse matrices roughly fall into two categories. In the first category, the matrices all have 0s at the same known positions. For instance, in tridiagonal matrices, the only nonzero elements are either on the diagonal or on subdiagonals just below or above the diagonal, and all other elements are guaranteed to be 0 , such as

$$
\left(\begin{array}{lllll}
2 & 0 & 0 & 0 & 0  \tag{11.43}\\
1 & 1 & 8 & 0 & 0 \\
0 & 1 & 5 & 8 & 0 \\
0 & 0 & 0 & 8 & 8 \\
0 & 0 & 0 & 3 & 5
\end{array}\right)
$$

Code to deal with such matrices can then access the nonzero elements based on this knowledge.
In the second category, each matrix that our code handles will typically have its nonzero matrices in different, "random," positions. A number of methods have been developed for storing amorphous sparse matrices, such as the Compressed Sparse Row format, which we'll code in this C struct, representing an mxn matrix A , with k nonzero entries:

```
struct {
    int m,n; // numbers of rows and columns of A
    float *avals; // the nonzero values of A, in row-major order; length k
    int *cols; // avals[i] is in column cols[i] in A; length k
    int *rowplaces; // rowplaces[i] is the index in avals for the 1st
            // nonzero element of row i in A (but last element
            // is k); length m+1
}
```

For the matrix in (if we were not to exploit its tridiagonal nature, and just treat it as amorphous):

[^44]- m,n: 5,5
- avals: $2,1,1,8,1,5,8,8,8,3,5$
- cols: $0,0,1,2,1,2,3,3,4,3,4$
- rowplaces: $0,2,4,6,9,11$

The array rowplaces tells us which parts of avals go into which rows of A:

| row in A | indices in rowplaces |
| ---: | ---: |
| 0 | $0-1$ |
| 1 | $2-3$ |
| 2 | $4-5$ |
| 3 | $6-8$ |
| 4 | $9-10$ |

### 11.8 Libraries

Of course, remember that CUDA provides some excellent matrix-operation routines, in CUBLAS. There is also the CUSP library for sparse matrices (i.e. those with a lot of 0 s ). Note too the CULA library (not developed by NVIDIA, but using CUDA).

More general (i.e. non-CUDA) parallel libraries for linear algebra include ScalaPACK and PLAPACK.

## Chapter 12

## Introduction to Parallel Sorting

Sorting is one of the most common operations in parallel processing applications. For example, it is central to many parallel database operations, and important in areas such as image processing, statistical methodology and so on. A number of different types of parallel sorting schemes have been developed. Here we look at some of these schemes.

### 12.1 Quicksort

You are probably familiar with the idea of quicksort: First break the original array into a "smallelement" pile and a "large-element" pile, by comparing to a pivot element. In a naive implementation, the first element of the array serves as the pivot, but better performance can be obtained by taking, say, the median of the first three elements. Then "recurse" on each of the two piles, and then string the results back together again.

This is an example of the divide and conquer approach seen in so many serial algorithms. It is easily parallelized (though load-balancing issues may arise). Here, for instance, we might assign one pile to one thread and the other pile to another thread.

Suppose the array to be sorted is named $\mathbf{x}$, and consists of $\mathbf{n}$ elements.

### 12.1.1 The Separation Process

A major issue is how we separate the data into piles.
In a naive implementation, the piles would be put into new arrays, but this is bad in two senses: It wastes memory space, and wastes time, since much copying of arrays needs to be done. A better
implementation places the two piles back into the original array $\mathbf{x}$. The following C code does that.
The function separate() is intended to be used in a recursive quicksort operation. It operates on $\mathbf{x}[1]$ through $\mathbf{x}[\mathbf{h}]$, a subarray of $\mathbf{x}$ that itself may have been formed at an earlier stage of the recursion. It forms two piles from those elements, and placing the piles back in the same region $\mathbf{x}[\mathbf{l}]$ through $\mathbf{x}[\mathbf{h}]$. It also has a return value, showing where the first pile ends.

```
int separate(int l, int h)
{ int ref,i,j,k,tmp;
    ref = x [h]; i = 1-1; j = h;
    do {
        do i++; while (x[i] < ref && i < h);
        do j--; while (x[j] > ref && j > l);
        tmp = x[i]; x[i] = x[j]; x[j] = tmp;
    } while (j > i);
    x[j] = x[i]; x[i] = x[h]; x[h] = tmp;
    return i;
}
```

The function separate() rearranges the subarray, returning a value $\mathbf{m}$, so that:

- $\mathbf{x}[\mathbf{1}]$ through $\mathbf{x}[\mathbf{m - 1}]$ are less than $\mathbf{x}[\mathbf{m}]$,
- $\mathbf{x}[\mathbf{m}+\mathbf{1}]$ through $\mathbf{x}[\mathbf{h}]$ are greater than $\mathbf{x}[\mathbf{m}]$, and
- $\mathbf{x}[\mathbf{m}]$ is in its "final resting place," meaning that $\mathbf{x}[\mathbf{m}]$ will never move again for the remainder of the sorting process. (Another way of saying this is that the current $\mathbf{x}[\mathbf{m}]$ is the $\mathbf{m}$-th smallest of all the original $\mathbf{x}[\mathbf{i}], \mathbf{i}=0,1, \ldots, \mathbf{n}-1$.)

By the way, $\mathbf{x}[\mathbf{l}]$ through $\mathbf{x}[\mathbf{m}-\mathbf{1}]$ will also be in their final resting places as a group. They may be exchanging places with each other from now on, but they will never again leave the range $\mathbf{i}$ though $\mathbf{m - 1}$ within the $\mathbf{x}$ array as a whole. A similar statement holds for $\mathrm{x}[\mathbf{m}+\mathbf{1}]$ through $\mathrm{x}[\mathbf{n} \mathbf{- 1}]$.

Another approach is to do a prefix scan. As an illustration, consider the array

$$
\begin{array}{lllllllll}
28 & 35 & 12 & 5 & 13 & 6 & 8 & 10 & 168 \tag{12.1}
\end{array}
$$

We'll take the first element, 28 , as the pivot, and form a new array of 1 s and 0 s, where 1 means "less than the pivot":

$$
\begin{array}{rrrrrrrrr}
28 & 35 & 12 & 5 & 13 & 6 & 48 & 10 & 168 \\
1 & 0 & 1 & 1 & 1 & 1 & 0 & 1 & 0
\end{array}
$$

Now form the prefix scan (Chapter 10) of that second array, with respect to addition. It will be an exclusive scan (Section 10.3). This gives us

| 28 | 35 | 12 | 5 | 13 | 6 | 48 | 10 | 168 |
| ---: | ---: | ---: | ---: | ---: | ---: | ---: | ---: | ---: |
| 0 | 0 | 1 | 1 | 1 | 1 | 0 | 1 | 0 |
| 0 | 0 | 0 | 1 | 2 | 3 | 3 | 4 | 4 |

Now, the key point is that for every element 1 in that second row, the corresponding element in the third row shows where the first-row element should be placed under the separation operation! Here's why:

The elements $12,5,13,6$ and 10 should go in the first pile, which in an in-place separation would means indices $0,1,2,3$, and 4 . Well, as you can see above, these are precisely the values shown in the third row for $12,5,13,6$ and 10 , all of which have 1 s in the second row.

The pivot, 28 , then should immediately follow that low pile, i.e. it should be placed at index 5 .
We can simply place the high pile at the remaining indicies, 6 through 8 (though we'll do it more systematically below).

In general for an array of length k , we:

- form the second row of 1 s and 0 s indicating $<$ pivot
- form the third row, the exclusive prefix scan
- for each 1 in the second row, place the corresponding element in row 1 into the spot indicated by row 3
- place the pivot in the place indicated by 1 plus m , the largest value in row 3
- form row 4 , equal to ( $0,1, \ldots, \mathrm{k}-1$ ) minus row 3 plus m
- for each 0 in the second row, place the corresponding element in row 1 into the spot indicated by row 4

Note that this operation, using scan, could be used an an alternative to the separate() function above. But it could be done in parallel; more on this below.

### 12.1.2 Example: OpenMP Quicksort

Here is OpenMP code which performs quicksort in the shared-memory paradigm (adapted from code in the OpenMP Source Code Repository, http://www.pcg.ull.es/ompscr/):

```
void qs(int *x, int l, int h)
{ int newl[2], newh[2], i, m;
    m = separate(x,l,h);
    newl[0] = l; newh[0] = m-1;
```

```
    newl[1] = m+1; newh[1] = h;
    #pragma omp parallel
    {
        #pragma omp for nowait
        for (i = 0; i < 2; i++)
        qs(newl[i],newh[i]);
    }
```

\}

Note the nowait clause. Since different threads are operating on different portions of the array, they need not be synchronized.

Recall that another implementation, using the task directive, was given earlier in Section 4.5.
In both of these implementations, we used the function separate() defined above. So, different threads apply different separation operations to different subarrays. An alternative would be to place the parallelism in the separation operation itself, using the parallel algorithms for prefix scan in Chapter 10.

### 12.1.3 Hyperquicksort

This algorithm was originally developed for hypercubes, but can be used on any message-passing system having a power of 2 for the number of nodes ${ }^{1}$

It is assumed that at the beginning each PE contains some chunk of the array to be sorted. After sorting, each PE will contain some chunk of the sorted array, meaning that:

- each chunk is itself in sorted form
- for all cases of $i<j$, the elements at PE i are less than the elements at PE j

If the sorted array itself were our end, rather than our means to something else, we could now collect it at some node, say node 0 . If, as is more likely, the sorting is merely an intermediate step in a larger distributed computation, we may just leave the chunks at the nodes and go to the next phase of work.

Say we are on a d-cube. The intuition behind the algorithm is quite simple:

```
for i = d downto 1
    for each i-cube:
        root of the i-cube broadcasts its median to all in the i-cube,
        to serve as pivot
        consider the two (i-1)-subcubes of this i-cube
```

[^45]```
each pair of partners in the (i-1)-subcubes exchanges data:
    low-numbered PE gives its partner its data larger than pivot
    high-numbered PE gives its partner its data smaller than pivot
```

To avoid deadlock, have the lower-numbered partner send then receive, and vice versa for the higher-numbered one. Better, in MPI, use MPI_SendRcv().

After the first iteration, all elements in the lower (d-1)-cube are less than all elements in higher (d-1)-cube. After d such steps, the array will be sorted.

### 12.2 Mergesorts

### 12.2.1 Sequential Form

In its serial form, mergesort has the following pseudocode:

```
// initially called with l = 0 and h = n-1, where n is the length of the
// array and is assumed here to be a power of 2
void seqmergesort(int *x, int l, int h)
{ seqmergesort(x,0,h/2-1);
    seqmergesort (x,h/2,h);
    merge(x,l,h);
}
```

The function merge() should be done in-place, i.e. without using an auxiliary array. It basically codes the operation shown in pseudocode for the message-passing case in Section 12.2.3.

### 12.2.2 Shared-Memory Mergesort

This is similar to the patterns for shared-memory quicksort in Section 12.1.2 above.

### 12.2.3 Message Passing Mergesort on a Tree Topology

First, we organize the processing nodes into a binary tree. This is simply from the point of view of the software, rather than a physical grouping of the nodes. We will assume, though, that the number of nodes is one less than a power of 2 .

To illustrate the plan, say we have seven nodes in all. We could label node 0 as the root of the tree, label nodes 1 and 2 to be its two children, label nodes 3 and 4 to be node 1's children, and finally label nodes 5 and 6 to be node 2's children.

It is assumed that the array to be sorted is initially distributed in the leaf nodes (recall a similar situation for hyperquicksort), i.e. nodes $3-6$ in the above example. The algorithm works best if there are approximately the same number of array elements in the various leaves.

In the first stage of the algorithm, each leaf node applies a regular sequential sort to its current holdings. Then each node begins sending its now-sorted array elements to its parent, one at a time, in ascending numerical order.

Each nonleaf node then will merge the lists handed to it by its two children. Eventually the root node will have the entire sorted array. Specifically, each nonleaf node does the following:
do
if my left-child datum < my right-child datum pass my left-child datum to my parent
else
pass my right-child datum to my parent
until receive the "no more data" signal from both children

There is quite a load balancing issue here. On the one hand, due to network latency and the like, one may get better performance if each node accumulates a chunk of data before sending to the parent, rather than sending just one datum at a time. Otherwise, "upstream" nodes will frequently have no work to do.

On the other hand, the larger the chunk size, the earlier the leaf nodes will have no work to do. So for any particular platform, there will be some optimal chunk size, which would need to be determined by experimentation.

### 12.2.4 Compare-Exchange Operations

These are key to many sorting algorithms.
A compare-exchange, also known as compare-split, simply means in English, "Let's pool our data, and then I'll take the lower half and you take the upper half." Each node executes the following pseudocode:

```
send all my data to partner
receive all my partner's data
if I have a lower id than my partner
    I keep the lower half of the pooled data
else
    I keep the upper half of the pooled data
```


### 12.2.5 Bitonic Mergesort

Definition: A sequence ( $a_{0}, a_{1}, . ., a_{k-1}$ ) is called bitonic if either of the following conditions holds:
(a) The sequence is first nondecreasing then nonincreasing, meaning that for some $r$

$$
\left(a_{0} \leq a_{1} \leq \ldots \leq a_{r} \geq a_{r+1} \geq a_{n-1}\right)
$$

(b) The sequence can be converted to the form in (a) by rotation, i.e. by moving the last k elements from the right end to the left end, for some k .

As an example of (b), the sequence ( $3,8,12,15,14,5,1,2$ ) can be rotated rightward by two element positions to form ( $1,2,3,8,12,15,14,5$ ). Or we could just rotate by one element, moving the 2 to forming ( $2,3,8,12,15,14,5,1$ ).

Note that the definition includes the cases in which the sequence is purely nondecreasing ( $\mathrm{r}=\mathrm{n}-1$ ) or purely nonincreasing $(r=0)$.

Also included are "V-shape" sequences, in which the numbers first decrease then increase, such as $(12,5,2,8,20)$. By (b), these can be rotated to form (a), with ( $12,5,2,8,20$ ) being rotated to form ( $2,8,20,12,5$ ), an "A-shape" sequence.
(For convenience, from here on I will use the terms increasing and decreasing instead of nonincreasing and nondecreasing.)

Suppose we have bitonic sequence ( $a_{0}, a_{1}, . ., a_{k-1}$ ), where k is a power of 2 . Rearrange the sequence by doing compare-exchange operations between $a_{i}$ and $\left.a_{n / 2+i}\right), \mathrm{i}=0,1, \ldots, \mathrm{n} / 2-1$. Then it is not hard to prove that the new $\left(a_{0}, a_{1}, . ., a_{k / 2-1}\right)$ and ( $\left.a_{k / 2}, a_{k / 2+1}, . ., a_{k-1}\right)$ are bitonic, and every element of that first subarray is less than or equal to every element in the second one.

So, we have set things up for yet another divide-and-conquer attack:

```
// x is bitonic of length n, n a power of 2
void sortbitonic(int *x, int n)
{ do the pairwise compare-exchange operations
    if (n > 2) {
        sortbitonic(x,n/2);
        sortbitonic(x+n/2,n/2);
    }
}
```

This can be parallelized in the same ways we saw for Quicksort earlier.
So much for sorting bitonic sequences. But what about general sequences?
We can proceed as follows, using our function sortbitonic() above:

1. For each $\mathrm{i}=0,2,4, \ldots, \mathrm{n}-2$ :

- Each of the pairs $\left(a_{i}, a_{i+1}\right), \mathrm{i}=0,2, \ldots, \mathrm{n}-2$ is bitonic, since any 2 -element array is bitonic!
- Apply sortbitonic () to $\left(a_{i}, a_{i+1}\right)$. In this case, we are simply doing a compare-exchange.
- If $\mathrm{i} / 2$ is odd, reverse the pair, so that this pair and the pair immediately preceding it now form a 4 -element bitonic sequence.

2. For each $i=0,4,8, \ldots, n-4$ :

- Apply sortbitonic() to ( $\left.a_{i}, a_{i+1}, a_{i+2}, a_{i+3}\right)$.
- If $\mathrm{i} / 4$ is odd, reverse the quartet, so that this quartet and the quartet immediately preceding it now form an 8 -element bitonic sequence.

3. Keep building in this manner, until get to a single sorted n-element list.

There are many ways to parallelize this. In the hypercube case, the algorithm consists of doing compare-exchange operations with all neighbors, pretty much in the same pattern as hyperquicksort.

### 12.3 The Bubble Sort and Its Cousins

### 12.3.1 The Much-Maligned Bubble Sort

Recall the bubble sort:

```
void bubblesort(int *x, int n)
{ for i = n-1 downto 1
    for j = 0 to i
        compare-exchange(x,i,j,n)
}
```

Here the function compare-exchange() is as in Section 12.2 .4 above. In the context here, it boils down to

```
if x[i] > x[j]
    swap x[i] and x[j]
```

In the first $\mathbf{i}$ iteration, the largest element "bubbles" all the way to the right end of the array. In the second iteration, the second-largest element bubbles to the next-to-right-end position, and so on.

You learned in your algorithms class that this is a very inefficient algorithm-when used serially. But it's actually rather usable in parallel systems.

For example, in the shared-memory setting, suppose we have one thread for each value of $\mathbf{i}$. Then those threads can work in parallel, as long as a thread with a larger value of $\mathbf{i}$ does not overtake a thread with a smaller $\mathbf{i}$, where "overtake" means working on a larger $\mathbf{j}$ value.

Once again, it probably pays to chunk the data. In this case, compare-exchange() fully takes on the meaning it had in Section 12.2.4.

### 12.3.2 A Popular Variant: Odd-Even Transposition

A popular variant of this is the odd-even transposition sort. The pseudocode for a sharedmemory version is:

```
// the argument "me" is this thread's ID
void oddevensort(int *x, int n, int me)
{ for i = 1 to n
    if i is odd
        if me is even
            compare-exchange(x,me,me+1,n)
            else // me is odd
            compare-exchange(x,me,me-1,n)
        else // i is even
            if me is even
            compare-exchange(x,me,me-1,n)
            else // me is odd
            compare-exchange(x,me,me+1,n)
```

If the second or third argument of compare-exchange() is less than 0 or greater than $\mathbf{n}$ - 1 , the function has no action.

This looks a bit complicated, but all it's saying is that, from the point of view of an even-numbered element of $\mathbf{x}$, it trades with its right neighbor during odd phases of the procedure and with its left neighbor during even phases.

Again, this is usually much more effective if done in chunks.

### 12.3.3 Example: CUDA Implementation of Odd/Even Transposition Sort

```
#include <stdio.h>
#include <stdlib.h>
#include <cuda.h>
// compare and swap; copies from the f to t, swapping f[i] and
// f[j] if the higher-index value is smaller; it is required that i < j
__device__ void cas(int *f,int *t,int i,int j, int n, int me)
{
    if (i < 0 || j >= n) return;
    if (me == i) {
```

```
        if (f[i] > f[j]) t[me] = f[j];
        else t[me] = f[i];
    } else { // me == j
        if (f[i] > f[j]) t[me] = f[i];
        else t[me] = f[j];
    }
}
// does one iteration of the sort
__global__ void oekern(int *da, int *daaux, int n, int iter)
{ int bix = blockIdx.x; // block number within grid
    if (iter % 2) {
        if (bix % 2) cas(da,daaux,bix-1,bix,n,bix);
        else cas(da,daaux,bix,bix+1,n,bix);
    } else {
        if (bix % 2) cas(da,daaux,bix,bix+1,n,bix);
        else cas(da,daaux,bix-1,bix,n,bix);
    }
}
// sorts the array ha, length n, using odd/even transp. sort;
// kept simple for illustration, no optimization
void oddeven(int *ha, int n)
{
    int *da;
    int dasize = n * sizeof(int);
    cudaMalloc((void **)&da,dasize);
    cudaMemcpy(da,ha,dasize, cudaMemcpyHostToDevice);
    // the array daaux will serve as "scratch space"
    int *daaux;
    cudaMalloc((void **)&daaux,dasize);
    dim3 dimGrid(n,1);
    dim3 dimBlock(1,1,1);
    int *tmp;
    for (int iter = 1; iter <= n; iter++) {
        oekern<<<dimGrid,dimBlock>>>(da,daaux,n,iter);
        cudaThreadSynchronize();
        if (iter < n) {
            // swap pointers
            tmp = da;
            da = daaux;
            daaux = tmp;
        } else
            cudaMemcpy(ha,daaux,dasize,cudaMemcpyDeviceToHost);
    }
}
```

Recall that in CUDA code, separate blocks of threads cannot synchronize with each other. Unless we deal with just a single block, this necessitates limiting the kernel to a single iteration of the algorithm, so that as iterations progress, execution alternates between the device and the host.

Moreover, we do not take advantage of shared memory. One possible solution would be to use _-syncthreads() within each block for most of the compare-and-exchange operations, and then having the host take care of the operations on the boundaries between blocks.

### 12.4 Shearsort

In some contexts, our hardware consists of a two-dimensional mesh of PEs. A number of methods have been developed for such settings, one of the most well known being Shearsort, developed by Sen, Shamir and the eponymous Isaac Scherson of UC Irvine. Again, the data is assumed to be initially distributed among the PEs. Here is the pseudocode:

```
for i = 1 to ceiling(log2(n)) + 1
    if i is odd
        sort each even row in descending order
        sort each odd row in ascending order
    else
        sort each column is ascending order
```

At the end, the numbers are sorted in a "snakelike" manner.
For example:

| 6 | 12 |
| :--- | :--- |
| 5 | 9 |


| 6 | 12 |
| :--- | :--- |
| 9 | 5 |


| 6 | 5 |
| :--- | :--- |
| 9 | 12 |


| 5 | $6 \downarrow$ |
| :--- | :--- |
| 12 | $\leftarrow 9$ |

No matter what kind of system we have, a natural domain decomposition for this problem would be for each process to be responsible for a group of rows. There then is the question about what to do during the even-numbered iterations, in which column operations are done. This can be handled via a parallel matrix transpose operation. In MPI, the function MPI_Alltoall() may be useful.

### 12.5 Bucket Sort with Sampling

For concreteness, suppose we are using MPI on message-passing hardware, say with 10 PEs. As usual in such a setting, suppose our data is initially distributed among the PEs.

Suppose we knew that our array to be sorted is a random sample from the uniform distribution on $(0,1)$. In other words, about $20 \%$ of our array will be in $(0,0.2), 38 \%$ will be in $(0.45,0.83)$ and so on.

What we could do is assign PE0 to the interval ( $0,0.1$ ), PE1 to $(0.1,0.2)$ etc. Each PE would look at its local data, and distribute it to the other PEs according to this interval scheme. Then each PE would do a local sort.

In general, we don't know what distribution our data comes from. We solve this problem by doing sampling. In our example here, each PE would sample some of its local data, and send the sample to PE0. From all of these samples, PE0 would find the decile values, i.e. 10th percentile, 20th percentile,..., 90th percentile. These values, called splitters would then be broadcast to all the PEs, and they would then distribute their local data to the other PEs according to these intervals.

OpenMP code for this was given in Section 1.3.2.6. Here is similar MPI code below (various improvements could be made, e.g. with broadcast):

```
// bucket sort, bin boundaries known in advance
// node 0 is manager, all else worker nodes; node 0 sends full data, bin
// boundaries to all worker nodes; i-th worker node extracts data for
// bin i-1, sorts it, sends sorted chunk back to node 0; node 0 places
// sorted results back in original array
// not claimed efficient; e.g. could be better to have manager place
// items into bins
#include <mpi.h>
#define MAX_N 100000 // max size of original data array
#define MAX_NPROCS 100 // max number of MPI processes
#define DATAMSG 0 // manager sending original data
#define BDRIES_MSG 0 // manager sending bin boundaries
#define CHUNKS_MSG 2 // workers sending their sorted chunks
int nnodes, //
    n, // size of full array
    me, // my node number
    fulldata [MAXN],
    tmp [MAXN],
    nbdries, // number of bin boundaries
    counts [MAX NPROCS];
float bdries[MAXNPROCS-2]; // bin boundaries
int debug, debugme;
init(int argc, char **argv)
{
    int i;
    debug = atoi(argv[3]);
    debugme = atoi(argv[4]);
    MPI_Init(&argc,&argv);
    MPI_Comm_size(MPLCOMM_WORLD,&nnodes );
```

```
    MPI_Comm_rank (MPLCOMM_WORLD,&me );
    nbdries = nnodes - 2;
    n = atoi(argv [1]);
    int k = atoi(argv[2]); // for random # gen
    // generate random data for test purposes
    for (i = 0; i < n; i++) fulldata[i] = rand() % k;
    // generate bin boundaries for test purposes
    for (i = 0; i < nbdries; i++) {
        bdries[i] = i * (k+1) / ((float) nnodes);
    }
}
void managernode()
{
    MPI_Status status;
    int i;
    int lenchunk; // length of a chunk received from a worker
    // send full data, bin boundaries to workers
    for (i = 1; i < nnodes; i++) {
        MPI_Send(fulldata,n,MPI_INT, i ,DATA_MSG,MPLCOMM_WORLD);
        MPI_Send(bdries,nbdries,MPIFLOAT, i ,BDRIES_MSG,MPLCOMMWWORLD);
    }
    // collect sorted chunks from workers, place them in their proper
    // positions within the original array
    int currposition = 0;
    for (i = 1; i < nnodes; i++) {
        MPI_Recv(tmp,MAXN,MPI_INT, i ,CHUNKS_MSG,MPLCOMMLWORLD,& status );
        MPI_Get_count(&status,MPI_INT,&lenchunk );
        memcpy(fulldata+currposition,tmp, lenchunk*sizeof(int));
        currposition += lenchunk;
    }
    if (n<25) {
        for (i = 0; i < n; i++) printf("%d ",fulldata [i]);
        printf("\n");
    }
}
// adds xi to the part array, increments npart, the length of part
void grab(int xi, int *part, int *npart)
{
    part[*npart] = xi;
    *npart += 1;
int cmpints(int *u, int *v)
{ if (*u<*v) return -1;
    if (*u > *v) return 1;
    return 0;
```

\}
\}

```
void getandsortmychunk(int *tmp, int n, int *chunk, int *lenchunk)
{
    int i, count = 0;
    int workernumber = me - 1;
        if (me = debugme) while (debug) ;
    for (i = 0; i < n; i++) {
            if (workernumber= 0) {
                    if (tmp[i] <= bdries[0]) grab(tmp[i],chunk,&count);
            }
            else if (workernumber < nbdries -1) {
                    if (tmp[i] > bdries[workernumber - 1] &&
                    tmp[i] <= bdries[workernumber]) grab(tmp[i], chunk,&count);
            } else
                if (tmp[i] > bdries[nbdries - 1]) grab(tmp[i],chunk,&count);
    }
    qsort(chunk, count, sizeof(int), cmpints);
    *lenchunk = count;
}
void workernode()
{
    int n,fulldata[MAXN], // size and storage of full data
        chunk [MAXN],
        lenchunk,
        nbdries; // number of bin boundaries
    float bdries[MAXNPROCS-1]; // bin boundaries
    MPI_Status status;
    MPI_Recv(fulldata ,MAXN, MPI_INT,0 ,DATA_MSG,MPLCOMMLWORLD,&status );
    MPI_Get_count(&status,MPI_INT,&n);
    MPI_Recv(bdries ,MAX_NPROCS-2,MPI_FLOAT,0 ,BDRIES_MSG,MPLCOMM_WORLD,& status );
    MPI_Get_count(&status,MPI_FLOAT,&nbdries);
    getandsortmychunk(fulldata,n,chunk,&lenchunk);
    MPI_Send (chunk, lenchunk,MPI_INT,0 ,CHUNKS_MSG,MPLCOMM_WORLD);
}
int main(int argc,char **argv)
{
    int i;
    init(argc, argv);
    if (me=0) managernode();
    else workernode();
    MPI_Finalize();
}
```


### 12.6 Radix Sort

The radix sort is essentially a special case of a bucket sort. If we have 16 threads, say, we could determine a datum's bucket by its lower 4 bits. As long as our data is uniformly distributed under the mod 16 operation, we would not need to do any sampling.

The CUDPP GPU library uses a radix sort. The buckets are formed one bit at a time, using segmented scan as above.

### 12.7 Enumeration Sort

This one is really simple. Take for instance the array $(12,5,13,18,6)$. There are 2 elements less than 12 , so in the end, it should go in position 2 of the sorted array, $(5,6,12,13,18)$.

Say we wish to sort $\mathbf{x}$, which for convenience we assume contains no tied values. Then the pseudocode for this algorithm, placing the results in $\mathbf{y}$, is

```
for all i in 0...n-1:
    count = 0
    elt = x[i]
    for all j in 0...n-1:
        if x[j] < elt then count++
    y[count] = elt
```

The outer (or inner) loop is easily parallelized.

## Chapter 13

## Parallel Computation for Audio and Image Processing

Mathematical computations involving images can become quite intensive, and thus parallel methods are of great interest. Here we will be primarily interested in methods involving Fourier analysis.

### 13.1 General Principles

### 13.1.1 One-Dimensional Fourier Series

A sound wave form graphs volume of the sound against time. Here, for instance, is the wave form for a vibrating reed $]^{1}$

[^46]

Recall that we say a function of time $\mathrm{g}(\mathrm{t})$ is periodic ("repeating," in our casual wording above) with period $T$ if if $g(u+T)=g(u)$ for all $u$. The fundamental frequency of $g()$ is then defined to be the number of periods per unit time,

$$
\begin{equation*}
f_{0}=\frac{1}{T} \tag{13.1}
\end{equation*}
$$

Recall also from calculus that we can write a function $\mathrm{g}(\mathrm{t})$ (not necessarily periodic) as a Taylor series, which is an "infinite polynomial":

$$
\begin{equation*}
g(t)=\sum_{n=0}^{\infty} c_{n} t^{n} \tag{13.2}
\end{equation*}
$$

The specific values of the $c_{n}$ may be derived by differentiating both sides of 13.2) and evaluating at $\mathrm{t}=0$, yielding

$$
\begin{equation*}
c_{n}=\frac{g^{(n)}(0)}{n!} \tag{13.3}
\end{equation*}
$$

where $g^{(j)}$ denotes the ith derivative of $g()$.
For instance, for $e^{t}$,

$$
\begin{equation*}
e^{t}=\sum_{n=0}^{\infty} \frac{1}{n!} t^{n} \tag{13.4}
\end{equation*}
$$

In the case of a repeating function, it is more convenient to use another kind of series representation, an "infinite trig polynomial," called a Fourier series. This is just a fancy name for a weighted sum
of sines and cosines of different frequencies. More precisely, we can write any repeating function $\mathrm{g}(\mathrm{t})$ with period T and fundamental frequency $f_{0}$ as

$$
\begin{equation*}
g(t)=\sum_{n=0}^{\infty} a_{n} \cos \left(2 \pi n f_{0} t\right)+\sum_{n=1}^{\infty} b_{n} \sin \left(2 \pi n f_{0} t\right) \tag{13.5}
\end{equation*}
$$

for some set of weights $a_{n}$ and $b_{n}$. Here, instead of having a weighted sum of terms

$$
\begin{equation*}
1, t, t^{2}, t^{3}, \ldots \tag{13.6}
\end{equation*}
$$

as in a Taylor series, we have a weighted sum of terms

$$
\begin{equation*}
1, \cos \left(2 \pi f_{0} t\right), \cos \left(4 \pi f_{0} t\right), \cos \left(6 \pi f_{0} t\right), \ldots \tag{13.7}
\end{equation*}
$$

and of similar sine terms. Note that the frequencies $n f_{0}$, in those sines and cosines are integer multiples of the fundamental frequency of $\mathrm{x}, f_{0}$, called harmonics.

The weights $a_{n}$ and $b_{n}, \mathrm{n}=0,1,2, \ldots$ are called the frequency spectrum of g() . The coefficients are calculated as follows $?^{2}$

$$
\begin{gather*}
a_{0}=\frac{1}{T} \int_{0}^{T} g(t) d t  \tag{13.8}\\
a_{n}=\frac{2}{T} \int_{0}^{T} g(t) \cos \left(2 \pi n f_{0} t\right) d t  \tag{13.9}\\
b_{n}=\frac{2}{T} \int_{0}^{T} g(t) \sin \left(2 \pi n f_{0} t\right) d t \tag{13.10}
\end{gather*}
$$

By analyzing these weights, we can do things like machine-based voice recognition (distinguishing one person's voice from another) and speech recognition (determining what a person is saying). If for example one person's voice is higher-pitched than that of another, the first person's weights will be concentrated more on the higher-frequency sines and cosines than will the weights of the second.

Since $g(t)$ is a graph of loudness against time, this representation of the sound is called the time domain. When we find the Fourier series of the sound, the set of weights $a_{n}$ and $b_{n}$ is said to be a

[^47]representation of the sound in the frequency domain. One can recover the original time-domain representation from that of the frequency domain, and vice versa, as seen in Equations 13.8, (13.9), 13.10) and (13.5).

In other words, the transformations between the two domains are inverses of each other, and there is a one-to-one correspondence between them. Every $g()$ corresponds to a unique set of weights and vice versa.

Now here is the frequency-domain version of the reed sound:


Note that this graph is very "spiky." In other words, even though the reed's waveform includes all frequencies, most of the power of the signal is at a few frequencies which arise from the physical properties of the reed.

Fourier series are often expressed in terms of complex numbers, making use of the relation

$$
\begin{equation*}
e^{i \theta}=\cos (\theta)+i \sin (\theta), \tag{13.11}
\end{equation*}
$$

where $i=\sqrt{-1}]^{3}$

[^48]The complex form of (13.5) is

$$
\begin{equation*}
g(t)=\sum_{j=-\infty}^{\infty} c_{j} e^{2 \pi i j \frac{t}{T}} \tag{13.12}
\end{equation*}
$$

The $c_{j}$ are now generally complex numbers. They are functions of the $a_{j}$ and $b_{j}$, and thus form the frequency spectrum.

Equation (13.12) has a simpler, more compact form than (13.5). Do you now see why I referred to Fourier series as trig polynomials? The series 13.12) involves the $\mathrm{j}^{\text {th }}$ powers of $e^{2 \pi \frac{t}{T}}$.

### 13.1.2 Two-Dimensional Fourier Series

Let's now move from sounds images. Just as we were taking time to be a continuous variable above, for the time being we are taking the position within an image to be continuous too; this is equivalent to having infinitely many pixels. Here $g()$ is a function of two variables, $g(u, v)$, where $u$ and $v$ are the horizontal and vertical coordinates of a point in the image, with $g(u, v)$ being the intensity of the image at that point. If it is a gray-scale image, the intensity is whiteness of the image at that point, typically with 0 being pure black and 255 being pure white. If it is a color image, a typical graphics format is to store three intensity values at a point, one for each of red, green and blue. The various colors come from combining three colors at various intensities.

The terminology changes a bit. Our original data is now referred to as being in the spatial domain, rather than the time domain. But the Fourier series coefficients are still said to be in the frequency domain.

### 13.2 Discrete Fourier Transforms

In sound and image applications, we seldom if ever know the exact form of the repeating function $g()$. All we have is a sampling from $g()$, i.e. we only have values of $g(t)$ for a set of discrete values of $t$.

In the sound example above, a typical sampling rate is 8000 samples per second ${ }_{\square}^{4}$ So, we may have $g(0), g(0.000125), g(0.000250), g(0.000375)$, and so on. In the image case, we sample the image

[^49]pixel by pixel.
Integrals like 13.8) now change to sums.

### 13.2.1 One-Dimensional Data

Let $X=\left(x_{0}, \ldots, x_{n-1}\right)$ denote the sampled values, i.e. the time-domain representation of g() based on our sample data. These are interpreted as data from one period of $g()$, with the period being $n$ and the fundamental frequency being $1 / \mathrm{n}$. The frequency-domain representation will also consist of n numbers, $c_{0}, \ldots, c_{n-1}$, defined as follows:

$$
\begin{equation*}
c_{k}=\frac{1}{n} \sum_{j=0}^{n-1} x_{j} e^{-2 \pi i j k / n}=\frac{1}{n} \sum_{j=0}^{n-1} x_{j} q^{j k} \tag{13.13}
\end{equation*}
$$

where

$$
\begin{equation*}
q=e^{-2 \pi i / n} \tag{13.14}
\end{equation*}
$$

again with $i=\sqrt{-1}$. The array C of complex numbers $c_{k}$ is called the discrete Fourier transform (DFT) of X. Note that (13.13) is basically a discrete analog of 13.9) and 13.10).
Note that instead of having infinitely many frequencies, we only have n of them, i.e. the n original data points $x_{j}$ map to n frequency weights $c_{k} \cdot{ }^{5}$
The quantity $q$ is a $\mathrm{n}^{\text {th }}$ root of 1 :

$$
\begin{equation*}
q^{n}=e^{-2 \pi i}=\cos (-2 \pi)+i \sin (-2 \pi)=1 \tag{13.15}
\end{equation*}
$$

Equation 13.13) can be written as

$$
\begin{equation*}
C=\frac{1}{n} A X \tag{13.16}
\end{equation*}
$$

[^50]where X is the vector $x_{j}$ and
\[

A=\left($$
\begin{array}{ccccc}
1 & 1 & 1 & \ldots & 1  \tag{13.17}\\
1 & q & q^{2} & \ldots & q^{n-1} \\
\ldots & \ldots & \ldots & \ldots & \ldots \\
1 & q^{n-1} & q^{2(n-1)} & \ldots & q^{(n-1)(n-1)}
\end{array}
$$\right)
\]

Here's R code to calculate A:

```
makeamat <- function(n,u) {
    m <- matrix(nrow=n, ncol=n)
    for (i in 1:n) {
        for (j in i:n) {
            if (i == j) {
                m[i,i] <- u^((i-1)^2)
            }
            else {
                m[i,j] <- u^((i-1)*(j-1))
                m[j,i] <- m[i,j]
            }
        }
    }
    m
}
```


### 13.2.2 Inversion

As in the continuous case, the DFT is a one-to-one transformation. so we can recover each domain from the other. The details are important:

The matrix A in 13.17 ) is a special case of Vandermonde matrices, known to be invertible. In fact, if we think of that matrix as a function of $\mathrm{q}, \mathrm{A}(\mathrm{q})$, then it turns out that

$$
\begin{equation*}
[A(q)]^{-1}=\frac{1}{n} A\left(\frac{1}{q}\right) \tag{13.18}
\end{equation*}
$$

Thus (13.16) becomes

$$
\begin{equation*}
X=n[A(q)]^{-1} C=A\left(\frac{1}{q}\right) C \tag{13.19}
\end{equation*}
$$

In nonmatrix terms:

$$
\begin{equation*}
x_{j}=\sum_{k=0}^{n-1} c_{k} e^{2 \pi i j k / n}=\sum_{k=0}^{n-1} c_{k} q^{-j k} \tag{13.20}
\end{equation*}
$$

Equation 13.20 is basically a discrete analog of 13.5.

### 13.2.2.1 Alternate Formulation

Equation (13.16) has a factor $1 / \mathrm{n}$ while (13.19) doesn't. In order to achieve symmetry, some authors of material on DFT opt to define the DFT and its inverse with $1 / \sqrt{n}$ in 13.13$)$ instead of $1 / \mathrm{n}$, and by adding a factor $1 / \sqrt{n}$ in 13.20 . They then include a factor $1 / \sqrt{n}$ in (13.17), with the result that $[A(q)]^{-1}=A(1 / q)$. Thus everything simplifies.

Other formulations are possible. For instance, the R fft() routine's documentation says it's "unnormalized," meaning that there is neither a $1 / n$ nor a $1 / \sqrt{n}$ in 13.20 . When using a DFT routine, be sure to determine what it assumes about these constant factors.

### 13.2.3 Two-Dimensional Data

The spectrum numbers $c_{r s}$ are double-subscripted, like the original data $x_{u v}$, the latter being the pixel intensity in row $u$, column $v$ of the image, $u=0,1, \ldots, n-1, v=0,1, \ldots, m-1$. Equation (13.13) becomes

$$
\begin{equation*}
c_{r s}=\frac{1}{n} \frac{1}{m} \sum_{j=0}^{n-1} \sum_{k=0}^{m-1} x_{j k} e^{-2 \pi i\left(\frac{j r}{n}+\frac{k s}{m}\right)} \tag{13.21}
\end{equation*}
$$

where $\mathrm{r}=0,1, \ldots, \mathrm{n}-1, \mathrm{~s}=0,1, \ldots, \mathrm{~m}-1$.
Its inverse is

$$
\begin{equation*}
x_{r s}=\sum_{j=0}^{n-1} \sum_{k=0}^{m-1} c_{j k} e^{2 \pi i\left(\frac{j r}{n}+\frac{k s}{m}\right)} \tag{13.22}
\end{equation*}
$$

### 13.3 Parallel Computation of Discrete Fourier Transforms

### 13.3.1 The Fast Fourier Transform

Speedy computation of a discrete Fourier transform was developed by Cooley and Tukey in their famous Fast Fourier Transform (FFT), which takes a "divide and conquer" approach:
Equation (13.13) can be rewritten as

$$
\begin{equation*}
c_{k}=\frac{1}{n}\left[\sum_{j=0}^{m-1} x_{2 j} q^{2 j k}+\sum_{j=0}^{m-1} x_{2 j+1} q^{(2 j+1) k},\right] \tag{13.23}
\end{equation*}
$$

where $m=n / 2$.
After some algebraic manipulation, this becomes

$$
\begin{equation*}
c_{k}=\frac{1}{2}\left[\frac{1}{m} \sum_{j=0}^{m-1} x_{2 j} z^{j k}+q^{k} \frac{1}{m} \sum_{j=0}^{m-1} x_{2 j+1} z^{j k}\right] \tag{13.24}
\end{equation*}
$$

where $z=e^{-2 \pi i / m}$.
A look at Equation (13.24) shows that the two sums within the brackets have the same form as Equation 13.13). In other words, Equation (13.24) shows how we can compute an n-point FFT from two $\frac{n}{2}$-point FFTs. That means that a DFT can be computed recursively, cutting the sample size in half at each recursive step.

In a shared-memory setting such as OpenMP, we could implement this recursive algorithm in the manners of Quicksort in Chapter 12 ,

In a message-passing setting, again because this is a divide-and-conquer algorithm, we can use the pattern of Hyperquicksort, also in Chapter 12 .

Some digital signal processing chips implement this in hardware, with a special interconnection network to implement this algorithm.

### 13.3.2 A Matrix Approach

The matrix form of 13.13 is

$$
\begin{equation*}
C=\frac{1}{n} A X \tag{13.25}
\end{equation*}
$$

where A is n xn . Element $(\mathrm{j}, \mathrm{k})$ of A is $q^{j k}$, while element j of X is $x_{j}$. This formulation of the problem then naturally leads one to use parallel methods for matrix multiplication, as in Chapter 11.

Divide-and-conquer tends not to work too well in shared-memory settings, because after some point, fewer and fewer threads will have work to do. Thus this matrix formulation is quite valuable.

### 13.3.3 Parallelizing Computation of the Inverse Transform

The form of the DFT (13.13) and its inverse 13.20 are very similar. For example, the inverse transform is again of a matrix form as in 13.25; ;even the new matrix looks a lot like the old one ${ }^{6}$
Thus the methods mentioned above, e.g. FFT and the matrix approach, apply to calculation of the inverse transforms too.

### 13.3.4 Parallelizing Computation of the Two-Dimensional Transform

Regroup (13.21) as:

$$
\begin{align*}
c_{r s} & =\frac{1}{n} \sum_{j=0}^{n-1}\left(\frac{1}{m} \sum_{k=0}^{m-1} x_{j k} e^{-2 \pi i\left(\frac{k s}{m}\right)}\right) e^{-2 \pi i\left(\frac{j r}{n}\right)}  \tag{13.26}\\
& =\frac{1}{n} \sum_{j=0}^{n-1} y_{j s} e^{-2 \pi i\left(\frac{j r}{n}\right)} \tag{13.27}
\end{align*}
$$

Note that $y_{j s}$, i.e. the expression between the large parentheses, is the $s^{\text {th }}$ component of the DFT of the $\mathrm{j}^{\text {th }}$ row of our data. And hey, the last expression 13.27) above is in the same form as 13.13)! Of course, this means we are taking the DFT of the spectral coefficients rather than observed data, but numbers are numbers.

[^51]In other words: To get the two-dimensional DFT of our data, we first get the one-dimensional DFTs of each row of the data, place these in rows, and then find the DFTs of each column. This property is called separability.

This certainly opens possibilities for parallelization. Each thread (shared memory case) or node (message passing case) could handle groups of rows of the original data, and in the second stage each thread could handle columns.

Or, we could interchange rows and columns in this process, i.e. put the j sum inside and k sum outside in the above derivation.

### 13.4 Available FFT Software

### 13.4.1 R

As of now, R only offers serial computation, through its function fft() . It works on both one- and two-dimensional (or more) data. If its argument inverse is set to TRUE, it will find the inverse.

Parallel computation of a two-dimensional transform can be easily accomplished by using fft() together with the approach in Section 13.3 .4 and one of the packages for parallel R in Chapter ??. Here's how to do it in snow:

```
parfft2<- function(cls ,m) {
    tmp <- parApply(cls ,m,1, fft)
    parApply(cls,tmp,1,fft)
}
```

Recall that when parApply () is called with a vector-valued function argument, the output from row $i$ of the input matrix is placed in column i of the output matrix. Thus in the second call above, we used rows (argument 1) instead of columns.

### 13.4.2 CUFFT

Remember that CUDA includes some excellent FFT routines, in CUFFT.

### 13.4.3 FFTW

FFTW ("Fastest Fourier Transform in the West") is available for free download at http://www. fftw.org. It includes versions callable from OpenMP and MPI.

### 13.5 Applications to Image Processing

In image processing, there are a number of different operations which we wish to perform. We will consider two of them here.

### 13.5.1 Smoothing

An image may be too "rough." There may be some pixels which are noise, accidental values that don't fit smoothly with the neighboring points in the image.

One way to smooth things out would be to replace each pixel intensity valu $\chi^{7}$ by the mean or median among the pixels neighbors. These could be the four immediate neighbors if just a little smoothing is needed, or we could go further out for a higher amount of smoothing. There are many variants of this.

But another way would be to apply a low-pass filter to the DFT of our image. This means that after we compute the DFT, we simply delete the higher harmonics, i.e. set $c_{r s}$ to 0 for the larger values of r and s . We then take the inverse transform back to the spatial domain. Remember, the sine and cosine functions of higher harmonics are "wigglier," so you can see that all this will have the effect of removing some of the wiggliness in our image - exactly what we wanted.

We can control the amount of smoothing by the number of harmonics we remove.
The term low-pass filter obviously alludes to the fact that the low frequencies "pass" through the filter but the high frequencies are blocked. Since we've removed the high-oscillatory components, the effect is a smoother image $8^{8}$

To do smoothing in parallel, if we just average neighbors, this is easily parallelized. If we try a low-pass filter, then we use the parallelization methods shown here earlier.

### 13.5.2 Example: Audio Smoothing in $R$

Below is code to do smoothing on sound. It inputs a sound sequence snd, and performs low-pass filtering, setting to 0 all DFT terms having $k$ greater than maxidx in (13.13).

```
p <- function(snd,maxidx) {
    four <- fft(snd)
    n <- length(four)
    newfour <- c(four[1:maxidx],rep(0,n-maxidx))
    return(Re(fft(newfour,inverse=T)/n))
}
```

[^52]
### 13.5.3 Edge Detection

In computer vision applications, we need to have a machine-automated way to deduce which pixels in an image form an edge of an object.

Again, edge-detection can be done in primitive ways. Since an edge is a place in the image in which there is a sharp change in the intensities at the pixels, we can calculate slopes of the intensities, in the horizontal and vertical directions. (This is really calculating the approximate values of the partial derivatives in those directions.)

But the Fourier approach would be to apply a high-pass filter. Since an edge is a set of pixels which are abruptly different from their neighbors, we want to keep the high-frequency components and block out the low ones.

Again, this means first taking the Fourier transform of the original, then deleting the low-frequency terms, then taking the inverse transform to go back to the spatial domain.

Below we have "before and after" pictures, first of original data and then the picture after an edge-detection process has been applied. ${ }^{9}$


[^53]

The second picture looks like a charcoal sketch! But it was derived mathematically from the original picture, using edge-detection methods.

Note that edge detection methods also may be used to determine where sounds ("ah," "ee") begin and end in speech-recognition applications. In the image case, edge detection is useful for face recognition, etc.

Parallelization here is similar to that of the smoothing case.

### 13.6 R Access to Sound and Image Files

In order to apply these transformations to sound and image files, you need to extra the actual data from the files. The formats are usually pretty complex. You can do this easily using the R tuneR and pixmap libraries.

After extracting the data, you can apply the transformations, then transform back to the time/spatial domain, and replace the data component of the original class.

### 13.7 Keeping the Pixel Intensities in the Proper Range

Normally pixel intensities are stored as integers between 0 and 255 , inclusive. With many of the operations mentioned above, both Fourier-based and otherwise, we can get negative intensity values, or values higher than 255 . We may wish to discard the negative values and scale down the positive ones so that most or all are smaller than 256 .

Furthermore, even if most or all of our values are in the range 0 to 255 , they may be near 0 , i.e. too faint. If so, we may wish to multiply them by a constant.

### 13.8 Does the Function $g()$ Really Have to Be Repeating?

It is clear that in the case of a vibrating reed, our loudness function $g(t)$ really is periodic. What about other cases?

A graph of your voice would look "locally periodic." One difference would be that the graph would exhibit more change through time as you make various sounds in speaking, compared to the one repeating sound for the reed. Even in this case, though, your voice is repeating within short time intervals, each interval corresponding to a different sound. If you say the word eye, for instance, you make an "ah" sound and then an "ee" sound. The graph of your voice would show one repeating pattern during the time you are saying "ah," and another repeating pattern during the time you are saying "ee." So, even for voices, we do have repeating patterns over short time intervals.

On the other hand, in the image case, the function may be nearly constant for long distances (horizontally or vertically), so a local periodicity argument doesn't seem to work there.

The fact is, though, that it really doesn't matter in the applications we are considering here. Even though mathematically our work here has tacitly assumed that our image is duplicated infinitely times (horizontally and vertically) ${ }^{10}$ we don't care about this. We just want to get a measure of "wiggliness," and fitting linear combinations of trig functions does this for us.

### 13.9 Vector Space Issues (optional section)

The theory of Fourier series (and of other similar transforms), relies on vector spaces. It actually is helpful to look at some of that here. Let's first discuss the derivation of (13.13).

Define X and C as in Section 13.2. X's components are real, but it is also a member of the vector space V of all n -component arrays of complex numbers.

For any complex number $\mathrm{a}+\mathrm{bi}$, define its conjugate, $\overline{a+b i}=a-b i$. Note that

$$
\begin{equation*}
\overline{e^{i \theta}}=\cos \theta-i \sin \theta==\cos (-\theta)+i \sin (-\theta)=e^{-i \theta} \tag{13.28}
\end{equation*}
$$

Define an inner product ("dot product"),

$$
\begin{equation*}
[u, w]=\frac{1}{n} \sum_{j=0}^{n-1} u_{j} \bar{w}_{j} . \tag{13.29}
\end{equation*}
$$

[^54]Define

$$
\begin{equation*}
v_{h}=\left(1, q^{-h}, q^{-2 h}, \ldots, q^{-(n-1) h}\right), h=0,1, \ldots, n-1 . \tag{13.30}
\end{equation*}
$$

Then it turns out that the $v_{h}$ form an orthonormal basis for $V{ }^{11}$ For example, to show orthnogonality, observe that for $r \neq s$

$$
\begin{align*}
{\left[v_{r}, v_{s}\right] } & =\frac{1}{n} \sum_{j=0}^{n-1} v_{r j} \overline{v_{s}}  \tag{13.31}\\
& =\frac{1}{n} \sum_{j=0} q^{j(-r+s)}  \tag{13.32}\\
& =\frac{1-q^{(-r+s) n}}{n(1-q)}  \tag{13.33}\\
& =0, \tag{13.34}
\end{align*}
$$

due to the identity $1+y+y^{2}+\ldots .+y^{k}=\frac{1-y^{k+1}}{1-y}$ and the fact that $q^{n}=1$. In the case $\mathrm{r}=\mathrm{s}$, the above computation shows that $\left[v_{r}, v_{s}\right]=1$.

The DFT of X, which we called C, can be considered the "coordinates" of X in V, relative to this orthonormal basis. The kth coordinate is then $\left[X, v_{k}\right]$, which by definition is 13.13].
The fact that we have an orthonormal basis for V here means that the matrix $\mathrm{A} / \mathrm{n}$ in $(13.25)$ is an orthogonal matrix. For real numbers, this means that this matrix's inverse is its transpose. In the complex case, instead of a straight transpose, we do a conjugate transpose, $B=\overline{A / n}^{t}$, where t means transpose. So, B is the inverse of A/n. In other words, in 13.25), we can easily get back to X from C, via

$$
\begin{equation*}
X=B C=\frac{1}{n} \bar{A}^{t} C \tag{13.35}
\end{equation*}
$$

It's really the same for the nondiscrete case. Here the vector space consists of all the possible periodic functions $g()$ (with reasonable conditions placed regarding continuity etc.) forms the vector space, and the sine and cosine functions form an orthonormal basis. The $a_{n}$ and $b_{n}$ are then the "coordinates" of g() when the latter is viewed as an element of that space.

[^55]
### 13.10 Bandwidth: How to Read the San Francisco Chronicle Business Page (optional section)

The popular press, especially business or technical sections, often uses the term bandwidth. What does this mean?

Any transmission medium has a natural range $\left[f_{\min }, f_{\max }\right]$ of frequencies that it can handle well. For example, an ordinary voice-grade telephone line can do a good job of transmitting signals of frequencies in the range 0 Hz to 4000 Hz , where " Hz " means cycles per second. Signals of frequencies outside this range suffer fade in strength, i.e are attenuated, as they pass through the phone line ${ }^{12}$

We call the frequency interval [0,4000] the effective bandwidth (or just the bandwidth) of the phone line.

In addition to the bandwidth of a medium, we also speak of the bandwidth of a signal. For instance, although your voice is a mixture of many different frequencies, represented in the Fourier series for your voice's waveform, the really low and really high frequency components, outside the range [340,3400], have very low power, i.e. their $a_{n}$ and $b_{n}$ coefficients are small. Most of the power of your voice signal is in that range of frequencies, which we would call the effective bandwidth of your voice waveform. This is also the reason why digitized speech is sampled at the rate of 8,000 samples per second. A famous theorem, due to Nyquist, shows that the sampling rate should be double the maximum frequency. Here the number 3,400 is "rounded up" to 4,000 , and after doubling we get 8,000 .

Obviously, in order for your voice to be heard well on the other end of your phone connection, the bandwidth of the phone line must be at least as broad as that of your voice signal, and that is the case.

However, the phone line's bandwidth is not much broader than that of your voice signal. So, some of the frequencies in your voice will fade out before they reach the other person, and thus some degree of distortion will occur. It is common, for example, for the letter ' f ' spoken on one end to be mis-heard as 's'on the other end. This also explains why your voice sounds a little different on the phone than in person. Still, most frequencies are reproduced well and phone conversations work well.

We often use the term "bandwidth" to literally refer to width, i.e. the width of the interval [ $\left.f_{\text {min }}, f_{\text {max }}\right]$.

There is huge variation in bandwidth among transmission media. As we have seen, phone lines have bandwidth intervals covering values on the order of $10^{3}$. For optical fibers, these numbers are more on the order of $10^{15}$.

[^56]The radio and TV frequency ranges are large also, which is why, for example, we can have many AM radio stations in a given city. The AM frequency range is divided into subranges, called channels. The width of these channels is on the order of the 4000 we need for a voice conversation. That means that the transmitter at a station needs to shift its content, which is something like in the [ 0,4000 ] range, to its channel range. It does that by multiplying its content times a sine wave of frequency equal to the center of the channel. If one applies a few trig identities, one finds that the product signal falls into the proper channel!

Accordingly, an optical fiber could also carry many simultaneous phone conversations.
Bandwidth also determines how fast we can set digital bits. Think of sending the sequence $10101010 .$. If we graph this over time, we get a "squarewave" shape. Since it is repeating, it has a Fourier series. What happends if we double the bit rate? We get the same graph, only horizontally compressed by a factor of two. The effect of this on this graph's Fourier series is that, for example, our former $a_{3}$ will now be our new $a_{6}$, i.e. the $2 \pi \cdot 3 f_{0}$ frequency cosine wave component of the graph now has the double the old frequency, i.e. is now $2 \pi \cdot 6 f_{0}$. That in turn means that the effective bandwidth of our 10101010... signal has doubled too.

In other words: To send high bit rates, we need media with large bandwidths.

## Chapter 14

## Parallel Computation in Statistics/Data Mining

How did the word statistics get supplanted by data mining? In a word, it is a matter of scale.
In the old days of statistics, a data set of 300 observations on 3 or 4 variables was considered large. Today, the widespread use of computers and the Web yield data sets with numbers of observations that are easily in the tens of thousands range, and in a number of cases even tens of millions. The numbers of variables can also be in the thousands or more.

In addition, the methods have become much more combinatorial in nature. In a classification problem, for instance, the old discriminant analysis involved only matrix computation, whereas a nearest-neighbor analysis requires far more computer cycles to complete.

In short, this calls for parallel methods of computation.

### 14.1 Itemset Analysis

### 14.1.1 What Is It?

The term data mining is a buzzword, but all it means is the process of finding relationships among a set of variables. In other words, it would seem to simply be a good old-fashioned statistics problem.

Well, in fact it is simply a statistics problem-but writ large, as mentioned earlier.
Major, Major Warning: With so many variables, the chances of picking up spurious relations
between variables is large. And although many books and tutorials on data mining will at least pay lip service to this issue (referring to it as overfitting), they don't emphasize it enough. 1

Putting the overfitting problem aside, though, by now the reader's reaction should be, "This calls for parallel processing," and he/she is correct. Here we'll look at parallelizing a particular problem, called itemset analysis, the most famous example of which is the market basket problem:

### 14.1.2 The Market Basket Problem

Consider an online bookstore that has records of every sale on the store's site. Those sales may be represented as a matrix S , whose $(\mathrm{i}, \mathrm{j})$ th element $S_{i j}$ is equal to either 1 or 0 , depending on whether the $\mathrm{i}^{\text {th }}$ sale included book $\mathrm{j}, \mathrm{i}=0,1, \ldots, \mathrm{~s}-1, \mathrm{j}=0,1, \ldots, \mathrm{t}-1$. So each row of S represents one sale, with the 1 s in that row showing which titles were bought. Each column of S represents one book title, with the 1s showing which sales transactions included that book.

Let's denote the entire line of book titles by $T_{0}, \ldots, T_{b-1}$. An itemset is just a subset of this. A frequent itemset is one which appears in many of sales transactions. But there is more to it than that. The store wants to choose some books for special ads, of the form "We see you bought books X and Y. We think you may be interested in Z."

Though we are using marketing as a running example here (which is the typical way that this subject is introduced), we will usually just refer to "items" instead of books, and to "database records" rather than sales transactions.

We have the following terminology:

- An association rule $I \rightarrow J$ is simply an ordered pair of disjoint itemsets I and $J$.
- The support of an an association rule $I \rightarrow J$ is the proportion of records which include both I and J.
- The confidence of an association rule $I \rightarrow J$ is the proportion of records which include J, among those records which include I.

Note that in probability terms, the support is basically $\mathrm{P}(\mathrm{I}$ and J$)$ while the confidence is $\mathrm{P}(\mathrm{J} \mid \mathrm{I})$. If the confidence is high in the book example, it means that buyers of the books in set I also tend to buy those in J. But this information is not very useful if the support is low, because it means that the combination occurs so rarely that it may not be worth our time to deal with it.

[^57]So, the user-let's call him/her the "data miner"-will first set thresholds for support and confidence, and then set out to find all association rules for which support and confidence exceed their respective thresholds.

### 14.1.3 Serial Algorithms

Various algorithms have been developed to find frequent itemsets and association rules. The most famous one for the former task is the Apriori algorithm. Even it has many forms. We will discuss one of the simplest forms here.

The algorithm is basically a breadth-first tree search. At the root we find the frequent 1-item itemsets. In the online bookstore, for instance, this would mean finding all individual books that appear in at least $r$ of our sales transaction records, where $r$ is our threshold.

At the second level, we find the frequent 2-item itemsets, e.g. all pairs of books that appear in at least r sales records, and so on. After we finish with level i, we then generate new candidate itemsets of size $i+1$ from the frequent itemsets we found of size i.

The key point in the latter operation is that if an itemset is not frequent, i.e. has support less than the threshold, then adding further items to it will make it even less frequent. That itemset is then pruned from the tree, and the branch ends.

Here is the pseudocode:
set $F_{1}$ to the set of 1-item itemsets whose support exceeds the threshold
for $\mathrm{i}=2$ to b
$F_{i}=\phi$
for each I in $F_{i-1}$
for each K in $F_{1}$
$Q=I \cup K$
if support( Q ) exceeds support threshold add Q to $F_{i}$
if $F_{i}$ is empty break
return $\cup_{i} F_{i}$

In other words, we are building up the itemsets of size i from those of size i-1, adding all possible choices of one element to each of the latter.

Again, there are many refinements of this, which shave off work to be done and thus increase speed. For example, we should avoid checking the same itemsets twice, e.g. first $\{1,2\}$ then $\{2,1\}$. This can be accomplished by keeping itemsets in lexicographical order. We will not pursue any refinements here.

### 14.1.4 Parallelizing the Apriori Algorithm

Clearly there is lots of opportunity for parallelizing the serial algorithm above. Both of the inner for loops can be parallelized in straightforward ways; they are "embarrassingly parallel." There are of course critical sections to worry about in the shared-memory setting, and in the message-passing setting one must designate a manager node in which to store the $F_{i}$.

However, as more and more refinements are made in the serial algorithm, then the parallelism in this algorithm become less and less "embarrassing." And things become more challenging if the storage needs of the $F_{i}$, and of their associated "accounting materials" such as a directory showing the current tree structure (done via hash trees), become greater than what can be stored in the memory of one node, say in the message-passing case.

In other words, parallelizing the market basket problem can be very challenging. The interested reader is referred to the considerable literature which has developed on this topic.

### 14.2 Probability Density Estimation

Let X denote some quantity of interest in a given population, say people's heights. Technically, the probability density function of X , typically denoted by f , is a function on the real line with the following properties:

- $f(t) \geq 0$ for all t
- for any $r<s$,

$$
\begin{equation*}
P(r<X<s)=\int_{r}^{s} f(t) d t \tag{14.1}
\end{equation*}
$$

(Note that this implies that f integrates to 1.)

This seems abstract, but it's really very simple: Say we have data on $\mathrm{X}, \mathrm{n}$ sample values $X_{1}, \ldots, X_{n}$, and we plot a histogram from this data. Then $f$ is what the histogram is estimating. If we have more and more data, the histogram gets closer and closer to the true $\mathrm{f} \cdot \square^{2}$

So, how do we estimate f , and how do we use parallel computing to reduce the time needed?

[^58]
### 14.2.1 Kernel-Based Density Estimation

Histogram computation breaks the real down into intervals, and then counts how many $X_{i}$ fall into each interval. This is fine as a crude method, but one can do better.

No matter what the interval width is, the histogram will consist of a bunch of rectanges, rather than a smooth curve. This problem basically stems from a lack of weighting on the data.

For example, suppose we are estimating $\mathrm{f}(25.8)$, and suppose our histogram interval is [24.0,26.0], with 54 points falling into that interval. Intuitively, we can do better if we give the points closer to 25.8 more weight.

One way to do this is called kernel-based density estimation, which for instance in R is handled by the function density ().

We need a set of weights, more precisely a weight function k , called the kernel. Any nonnegative function which integrates to 1 -i.e. a density function in its own right-will work. Typically k is taken to be the Gaussian or normal density function,

$$
\begin{equation*}
k(u)=\frac{1}{\sqrt{2 \pi}} e^{-0.5 u^{2}} \tag{14.2}
\end{equation*}
$$

Our estimator is then

$$
\begin{equation*}
\widehat{f}(t)=\frac{1}{n h} \sum_{i=1}^{n} k\left(\frac{t-X_{i}}{h}\right) \tag{14.3}
\end{equation*}
$$

In statistics, it is customary to use the ^symbol (pronounced "hat") to mean "estimate of." Here $\widehat{f}$ means the estimate of f .

Note carefully that we are estimating an entire function! There are infinitely many possible values of t , thus infinitely many values of $\mathrm{f}(\mathrm{t})$ to be estimated. This is reflected in 14.3), as $\widehat{f}(t)$ does indeed give a (potentially) different value for each t .

Here h , called the bandwidth, is playing a role analogous to the interval width in the case of histograms. We must choose the value of $h$, just like for a histogram we must choose the bin width $3^{3}$

Again, this looks very abstract, but all it is doing is assigning weights to the data. Consider our example above in which we wish to estimate $f(25.8)$, i.e. $t=25.8$ and suppose we choose $h$ to be 6.0. If say, $X_{88}$ is 1209.1 , very far as away from 25.8 , we don't want this data point to have much

[^59]weight in our estimation of $f(25.8)$. Well, it won't have much weight at all, because the quantity
\[

$$
\begin{equation*}
u=\frac{25.8-88}{6} \tag{14.4}
\end{equation*}
$$

\]

will be very large, and $\sqrt{14.2}$ will be tiny, as $u$ will be way, way out in the left tail.
Now, keep all this in perspective. In the end, we will be plotting a curve, just like we do with a histogram. We simply have a more sophiticated way to do this than plotting a histogram. Following are the graphs generated first by the histogram method, then by the kernel method, on the same data:



There are many ways to parallelize this computation, such as:

- Remember, we are going to compute (14.3) for many values of $t$. So, we can just have each process compute a block of those values.
- We may wish to try several different values of $h$, just as we might try several different interval widths for a histogram. We could have each process compute using its own values of $h$.
- It can be shown that (14.3) has the form of something called a convolution. The theory of convolution would take us too far afield $\int^{4}$ but this fact is useful here, as the Fourier transform of a convolution can be shown to be the product of the Fourier transforms of the two convolved components $\sqrt{5}^{5}$ In other words, this reduces the problem to that of parallelizing Fourier transforms - something we know how to do, from Chapter 13 ,

If you've seen the term before and are curious as to how this is a convolution, read on:
Write (14.3) as

$$
\begin{equation*}
\widehat{f}(t)=\sum_{i=1}^{n} \frac{1}{h} k\left(\frac{t-X_{i}}{h}\right) \cdot \frac{1}{n} \tag{14.5}
\end{equation*}
$$

Now consider two artificial random variables $U$ and $V$, created just for the purpose of facilitating computation, defined as follows.

The random variable U takes on the values ih with probability $g \cdot \frac{1}{h} k(i), \mathrm{i}=-\mathrm{c},-\mathrm{c}+1, \ldots, 0,1, \ldots, \mathrm{c}$ for some value of c that we choose to cover most of the area under $k$, with $g$ chosen so that the probabilities sum to 1 . The random variable V takes on the values $X_{1}, \ldots, X_{n}$ (considered fixed here), with probability $1 / \mathrm{n}$ each. U and V are set to be independent.

Then ( g times) 14.5 becomes $\mathrm{P}(\mathrm{U}+\mathrm{V}=\mathrm{t}$ ), exactly what convolution is about, the probability mass function (or density, in the continuous case) of a random variable arising as the sum of two independent nonnegative random variables.
${ }^{5}$ Again, if you have some background in probability and have see characteristic functions, this fact comes from

### 14.2.2 Histogram Computation for Images

In image processing, histograms are used to find tallies of how many pixels there are of each intensity. (Note that there is thus no interval width issue, as there is a separate "interval" value for each possible intensity level.) The serial pseudocode is:

```
for i = 1,\ldots,numintenslevels:
    count = 0
    for row = 1,\ldots,numrows:
        for col = 1,...,numcols:
            if image[row][col] == i: count++
    hist[i] = count
```

On the surface, this is certainly an "embarrassingly parallel" problem. In OpenMP, for instance, we might have each thread handle a block of rows of the image, i.e. parallelize the for row loop. In CUDA, we might have each thread handle an individual pixel, thus parallelizing the nested for row/col loops.

However, to make this go fast is a challenge, say in CUDA, due to issues of what to store in shared memory, when to swap it out, etc. A very nice account of fine-tuning this computation in CUDA is given in Histogram Calculation in CUDA, by Victor Podlozhnyuk of NVIDIA, 2007, http:// developer.download.nvidia.com/compute/cuda/1_1/Website/projects/histogram256/doc/histogram. pdf. The actual code is at http://developer.download.nvidia.com/compute/cuda/sdk/website/ Data-Parallel_Algorithms.html\#histogram. A summary follows:
(Much of the research into understand Podlozhnyuk's algorithm was done by UC Davis graduate student Spencer Mathews.)
Podlozhnyuk's overall plan is to have the threads compute subhistograms for various chunks of the image, then merge the subhistograms to create the histogram for the entire data set. Each thread will handle $1 / \mathrm{k}$ of the image's pixels, where k is the total number of threads in the grid, i.e. across all blocks.

In Podlozhnyuk's first cut at the problem, he maintains a separate subhistogram for each thread. He calls this version of the code histogram64. The name stems from the fact that only 64 intensity levels are used, i.e. the more significant 6 bits of each pixel's data byte. The reason for this restriction will be discussed later.

Each thread will store its subhistogram as an array of bytes; the count of pixels that a thread finds to have intensity i will be stored in the $\mathrm{i}^{\text {th }}$ byte of this array. Considering the content of a byte as an unsigned number, that means that each thread can process only 255 pixels.
the fact that the characteristic function of the sum of two independent random variables is equal to the product of the characteristic functions of the two variables.

The subhistograms will be stored together in a two-dimensional array, the $\mathrm{j}^{\text {th }}$ being the subhistogram for thread j . Since the subhistograms are accessed repeatedly, we want to store this twodimensional array in shared memory. (Since each pixel will be read only once, there would be no value in storing it in shared memory, so it is in global memory.)

The main concern is bank conflicts. As the various threads in a block write to the two-dimensional array, they may collide with each other, i.e. try to write to different locations within the same bank. But Podlozhnyuk devised a clever way to stagger the accesses, so that in fact there are no bank conflicts at all.

In the end, the many subhistograms within a block must be merged, and those merged counts must in turn be merged across all blocks. The former operation is done again by careful ordering to avoid any bank conflicts, and then the latter is done atomicAdd().

Now, why does histogram64 tabulate image intensities at only 6 -bit granularity? It's simply a matter of resource limitations. Podlozhnyuk notes that NVIDIA says that for best efficiency, there should be between 128 and 256 threads per block. He takes the middle ground, 192. With 16K of shared memory per block, $16 \mathrm{~K} / 192$ works out to about 85 bytes per thread. That eliminates computing a histogram for the full 8 -bit image data, with 256 intensity levels, which would require 256 bytes for each thread.

Accordingly, Podlozhnyuk offers histogram256, which refines the process, by having one subhistogram per warp, instead of per thread. This allows the full 8-bit data, 256 levels, to be tabulated, one word devoted to each count, rather than just one byte. A subhistogram is now a table, 256 rows by 32 columns (one column for each thread in the warp), with each table entry being 4 bytes ( 1 byte is not sufficient, as 32 threads are tabulating with it).

### 14.3 Clustering

Suppose you have data consisting of (X,Y) pairs, which when plotted look like this:


It looks like there may be two or three groups here. What clustering algorithms do is to form groups, both their number and their membership, i.e. which data points belong to which groups. (Note carefully that there is no "correct" answer here. This is merely an exploratory data analysis tool.)

Clustering is used is many diverse fields. For instance, it is used in image processing for segmentation and edge detection.

Here we have to two variables, say people's heights and weights. In general we have many variables, say p of them, so whatever clustering we find will be in p-dimensional space. No, we can't picture it very easily of p is larger than (or even equal to) 3 , but we can at least identify membership, i.e. John and Mary are in group 1, Jenny is in group 2, etc. We may derive some insight from this.

There are many, many types of clustering algorithms. Here we will discuss the famous k-means algorithm, developed by Prof. Jim MacQueen of the UCLA business school.

The method couldn't be simpler. Choose k , the number of groups you want to form, and then run this:

```
# form initial groups from the first k data points (or choose randomly)
for i = 1,...,k:
    group[i] = (x[i],y[i])
    center[i] = (x[i],y[i])
do:
```

```
    for j = 1,...,n:
        find the closest center[i] to (x[j],y[j])
        cl[j] = the i you got in the previous line
    for i = 1,\ldots,k:
        group[i] = all (x[j],y[j]) such that cl[j] = i
    center[i] = average of all (x,y) in group[i]
until group memberships do not change from one iteration to the next
```

Definitions of terms:

- Closest means in p-dimensional space, with the usual Euclidean distance: The distance from $\left(a_{1}, \ldots, a_{p}\right.$ to $\left(b_{1}, \ldots, b_{p}\right.$ is

$$
\begin{equation*}
\sqrt{\left(b_{1}-a_{1}\right)^{2}+\ldots+\left(b_{p}-a_{p}\right)^{2}} \tag{14.6}
\end{equation*}
$$

Other distance definitions could be used too, of course.

- The center of a group is its centroid, which is a fancy name for taking the average value in each component of the data points in the group. If $\mathrm{p}=2$, for example, the center consists of the point whose X coordinate is the average X value among members of the group, and whose Y coordinate is the average Y value in the group.


### 14.3.1 Example: k-Means Clustering in R

In terms of parallelization, again we have an embarrassingly parallel problem. Here's snow code for it:

```
# snow version of k-means clustering problem
# returns distances from x to each vector in y;
# here x is a single vector and y is a bunch of them
#
# define distance between 2 points to be the sum of the absolute values
# of their componentwise differences; e.g. distance between (5,4.2) and
# (3,5.6) is 2 + 1.4 = 3.4
dst <- function(x,y) {
    tmpmat <- matrix(abs(x-y), byrow=T, ncol=length(x)) # note recycling
    rowSums(tmpmat)
}
# will check this worker's mchunk matrix against currctrs, the current
# centers of the groups, returning a matrix; row j of the matrix will
# consist of the vector sum of the points in mchunk closest to j-th
# current center, and the count of such points
findnewgrps <- function(currctrs) {
```

```
    ngrps <- nrow(currctrs)
    spacedim <- ncol(currctrs) # what dimension space are we in?
    # set up the return matrix
    sumcounts <- matrix(rep(0,ngrps*(spacedim + 1)), nrow=ngrps)
    for (i in 1:nrow(mchunk)) {
        dsts <- dst(mchunk[i,],t(currctrs))
        j <- which.min(dsts)
        sumcounts[j,] <- sumcounts[j,] + c(mchunk[i,],1)
    }
    sumcounts
}
parkm <- function(cls ,m, niters, initcenters) {
    n <- nrow (m)
    spacedim <- ncol(m) # what dimension space are we in?
    # determine which worker gets which chunk of rows of m
    options(warn=-1)
    ichunks <- split(1:n,1:length(cls))
    options(warn=0)
    # form row chunks
    mchunks <- lapply(ichunks,function(ichunk) m[ichunk,])
    mcf <- function(mchunk) mchunk <<- mchunk
    # send row chunks to workers; each chunk will be a global variable at
    # the worker, named mchunk
    invisible(clusterApply(cls,mchunks,mcf))
    # send dst() to workers
    clusterExport(cls,"dst")
    # start iterations
    centers <- initcenters
    for (i in 1:niters) {
        sumcounts <- clusterCall(cls,findnewgrps, centers)
        tmp <- Reduce("+",sumcounts)
        centers <- tmp[,1:spacedim] / tmp[,spacedim+1]
        # if a group is empty, let's set its center to 0s
        centers[is.nan(centers)] <- 0
    }
    centers
}
```


### 14.4 Principal Component Analysis (PCA)

Consider data consisting of (X,Y) pairs as we saw in Section 14.3. Suppose X and Y are highly correlated with each other. Then for some constants c and d,

$$
\begin{equation*}
Y \approx c+d X \tag{14.7}
\end{equation*}
$$

Then in a sense there is really just one random variable here, as the second is nearly equal to some linear combination of the first. The second provides us with almost no new information, once we have the first. In other words, even though the vector (X,Y) roams in two-dimensional space, it usually sticks close to a one-dimensional object, namely the line (14.7).

Now think again of p variables. It may be the case that there exist $\mathrm{r}<\mathrm{p}$ variables, consisting of linear combinations of the p variables, that carry most of the information of the full set of p variables. If $r$ is much less than $p$, we would prefer to work with those $r$ variables. In data mining, this is called dimension reduction.

It can be shown that we can find these r variables by finding the r eigenvectors corresponding to the $r$ largest eigenvalues of a certain matrix. So again we have a matrix formulation, and thus parallelizing the problem can be done easily by using methods for parallel matrix operations. We discussed parallel eigenvector algorithms in Section 11.6 .

### 14.5 Monte Carlo Simulation

Monte Carlo simulation is typically (though not always) used to find probabilistic quantities such as probabilities and expected values. Consider a simple example problem:

An urn contains blue, yellow and green marbles, in numbers 5, 12 and 13, respectively. We choose 6 marbles at random. What is the probability that we get more yellow marbles than than green and more green than blue?

We could find the approximate answer by simulation:

```
count = 0
for i = 1,\ldots,n
    simulate drawing 6 marbles
    if yellows > greens > blues then count = count + 1
calculate approximate probability as count/n
```

The larger n is, the more accurate will be our approximate probability.
At first glance, this problem seems quite embarrassingly parallel. Say we are on a shared memory machine running 10 threads and wish to have $\mathrm{n}=100000$. Then we simply have each of our threads run the above code with $\mathrm{n}=10000$, and then average our 10 results.

The trouble with this, though, is that it assumes that the random numbers used by each thread are independent of the others. A naive approach, say by calling random() in the C library, will not achieve such independence. With some random number libraries, in fact, you'll get the same stream for each thread, certainly not what you want.

A number of techniques have been developed for generating parallel independent random number streams. We will not pursue the technical details here, but will give links to code for them.

- The NVIDIA CUDA SDK includes a parallel random number generator, the Mersenne Twister. The CURAND library has more.
- RngStream can be used with, for example, OpenMP and MPI.
- SPRNG is aimed at MPI, but apparently usable in shared memory settings as well. Rsprng is an $R$ interface to SPRNG.
- OpenMP: An OpenMP version of the Mersenne Twister is available at http://www.pgroup. com/lit/articles/insider/v2n2a4.htm. Other parallel random number generators for OpenMP are available via a Web search.

There are many, many more.

## Appendix A

## Miscellaneous Systems Issues

The material in this appendix pops up often throughout the book, so it's worth the time spent. For further details, see my computer systems book, http://heather.cs.ucdavis.edu/~matloff/50/ PLN/CompSystsBook.pdf.

## A. 1 Timesharing

## A.1.1 Many Processes, Taking Turns

Suppose you and someone else are both using the computer pc12 in our lab, one of you at the console and the other logged in remotely. Suppose further that the other person's program will run for five hours! You don't want to wait five hours for the other person's program to end. So, the OS arranges things so that the two programs will take turns running, neither of them running to completion all at once. It won't be visible to you, but that is what happens.

Timesharing involves having several programs running in what appears to be a simultaneous manner. (These programs could be from different users or the same user; in our case with threaded code, several processes actually come from a single invocation of a program.) If the system has only one CPU, which we'll assume temporarily, this simultaneity is of course only an illusion, since only one program can run at any given time, but it is a worthwhile illusion, as we will see.

First of all, how is this illusion attained? The answer is that we have the programs all take turns running, with each turn - called a quantum or timeslice - being of very short duration, for example 50 milliseconds. (We'll continue to assume 50 ms quanta below.)

Say we have four programs, $\mathbf{u}, \mathbf{v}$, $\mathbf{x}$ and $\mathbf{y}$, running currently. What will happen is that first $\mathbf{u}$ runs for 50 milliseconds, then $\mathbf{u}$ is suspended and $\mathbf{v}$ runs for 50 milliseconds, then $\mathbf{v}$ is suspended
and $\mathbf{x}$ runs for 50 milliseconds, and so on. After $\mathbf{y}$ gets its turn, then $\mathbf{u}$ gets a second turn, etc. Since the turn-switching, formally known as context-switching ${ }^{1}$ is happening so fast (every 50 milliseconds), it appears to us humans that each program is running continuously (though at one-fourth speed), rather than on and off, on and off, etc. $2^{2}$

But how can the OS enforce these quanta? For example, how can the OS force the program u above to stop after 50 milliseconds? The answer is, "It can't! The OS is dead while $\mathbf{u}$ is running." Instead, the turns are implemented via a timing device, which emits a hardware interrupt at the proper time. For example, we could set the timer to emit an interrupt every 50 milliseconds. When the timer goes off, it sends a pulse of current (the interrupt) to the CPU, which is wired up to suspend its current process and jump to the driver of the interrupting device (here, the timer). Since the driver is in the OS, the OS is now running!

We will make such an assumption here. However, what is more common is to have the timer interrupt more frequently than the desired quantum size. On a PC, the 8253 timer interrupts 100 times per second. Every sixth interrupt, the OS will perform a context switch. That results in a quantum size of 60 milliseconds. But this can be changed, simply by changing the count of interrupts needed to trigger a context switch.

The timer device driver saves all u's current register values, including its Program Counter value (the address of the current instruciton) and the value in its EFLAGS register (flags that record, for instance, whether the last instruction produced a 0 result). Later, when u's next turn comes, those values will be restored, and $\mathbf{u}$ including its PC value and the value in its EFLAGS register. Later, when u's next turn comes, those values will be restored, and $\mathbf{u}$ will resume execution as if nothing ever happened. For now, though, the OS routine will restore v's's previously-saved register values, making sure to restore the PC value last of all. That last action forces a jump from the OS to $\mathbf{v}$, right at the spot in $\mathbf{v}$ where $\mathbf{v}$ was suspended at the end of its last quantum. (Again, the CPU just "minds its own business," and does not "know" that one program, the OS, has handed over control to another, $\mathbf{v}$; the CPU just keeps performing its fetch/execute cycle, fetching whatever the PC points to, oblivious to which process is running.)

A process' turn can end early, if the current process voluntarily gives us control of the CPU. Say the process reaches a point at which it is supposed to read from the keyboard, with the source code calling, say, $\operatorname{scanf}()$ or cin. The process will make a systems call to do this (the compiler placed that there), which means the OS will now be running! The OS will mark this process as being in Sleep state, meaning that it's waiting for some action. Later, when the user for that process hits a key, it will cause an interrupt, and since the OS contains the keyboard device driver, this means the OS will then start running. The OS will change the process' entry in the process table from Sleep to Run-meaning only that it is ready to be given a turn. Eventually, after some other process'

[^60]turn ends, the OS will give this process its next turn.
On a multicore machine, several processes can be physically running at the same time, but the operation is the same as above. On a Linux system, to see all the currently running threads, type
ps -eLf

## A. 2 Memory Hierarchies

## A.2.1 Cache Memory

Memory (RAM) is usually not on the processor chip, which makes it "far away." Signals must go through thicker wires than the tiny ones inside the chip, which slows things down. And of course the signal does have to travel further. All this still occurs quite quickly by human standards, but not relative to the blinding speeds of modern CPUs.

Accordingly, a section of the CPU chip is reserved for a cache, which at any given time contains a copy of part of memory. If the requested item (say $\mathbf{x}$ above) is found in the cache, the CPU is in luck, and access is quick; this is called a cache hit. If the CPU is not lucky (a cache miss), it must bring in the requested item from memory.

Caches organize memory by chunks called blocks. When a cache miss occurs, the entire block containing the requested item is brought into the cache. Typically a block currently in the cache must be evicted to make room for the new one.

## A.2.2 Virtual Memory

Most modern processor chips have virtual memory (VM) capability, and most general-purpose OSs make use of it.

## A.2.2.1 Make Sure You Understand the Goals

VM has the following basic goals:

## - Overcome limitations on memory size:

We want to be able to run a program, or collectively several programs, whose memory needs are larger than the amount of physical memory available.

- Relieve the compiler and linker of having to deal with real addresses

We want to facilitate relocation of programs, meaning that the compiler and linker do not have to worry about where in memory a program will be loaded when it is run. They can, say, arrange for every program to be loaded into memory starting at address 20200, without fear of conflict, as the actualy address will be different.

## - Enable security:

We want to ensure that one program will not accidentally (or intentionally) harm another program's operation by writing to the latter's area of memory, read or write to another program's I/O streams, etc.

## A.2.2.2 How It Works

Suppose a variable $\mathbf{x}$ has the virtual address 1288 , i.e. $\& x=1288$ in a $C / C++$ program. But, when the OS loads the program into memory for execution, it rearranges everything, and the actual physical address of $\mathbf{x}$ may be, say, 5088 .

The high-order bits of an address are considered to be the page number of that address, with the lower bits being the offset within the page. For any given item such as $\mathbf{x}$, the offset is the same in both its virtual and physical addresses, but the page number differs.

To illustrate this simply, suppose that our machine uses base-10 numbering instead of base-2, and that page size is 100 bytes. Then $\mathbf{x}$ above would be in offset 88 of virtual page 12. Its physical page would be 50 , with the same offset. In other words, $\mathbf{x}$ is stored 88 bytes past the beginning of page 50 in memory.

The correspondences between virtual and physical page numbers is given in the page table, which is simply an array in the OS. The OS will set up this array at the time it loads the program into memory, so that the virtual-to-physical address translations can be done.

Those translations are done by the hardware. When the CPU executes a machine instruction that specifies access to 1288 , the CPU will do a lookup on the page table, in the entry for virtual page 12 , and find that the actual page is 50 . The CPU will then know that the true location is 5088 , and it would place 5088 in the address lines in the system bus to then access 5088 .

On the other hand, x may not currently be resident in memory at all, in which case the page table will mark it as such. If the CPU finds that page 12 is nonresident, we say a page fault occurs, and this will cause an internal interrupt, which in turn will cause a jump to the operating system (OS). The OS will then read the page containing x in from disk, place it somewhere in memory, and then update the page table to show that virtual page 12 is now in some physical page in memory. The OS will then execute an interrupt return instruction, and the CPU will restart the instruction which triggered the page fault.

## A.2.3 Performance Issues

Upon a cache miss, the hardware will need to read an entire block from memory, and if an eviction is involved, an entire block will be written as well, assuming a write-back policy. (See the reference at the beginning of this appendix.) All this is obviously slow.

The cach ${ }^{3}$ is quite small relative to memory, so you might guess that cache misses are very frequent. Actually, though, they aren't, due to something called locality of reference. This term refers to the fact that most programs tend to either access the same memory item repeatedly within short time periods (temporal locality), and/or access items within the same block often during short periods (spatial locality). Hit rates are typically well above $90 \%$. Part of this depends on having a good block replacement policy, which decides which block to evict (hopefully one that won't be needed again soon!).

A page fault is pretty catastrophic in performance terms. Remember, the disk speed is on a mechanical scale, not an electronic one, so it will take quite a while for the OS to service a page fault, much worse than for a cache miss. So the page replacement policy is even more important as well.

On Unix-family machines, the time command not only tells how long your program ran, but also how many page faults it caused. Note that since the OS runs every time a page fault occurs, it can keep track of the number of faults. This is very different from a cache miss, which although seems similar to a page fault in many ways, the key point is that a cache miss is handled solely in hardware, so no program can count the number of misses.$^{4}$

Note that in a VM system each memory access becomes two memory accesses - the page table read and the memory access itself. This would kill performance, so there is a special cache just for the page table, called the Translation Lookaside Buffer.

## A. 3 Array Issues

## A.3.1 Storage

It is important to understand how compilers store arrays in memory, an overview of which will now be presented.

Consider the array declaration

```
int y[100];
```

[^61]The compiler will store this in 100 consecutive words of memory. You may recall that in C/C++, an expression consisting of an array name, no subscript, is a pointer to the array. Well, more specifically, it is the address of the first element of the array.

An array element, say $y[8]$ actually means the same as the $\mathrm{C} / \mathrm{C}++$ pointer expression $\mathrm{y}+8$, which in turn means "the word 8 ints past the beginning of $\mathbf{y}$."

Two-dimensional arrays, say

```
int z[3][10];
```

exist only in our imagination. They are actually treated as one-dimensional arrays, in the above case consisting of $3 \times 10=30$ elements. C/C ++ arranges this in row-major order, meaning that all of row 0 comes first, then all of row 1 and so on. So for instance z [2][5] is stored in element $10+10+5$ of $\mathbf{z}$, and we could for example set that element to 8 with the code

$$
z[25]=8 ;
$$

or

$$
*(z+25)=8 ;
$$

Note that if we have a c-column two-dimensional array, element $(\mathrm{i}, \mathrm{j})$ is stored in the word $i \times c+j$ of the array. You'll see this fact used a lot in this book, and in general in code written in the parallel processing community.

## A.3.2 Subarrays

The considerations in the last section can be used to access subarrays. For example, here is code to find the sum of a float array of length $k$ :

```
float sum(float x,int k)
{ float s = 0.0; int i;
    for (i = 0; i < k; i++) s += x[i];
    return s;
}
```

Quite ordinary, but suppose we wish to find the sum in row 2 of the two-dimensional array $\mathbf{z}$ above. We could do this as $\operatorname{sum}(z+20,10)$.

## A.3.3 Memory Allocation

Very often one needs to set up an array whose size is not known at compile time. You are probably accustomed to doing this via malloc() or new. However, in large parallel programs, this approach may be quite slow.

With an array with size known at compile time, and which is declared local to some function, it will be allocated on the stack and you might run out of stack space. The easiest solution is probably to make the array global, of fixed size.

To accommodate larger arrays under gcc on a 64 -bit system, use the -mcmodel=medium command line option.

## Appendix B

## Review of Matrix Algebra

This book assumes the reader has had a course in linear algebra (or has self-studied it, always the better approach). This appendix is intended as a review of basic matrix algebra, or a quick treatment for those lacking this background.

## B. 1 Terminology and Notation

A matrix is a rectangular array of numbers. A vector is a matrix with only one row (a row vector or only one column (a column vector).

The expression, "the $(i, j)$ element of a matrix," will mean its element in row $i$, column $j$.
Please note the following conventions:

- Capital letters, e.g. A and X, will be used to denote matrices and vectors.
- Lower-case letters with subscripts, e.g. $a_{2,15}$ and $x_{8}$, will be used to denote their elements.
- Capital letters with subscripts, e.g. $A_{13}$, will be used to denote submatrices and subvectors.

If A is a square matrix, i.e. one with equal numbers n of rows and columns, then its diagonal elements are $a_{i i}, \mathrm{i}=1, \ldots, \mathrm{n}$.

The norm (or length) of an n-element vector $\mathbf{X}$ is

$$
\begin{equation*}
\|X\|=\sqrt{\sum_{i=1}^{n} x_{i}^{2}} \tag{B.1}
\end{equation*}
$$

## B.1.1 Matrix Addition and Multiplication

- For two matrices have the same numbers of rows and same numbers of columns, addition is defined elementwise, e.g.

$$
\left(\begin{array}{ll}
1 & 5  \tag{B.2}\\
0 & 3 \\
4 & 8
\end{array}\right)+\left(\begin{array}{ll}
6 & 2 \\
0 & 1 \\
4 & 0
\end{array}\right)=\left(\begin{array}{ll}
7 & 7 \\
0 & 4 \\
8 & 8
\end{array}\right)
$$

- Multiplication of a matrix by a scalar, i.e. a number, is also defined elementwise, e.g.

$$
0.4\left(\begin{array}{ll}
7 & 7  \tag{B.3}\\
0 & 4 \\
8 & 8
\end{array}\right)=\left(\begin{array}{cc}
2.8 & 2.8 \\
0 & 1.6 \\
3.2 & 3.2
\end{array}\right)
$$

- The inner product or dot product of equal-length vectors X and Y is defined to be

$$
\begin{equation*}
\sum_{k=1}^{n} x_{k} y_{k} \tag{B.4}
\end{equation*}
$$

- The product of matrices A and B is defined if the number of rows of B equals the number of columns of A (A and B are said to be conformable). In that case, the (i,j) element of the product C is defined to be

$$
\begin{equation*}
c_{i j}=\sum_{k=1}^{n} a_{i k} b_{k j} \tag{B.5}
\end{equation*}
$$

For instance,

$$
\left(\begin{array}{ll}
7 & 6  \tag{B.6}\\
0 & 4 \\
8 & 8
\end{array}\right)\left(\begin{array}{ll}
1 & 6 \\
2 & 4
\end{array}\right)=\left(\begin{array}{cc}
19 & 66 \\
8 & 16 \\
24 & 80
\end{array}\right)
$$

It is helpful to visualize $c_{i j}$ as the inner product of row i of A and column j of B , e.g. as shown in bold face here:

$$
\left(\begin{array}{ll}
\mathbf{7} & \mathbf{6}  \tag{B.7}\\
0 & 4 \\
8 & 8
\end{array}\right)\left(\begin{array}{ll}
\mathbf{1} & 6 \\
\mathbf{2} & 4
\end{array}\right)=\left(\begin{array}{ll}
\mathbf{7} & 70 \\
8 & 16 \\
8 & 80
\end{array}\right)
$$

- Matrix multiplicatin is associative and distributive, but in general not commutative:

$$
\begin{gather*}
A(B C)=(A B) C  \tag{B.8}\\
A(B+C)=A B+A C  \tag{B.9}\\
A B \neq B A \tag{B.10}
\end{gather*}
$$

## B. 2 Matrix Transpose

- The transpose of a matrix A, denoted $A^{\prime}$ or $A^{T}$, is obtained by exchanging the rows and columns of A, e.g.

$$
\left(\begin{array}{ll}
7 & 70  \tag{B.11}\\
8 & 16 \\
8 & 80
\end{array}\right)^{\prime}=\left(\begin{array}{ccc}
7 & 8 & 8 \\
70 & 16 & 80
\end{array}\right)
$$

- If $A+B$ is defined, then

$$
\begin{equation*}
(A+B)^{\prime}=A^{\prime}+B^{\prime} \tag{B.12}
\end{equation*}
$$

- If A and B are conformable, then

$$
\begin{equation*}
(A B)^{\prime}=B^{\prime} A^{\prime} \tag{B.13}
\end{equation*}
$$

## B. 3 Linear Independence

Equal-length vectors $X_{1}, \ldots, X_{k}$ are said to be linearly independent if it is impossible for

$$
\begin{equation*}
a_{1} X_{1}+\ldots+a_{k} X_{k}=0 \tag{B.14}
\end{equation*}
$$

unless all the $a_{i}$ are 0 .

## B. 4 Determinants

Let A be an nxn matrix. The definition of the determinant of $A$, $\operatorname{det}(A)$, involves an abstract formula featuring permutations. It will be omitted here, in favor of the following computational method.

Let $A_{-(i, j)}$ denote the submatrix of A obtained by deleting its $\mathrm{i}^{\text {th }}$ row and $\mathrm{j}^{\text {th }}$ column. Then the determinant can be computed recursively across the $\mathrm{k}^{\text {th }}$ row of A as

$$
\begin{equation*}
\operatorname{det}(A)=\sum_{m=1}^{n}(-1)^{k+m} \operatorname{det}\left(A_{-(k, m)}\right) \tag{B.15}
\end{equation*}
$$

where

$$
\operatorname{det}\left(\begin{array}{ll}
s & t  \tag{B.16}\\
u & v
\end{array}\right)=s v-t u
$$

## B. 5 Matrix Inverse

- The identity matrix I of size n has 1 s in all of its diagonal elements but 0 s in all off-diagonal elements. It has the property that $\mathrm{AI}=\mathrm{A}$ and $\mathrm{IA}=\mathrm{A}$ whenever those products are defined.
- The A is a square matrix and $\mathrm{AB}=\mathrm{I}$, then B is said to be the inverse of A , denoted $A^{-1}$. Then $\mathrm{BA}=\mathrm{I}$ will hold as well.
- $A^{-1}$ exists if and only if its rows (or columns) are linearly independent.
- $A^{-1}$ exists if and only if $\operatorname{det}(A) \neq 0$.
- If A and B are square, conformable and invertible, then AB is also invertible, and

$$
\begin{equation*}
(A B)^{-1}=B^{-1} A^{-1} \tag{B.17}
\end{equation*}
$$

## B. 6 Eigenvalues and Eigenvectors

Let $A$ be a square matrix. ${ }^{1}$

[^62]- A scalar $\lambda$ and a nonzero vector X that satisfy

$$
\begin{equation*}
A X=\lambda X \tag{B.18}
\end{equation*}
$$

are called an eigenvalue and eigenvector of A, respectively.

- A matrix U is said to be orthogonal if its rows have norm 1 and are orthogonal to each other, i.e. their inner product is $0 . \mathrm{U}$ thus has the property that $U U^{\prime}=I$ i.e. $U^{-1}=U$.
- If A is symmetric and real, then it is diagonalizable, i.e there exists an orthogonal matrix U such that

$$
\begin{equation*}
U^{\prime} A U=D \tag{B.19}
\end{equation*}
$$

for a diagonal matrix D . The elements of D are the eigenvalues of A , and the columns of U are the eigenvectors of A .

## Appendix C

## R Quick Start

Here we present a quick introduction to the R data/statistical programming language. Further learning resources are listed at http://heather.cs.ucdavis.edu/~/matloff/r.html.
$R$ syntax is similar to that of C . It is object-oriented (in the sense of encapsulation, polymorphism and everything being an object) and is a functional language (i.e. almost no side effects, every action is a function call, etc.).

## C. 1 Correspondences

| aspect | $\mathrm{C} / \mathrm{C}++$ | R |
| :--- | :--- | :--- |
| assignment | $=$ | <- (or $=$ ) |
| array terminology | array | vector, matrix, array |
| subscripts | start at 0 | start at 1 |
| array notation | $\mathrm{m}[2][3]$ | $\mathrm{m}[2,3]$ |
| 2-D array storage | row-major order | column-major order |
| mixed container | struct, members accessed by . | list, members acessed by $\$$ or [[ ]] |
| return mechanism | return | return() or last value computed |
| primitive types | int, float, double, char, bool | integer, float, double, character, logical |
| logical values | true, false | TRUE, FALSE (abbreviated T, F) |
| mechanism for combining modules | include, link | library() |
| run method | batch | interactive, batch |

## C. 2 Starting R

To invoke R, just type "R" into a terminal window. On a Windows machine, you probably have an R icon to click.

If you prefer to run from an IDE, you may wish to consider ESS for Emacs, StatET for Eclipse or RStudio, all open source. ESS is the favorite among the "hard core coder" types, while the colorful, easy-to-use, RStudio is a big general crowd pleaser. If you are already an Eclipse user, StatET will be just what you need.
$R$ is normally run in interactive mode, with $>$ as the prompt. Among other things, that makes it easy to try little experiments to learn from; remember my slogan, "When in doubt, try it out!"

## C. 3 First Sample Programming Session

Below is a commented R session, to introduce the concepts. I had a text editor open in another window, constantly changing my code, then loading it via R's source() command. The original contents of the file odd.R were:

```
oddcount <- function(x) {
    k <- 0 # assign 0 to k
    for (n in x) {
            if (n %% 2 = 1) k <- k+1 # %% is the modulo operator
    }
    return(k)
}
```

By the way, we could have written that last statement as simply

```
k k
```

because the last computed value of an $R$ function is returned automatically.
The R session is shown below. You may wish to type it yourself as you go along, trying little experiments of your own along the way ${ }^{1}$

```
> source("odd.R") # load code from the given file
> ls() # what objects do we have?
    [1] "oddcount"
    > # what kind of object is oddcount (well, we already know)?
> class(oddcount)
    [1] "function"
> # while in interactive mode, and not inside a function, can print
```

[^63]```
> # any object by typing its name; otherwise use print(), e.g. print(x+y)
> oddcount # a function is an object, so can print it
function(x) {
    k <- 0 # assign 0 to k
        for (n in x) {
            if (n %% 2 = 1) k <- k+1 # %% is the modulo operator
    }
    return(k)
}
> # let's test oddcount(), but look at some properties of vectors first
> y <- c(5,12,13,8,88) # c() is the concatenate function
> y
[1] 5 12 13 8 88
> y[2] #R subscripts begin at 1, not 0
[1] 12
>y[2:4] # extract elements 2, 3 and 4 of y
[1] 12 13 8
>y[c(1,3:5)] # elements 1, 3, 4 and 5
[1] 5 13 8 88
> oddcount(y) # should report 2 odd numbers
[1] 2
> # change code (in the other window) to vectorize the count operation,
> # for much faster execution
> source(" odd.R")
> oddcount
function(x) {
    x1 <- (x %% 2 = 1) # x1 now a vector of TRUEs and FALSEs
    x2 <- x[x1] # x2 now has the elements of x that were TRUE in x1
    return(length(x2))
}
> # try it on subset of y, elements 2 through 3
> oddcount(y[2:3])
[1] 1
> # try it on subset of y, elements 2, 4 and 5
> oddcount(y[c(2,4,5)])
[1] 0
> # further compactify the code
> source(" odd.R")
> oddcount
function(x) {
    length(x[x %% 2 = 1]) # last value computed is auto returned
}
> oddcount(y) # test it
[1] 2
# and even more compactification, making use of the fact that TRUE and
```

```
# FALSE are treated as 1 and 0
> oddcount <- function(x) sum(x %% 2 = 1)
# make sure you understand the steps that that involves: x is a vector,
# and thus x %% 2 is a new vector, the result of applying the mod 2
# operation to every element of x; then x %% 2 = 1 applies the =1
# operation to each element of that result, yielding a new vector of TRUE
# and FALSE values; sum() then adds them (as 1s and 0s)
# we can also determine which elements are odd
> which(y %% 2=1)
[1] 1 3
> # now have ftn return odd count AND the odd numbers themselves, using
> # the R list type
> source("odd.R")
> oddcount
function(x) {
    x1<- x[x %% 2=1]
    return(list(odds=x1, numodds=length(x1)))
}
> # R's list type can contain any type; components delineated by $
> oddcount(y)
$odds
[1] 5 13
$numodds
[1] 2
> ocy <- oddcount(y) # save the output in ocy, which will be a list
> ocy
$odds
[1] 5 13
$numodds
[1] 2
> ocy$odds
[1] 5 13
> ocy [[1]] # can get list elements using [[ ]] instead of $
[1] 5 13
> ocy[[2]]
[1] 2
```

Note that the function of the R function function() is to produce functions! Thus assignment is used. For example, here is what odd.R looked like at the end of the above session:

```
oddcount <- function(x) {
    x1<- x[x %% 2=1]
    return(list(odds=x1, numodds=length(x1)))
}
```

We created some code, and then used function() to create a function object, which we assigned to oddcount.

Note that we eventually vectorized our function oddcount(). This means taking advantage of the vector-based, functional language nature of $R$, exploiting R's built-in functions instead of loops. This changes the venue from interpreted R to C level, with a potentially large increase in speed. For example:

```
> x <- runif(1000000) # 1000000 random numbers from the interval (0, 1)
> system.time(\operatorname{sum}(x))
    user system elapsed
    0.008 0.000 0.006
> system.time({s <- 0; for (i in 1:1000000) s <- s + x[i]})
    user system elapsed
    2.776 0.004 2.859
```


## C. 4 Second Sample Programming Session

A matrix is a special case of a vector, with added class attributes, the numbers of rows and columns.

```
> # "rowbind() function combines rows of matrices; there's a cbind() too
>m1<- rbind (1:2,c(5,8))
>m1
            [,1] [, 2]
[1,] 1 2
[2,] 5 8
>rbind (m1, c(6,-1))
    [,1] [,2]
[1,] 1 2
[2,] 5 8
[3,] 6 -1
> # form matrix from 1,2,3,4,5,6, in 2 rows; R uses column-major storage
> m2<- matrix (1:6, nrow=2)
>m2
            [,1] [, 2] [, 3]
[1,] 1
[2,]
>ncol(m2)
[1] 3
nrow (m2)
[1] 2
> m2[2,3] # extract element in row 2, col 3
[1] }
# get submatrix of m2, cols 2 and 3, any row
> m3<- m2[, 2:3]
>m3
```

```
        [,1] [, 2]
[1,] 3 5
[2,] 4 6
> m1 * m3 # elementwise multiplication
    [,1] [, 2]
[1,] 3 10
[2,] 20 48
> 2.5 * m3 # scalar multiplication (but see below)
    [,1] [, 2]
[1,] 7.5 12.5
[2,] 10.0 15.0
> m1 %*% m3 # linear algebra matrix multiplication
    [,1] [, 2]
[1,] 11 17
[2,] 47 73
> # matrices are special cases of vectors, so can treat them as vectors
> sum(m1)
[1] 16
> ifelse(m2 %%3 = 1,0,m2) # (see below)
    [,1] [, 2] [, 3]
[1,] 0}<030
```

The "scalar multiplication" above is not quite what you may think, even though the result may be. Here's why:

In R, scalars don't really exist; they are just one-element vectors. However, R usually uses recycling, i.e. replication, to make vector sizes match. In the example above in which we evaluated the express $2.5 * \mathrm{~m} 3$, the number 2.5 was recycled to the matrix

$$
\left(\begin{array}{ll}
2.5 & 2.5  \tag{C.1}\\
2.5 & 2.5
\end{array}\right)
$$

in order to conform with $\mathbf{m 3}$ for (elementwise) multiplication.
The ifelse() function is another example of vectorization. Its call has the form

```
ifelse(boolean vectorexpression1, vectorexpression2, vectorexpression3)
```

All three vector expressions must be the same length, though R will lengthen some via recycling. The action will be to return a vector of the same length (and if matrices are involved, then the result also has the same shape). Each element of the result will be set to its corresponding element in vectorexpression2 or vectorexpression3, depending on whether the corresponding element in vectorexpression1 is TRUE or FALSE.

In our example above,

```
> ifelse(m2 %%3=1,0,m2) # (see below)
```

the expression $\mathrm{m} 2 \% \% 3==1$ evaluated to the boolean matrix

$$
\left(\begin{array}{lll}
T & F & F  \tag{C.2}\\
F & T & F
\end{array}\right)
$$

(TRUE and FALSE may be abbreviated to T and F .)
The 0 was recycled to the matrix

$$
\left(\begin{array}{lll}
0 & 0 & 0  \tag{C.3}\\
0 & 0 & 0
\end{array}\right)
$$

while vectorexpression3, m2, evaluated to itself.

## C. 5 Third Sample Programming Session

This time, we focus on vectors and matrices.

```
> m<- rbind(1:3,c(5,12,13)) # "row bind," combine rows
>m
    [,1] [ [, 2] [ [, 3]
[2,] 5
> t(m) # transpose
    [,1] [, 2]
[1,] 1 5
[2,] 2 12
[3,] 3 13
> ma <- m[,1:2]
>ma
lrrer
> rep(1,2) # "repeat," make multiple copies
[1] 1 1
> ma %*% rep (1,2) # matrix multiply
[1,] 3
[2,] 17
> solve(ma,c(3,17)) # solve linear system
[1] 1 1
```

```
> solve(ma) # matrix inverse
    [,1] [, 2]
[1,] 6.0 -1.0
[2,] -2.5 0.5
```


## C. 6 The R List Type

The R list type is, after vectors, the most important R construct. A list is like a vector, except that the components are generally of mixed types.

## C.6.1 The Basics

Here is example usage:

```
>g<- list (x = 4:6, s = "abc")
>g
$x
[1] 4 5 6
$s
[1] "abc"
> g$x # can reference by component name
[1] 4 5 6
> g$s
[1] "abc"
> g[[1]] # can reference by index, but note double brackets
[1] 4 5 6
> g[[2]]
[1] "abc"
> for (i in 1:length(g)) print(g[[i]])
[1] 4 5 6
[1] "abc"
```


## C.6.2 The Reduce() Function

One often needs to combine elements of a list in some way. One approach to this is to use Reduce():

```
> x <- list (4:6, c(1,6,8))
>x
[[1]]
[1] 4 5 6
```

```
[[2]]
[1] 1 6 8
>sum(x)
Error in sum(x) : invalid 'type' (list) of argument
> Reduce(sum,x)
[1] 30
```

Here Reduce() cumulatively applied R's sum() to $\mathbf{x}$. Of course, you can use it with functions you write yourself too.

Continuing the above example:

```
> Reduce(c,x)
[1] 4 5 6 1 6 8
```


## C.6.3 S3 Classes

R is an object-oriented (and functional) language. It features two types of classes, S3 and S4. I'll introduce S3 here.

An S3 object is simply a list, with a class name added as an attribute:

```
> j<- list(name="Joe", salary=55000, union=T)
> class(j) <- "employee"
> m <- list (name="Joe", salary=55000, union=F)
> class(m) <- "employee"
```

So now we have two objects of a class we've chosen to name "employee". Note the quotation marks.

We can write class generic functions:

```
print.employee <- function(wrkr) {
+ cat(wrkr$name,"\n")
+ cat("salary",wrkr$salary,"\n")
+ cat("union member",wrkr$union,"\n")
+ }
> print(j)
Joe
salary 55000
union member TRUE
> j
Joe
salary 55000
union member TRUE
```

What just happened? Well, print() in R is a generic function, meaning that it is just a placeholder for a function specific to a given class. When we printed $\mathbf{j}$ above, the R interpreter searched for a function print.employee(), which we had indeed created, and that is what was executed. Lacking this, R would have used the print function for R lists, as before:

```
> rm(print.employee) # remove the function, to see what happens with print
> j
$name
[1] "Joe"
$salary
[1] 55000
$union
[1] TRUE
attr(,"class")
[1] "employee"
```


## C.6.4 Handy Utilities

$R$ functions written by others, e.g. in base R or in the CRAN repository for user-contributed code, often return values which are class objects. It is common, for instance, to have lists within lists. In many cases these objects are quite intricate, and not thoroughly documented. In order to explore the contents of an object - even one you write yourself-here are some handy utilities:

- names(): Returns the names of a list.
- $\operatorname{str}()$ : Shows the first few elements of each component.
- summary (): General function. The author of a class $\mathbf{x}$ can write a version specific to $\mathbf{x}$, i.e. summary.x(), to print out the important parts; otherwise the default will print some bare-bones information.

For example:

```
> z <- list(a= runif(50), b = list(u=sample(1:100,25), v="blue sky"))
>
$a
    [1] 0.301676229 0.679918518 0.208713522 0.510032893 0.405027042
0.412388038
    [7] 0.900498062 0.119936222 0.154996457 0.251126218 0.928304164
0.979945937
[13] 0.902377363 0.941813898 0.027964137 0.992137908 0.207571134
0.049504986
```

```
[19] 0.092011899 0.564024424 0.247162004 0.730086786 0.530251779
0.562163986
[25] 0.360718988 0.392522242 0.830468427 0.883086752 0.009853107
0.148819125
[31] 0.381143870 0.027740959 0.173798926 0.338813042 0.371025885
0.417984331
[37] 0.777219084 0.588650413 0.916212011 0.181104510 0.377617399
0.856198893
[43] 0.629269146 0.921698394 0.878412398 0.771662408 0.595483477
0.940457376
[49] 0.228829858 0.700500359
$b
$b$u
    [1] 33 67 32 76 29 3 3 42 54 97 41 57 87 36 92 81 31 78 12 85 73 26 44
8640 43
$b$v
[1] "blue sky"
> names(z)
[1] "a" "b"
> str(z)
List of 2
    $ a: num [1:50] 0.302 0.68 0.209 0.51 0.405 \ldots.
    $ b:List of 2
        ..$ u: int [1:25] 33 67 32 76 29 3 42 54 97 41 ...
    ..$ v: chr "blue sky"
> names(z$b)
[1] "u" "v"
> summary(z)
    Length Class Mode
a 50 -none- numeric
b 2 -none- list
```


## C. 7 Graphics

$R$ excels at graphics, offering a rich set of capabilities, from beginning to advanced. In addition to the functions in base R, extensive graphics packages are available, such as lattice and ggplot2.

One point of confusion for beginniners involves saving an R graph that is currently displayed on the screen to a file. Here is a function for this, which I include in my R startup file, .Rprofile, in my home directory:

```
pr2file
function (filename)
{
    origdev <- dev.cur()
```

```
    parts <- strsplit(filename, ".", fixed = TRUE)
    nparts <- length(parts[[1]])
suff <- parts[[1]][nparts]
if (suff == "pdf") {
    pdf(filename)
}
else if (suff="png") {
    png(filename)
}
else jpeg(filename)
devnum <- dev.cur()
dev.set(origdev)
dev.copy(which = devnum)
dev.set(devnum)
dev.off()
dev.set(origdev)
}
```

The code, which I won't go into here, mostly involves manipulation of various R graphics devices. I've set it up so that you can save to a file of type either PDF, PNG or JPEG, implied by the file name you give.

## C. 8 Other Sources for Learning R

There are tons of resources for R on the Web. You may wish to start with the links at http: //heather.cs.ucdavis.edu/~matloff/r.html.

## C. 9 Online Help

R's help() function, which can be invoked also with a question mark, gives short descriptions of the R functions. For example, typing

```
> ?rep
```

will give you a description of R's rep() function.
An especially nice feature of $R$ is its example() function, which gives nice examples of whatever function you wish to query. For instance, typing

```
> example(wireframe())
```

will show examples - R code and resulting pictures - of wireframe(), one of R's 3-dimensional graphics functions.

## C. 10 Debugging in $R$

The internal debugging tool in $\mathrm{R}, \operatorname{debug}()$, is usable but rather primitive. Here are some alternatives:

- The RStudio IDE has a built-in debugging tool.
- The StatET IDE for R on Eclipse has a nice debugging tool. Works on all major platforms, but can be tricky to install.
- My own debugging tool, debugR, is extensive and easy to install, but for the time being is limited to Linux, Mac and other Unix-family systems. See http://heather.cs.ucdavis.edu/debugR.html.


## C. 11 Complex Numbers

If you have need for complex numbers, $R$ does handle them. Here is a sample of use of the main functions of interest:

```
za <- complex(real=2,imaginary=3.5)
za
[1] 2+3.5 i
>bb<- complex(real=1, imaginary=-5)
z zb
[1] 1-5i
> za * zb
[1] 19.5-6.5i
>Re(za)
[1] 2
> Im(za)
[1] 3.5
za^2
[1] -8.25+14i
> abs(za)
[1] 4.031129
> exp(complex (real=0,imaginary=pi/4))
[1] 0.7071068+0.7071068i
> cos(pi/4)
[1] 0.7071068
> sin(pi/4)
[1] 0.7071068
```

Note that operations with complex-valued vectors and matrices work as usual; there are no special complex functions.


[^0]:    ${ }^{1}$ The terminology gets confusing here. Although each core is a complete processor, people in the field tend to call the entire chip a "processor," referring to the cores, as, well, cores. In this book, the term processor will generally include cores, e.g. a dual-core chip will be considered to have two processors.

[^1]:    ${ }^{2}$ Here and below, the term Unix includes Linux.

[^2]:    ${ }^{3}$ There may be other processes running too. So the threads of our program must still take turns with other processes running on the machine.

[^3]:    ${ }^{4}$ Technically one could use locals in main() (or whatever function it is where the threads are created) for this purpose, but this would be so unwieldy that it is seldom done.
    ${ }^{5}$ Technically, we should say "shared by all threads" here, as a given thread does not always execute on the same processor, but at any instant in time each executing thread is at some processor, so the statement is all right.

[^4]:    ${ }^{6}$ In a more refined version, X would be parceled out to the nodes, just as the rows of A are.

[^5]:    ${ }^{1}$ You can download Gove's code from http://blogs.sun.com/d/resource/map_src.tar.bz2 Most relevant is listing7.64.c.

[^6]:    ${ }^{2}$ Note the qualifying term.
    ${ }^{3}$ This is still static, as the randomization is done at the outset, before starting computation.
    ${ }^{4}$ See Section 4.3.3.

[^7]:    ${ }^{5}$ Why are we calling it "communication" here? Recall that in shared-memory programming, the threads communicate through shared variables. When one thread increments nextchunk, it "communicates" that new value to the other threads by placing it in shared memory where they will see it, and as noted earlier contention among threads to shared memory is a major source of potential slowdown.

[^8]:    ${ }^{6}$ As noted in the Preface to this book, I occasionally refer here to my research, to illustrate for students the beneficial interaction between teaching and research.

[^9]:    ${ }^{1}$ Here thread 0 is considered "consecutive" to thread 15 , in a wraparound manner.

[^10]:    ${ }^{2}$ This sounds similar to the concept of a cache. However, it is very different. A cache contains a local copy of some data stored elsewhere. Here it is the data itself, not a copy, which is being stored locally.

[^11]:    ${ }^{3}$ For safety's sake, i.e. fault tolerance, even writes are typically acknowledged in multiprocessor systems.
    ${ }^{4}$ The picture may be cut off somewhat at the top and left edges. The upper-right output of the rectangle in the top row, leftmost position should connect to the dashed line which leads down to the second PE from the left. Similarly, the upper-left output of that same rectangle is a dashed lined, possibly invisible in your picture, leading down to the leftmost PE.

[^12]:    ${ }^{5}$ Note that the ' 1 ' in " $\mathrm{O}(1)$ " does not refer to the fact that only one communication can occur at a time. If we had, for example, a two-bus system, the bandwidth would still be $\mathrm{O}(1)$, since multiplicative constants do not matter. What $\mathrm{O}(1)$ means, again, is that as n grows, the bandwidth stays at a multiple of 1 , i.e. stays constant.

[^13]:    ${ }^{6}$ This discussion is for a mythical machine, but any real system works in this manner.

[^14]:    ${ }^{7}$ The reader may wish to review the basics of caches. See for example http://heather.cs.ucdavis.edu/~matloff/ 50/PLN/CompOrganization.pdf

[^15]:    ${ }^{8}$ We will follow commonly-used terminology here, distinguishing between a cache line and a memory block. Memory is divided in blocks, some of which have copies in the cache. The cells in the cache are called cache lines. So, at any given time, a given cache line is either empty or contains a copy (valid or not) of some memory block.
    ${ }^{9}$ Again, remember that ordinary bus arbitration methods would be used.

[^16]:    ${ }^{10}$ Many modern processors, including Pentium and MIPS, allow the programmer to mark some blocks as being noncacheable.
    ${ }^{11}$ Some protocols change between the two modes dynamically.

[^17]:    ${ }^{12}$ See Pentium Processor System Architecture, by D. Anderson and T. Shanley, Addison-Wesley, 1995. We have simplified the presentation here, by eliminating certain programmable options.

[^18]:    ${ }^{13}$ There are many variants of all of this, especially in the software distibuted shared memory realm, to be discussed later.

[^19]:    ${ }^{14}$ We call this "programmer-induced," since the programmer will include some special operation in her C/C ++ code which will be translated to MEMBAR.

[^20]:    ${ }^{15}$ As with many aspects of parallel programming, a good basic knowledge of operating systems is key. See the reference on page 7

[^21]:    ${ }^{16}$ The term object-based is not related to the term object-oriented programming.
    ${ }^{17}$ There are a number of important issues involved with this word eventually, as we will see later.
    ${ }^{18}$ The update may not occur immediately. More on this later.

[^22]:    ${ }^{19}$ Note, though, that we are not actually dealing with a cache here. Each node in the SDSM system will have a cache, of course, but a node's cache simply stores parts of that node's set of pages. The coherency across nodes is across pages, not caches. We must insure that a change made to a given page is eventually propropagated to pages on other nodes which correspond to this one.

[^23]:    ${ }^{20}$ Though, as mentioned in the comments, it is aimed more at message-passing contexts.

[^24]:    ${ }^{21}$ Writes will also be propagated at barrier operations, but two successive arrivals by a processor to a barrier can be considered to be a lock/unlock pair, by considering a departure from a barrier to be a "lock," and considering reaching a barrier to be an "unlock." So, we'll usually not mention barriers separately from locks in the remainder of this subsection.
    ${ }^{22}$ The set of changes is called a diff, remiscent of the Unix file-compare command. A copy, called a twin, had been made of the original page, which now will be used to produce the diff. This has substantial overhead. The Treadmarks people found that it took 167 microseconds to make a twin, and as much as 686 microseconds to make a diff.
    ${ }^{23}$ In JIAJIA, that location is normally fixed, but JIAJIA does include advanced programmer options which allow the location to migrate.

[^25]:    ${ }^{24}$ The only other option would be to use lock/unlock, but then their writing would not be simultaneous.
    ${ }^{25}$ If they are writing to the same variable, not just the same page, the programmer would use locks instead of a barrier, and the situation would not arise.
    ${ }^{26}$ Some hardware barriers have been proposed.
    ${ }^{27}$ I use the word processor here, but it could be just a thread on the one hand, or on the other hand a processing element in a message-passing context.

[^26]:    ${ }^{1}$ There is an issue here of thread startup time. The OMPi compiler sets up threads at the outset, so that that startup time is incurred only once. When a parallel construct is encountered, they are awakened. At the end of the construct, they are suspended again, until the next parallel construct is reached.

[^27]:    ${ }^{2}$ This is an OpenMP term. The for directive is another example of it. More on this below.

[^28]:    ${ }^{3}$ Note, though, that plain min and max would not help in our Dijkstra example above, as we not only need to find the minimum value, but also need the vertex which attains that value.

[^29]:    ${ }^{4}$ You may find certain subversions of GCC 4.1 can be used too.

[^30]:    ${ }^{5}$ If you are reading this presentation on OpenMP separately from the book, the book is at http://heather.cs. ucdavis.edu/~matloff/158/PLN/ParProcBook.pdf

[^31]:    ${ }^{1}$ This function cannot be called from the device itself. However, malloc() is available from the device, and device memory allocated by it can be copied to the host. See the NVIDIA programming guide for details.

[^32]:    ${ }^{2}$ Neither has the CPU-only version of the program been optimized. As pointed out by Bill Hsu, the row-major version of that program should run faster than the column-major one, due to cache consideration.

[^33]:    ${ }^{3}$ As February 2012, the next generation, Kepler, is soon to be released.

[^34]:    ${ }^{4}$ If you are reading this presentation on CUDA separately from the book, the book is at http://heather.cs. ucdavis.edu/~matloff/158/PLN/ParProcBook.pdf

[^35]:    ${ }^{1}$ Note that we used a .cpp suffix for the source file name, instead of .cu. Or, we can use the $-\mathrm{x} \mathbf{c u}$ if compiling with nvec.

[^36]:    ${ }^{2}$ If you search through the Thrust source code, you'll fine omp pragmas.
    ${ }^{3}$ Threads will not be set up if you use host arraysvectors.

[^37]:    ${ }^{4}$ We are still writing to temporary storage, but that will probably be in registers (since we don't create the entire map at once), thus fast to access.

[^38]:    ${ }^{1}$ Note that we number the digits from right to left, with the rightmost digit being digit 0 .

[^39]:    ${ }^{2}$ Note that this is indeed an i-dimensional cube, because the last i digits are free to vary.

[^40]:    ${ }^{3}$ The term scalable arises frequently in conversations on parallel processing. It means that this particular method of dealing with some aspect of parallel processing continues to work well as the system size increases. We say that the method scales.

[^41]:    ${ }^{1}$ Of course, some or all of these might be threads on the same machine.

[^42]:    ${ }^{2}$ The results from the various reducers will then in turn be reduced, yielding the max temps for all years.

[^43]:    ${ }^{3}$ There is a quick introduction to Python in Appendix ??

[^44]:    ${ }^{1}$ The term singular value is a synonym for eigenvalue.

[^45]:    ${ }^{1}$ See Chapter 7 for definitions of hypercube terms.

[^46]:    ${ }^{1}$ Reproduced here by permission of Prof. Peter Hamburger, Indiana-Purdue University, Fort Wayne. See http://www.ipfw.edu/math/Workshop/PBC.html

[^47]:    ${ }^{2}$ The get an idea as to how these formulas arise, see Section 13.9 . But for now, if you integrate both sides of 13.5 , you will at least verify that the formulas below do work.

[^48]:    ${ }^{3}$ There is basically no physical interpretation of complex numbers. Instead, they are just mathematical abstrac-

[^49]:    tions. However, they are highly useful abstractions, with the complex form of Fourier series, beginning with 13.12, being a case in point.

    It is not assumed that you know complex variables well. All that is required is knowledge of how to add, subtract, multiply and divide, and the definition of $|c|$ for complex c.
    ${ }^{4}$ See Section 13.10 for the reasons behind this.

[^50]:    ${ }^{5}$ Actually, in the case of $x_{j}$ real, which occurs with sound data, we really get only $\mathrm{n} / 2$ frequencies. The weight of the frequences after $\mathrm{k}=\mathrm{n} / 2$ turn out to be the conjugates of those before $\mathrm{n} / 2$, where the conjugate of $\mathrm{a}+\mathrm{bi}$ is defined to be a-bi.

[^51]:    ${ }^{6}$ In fact, one can obtain the new matrix easily from the old, as explained in Section 13.9 .

[^52]:    ${ }^{7}$ Remember, there may be three intensity values per pixel, for red, green and blue.
    ${ }^{8}$ Note that we may do more smoothing in some parts of the image than in others.

[^53]:    ${ }^{9}$ These pictures are courtesy of Bill Green of the Robotics Laboratory at Drexel University. In this case he is using a Sobel process instead of Fourier analysis, but the result would have been similar for the latter. See his Web tutorial at www.pages.drexel.edu/~weg22/edge.html, including the original pictures, which may not show up well in our printed book here.

[^54]:    ${ }^{10}$ And in the case of the cosine transform, implicitly we are assuming that the image flips itself on every adjacent copy of the image, first right-side up, then upside-own, then right-side up again, etc.

[^55]:    ${ }^{11}$ Recall that this means that these vectors are orthogonal to each other, and have length 1 , and that they span V .

[^56]:    ${ }^{12}$ And in fact will probably be deliberately filtered out.

[^57]:    ${ }^{1}$ Some writers recommend splitting one's data into a training set, which is used to discover relationships, and a validation set, which is used to confirm those relationships. It's a good idea, but overfitting can still occur even with this precaution.

[^58]:    ${ }^{2}$ The histogram must be scaled to have total area 1 . Most statistical programs have options for this.

[^59]:    ${ }^{3}$ Some statistical programs will choose default values, based on theory.

[^60]:    ${ }^{1}$ We are switching from the "context" of one program to another.
    ${ }^{2}$ Think of a light bulb turning on and off extremely rapidly, with half the time on and half off. If it is blinking rapidly enough, you won't see it go on and off. You'll simply see it as shining steadily at half brightness.

[^61]:    ${ }^{3}$ Or caches, plural, as there are often multiple levels of caches in today's machines.
    ${ }^{4}$ Note by the way that cache misses, though harmful to program speed, aren't as catastrophic as page faults, as the disk is not involved.

[^62]:    ${ }^{1}$ For nonsquare matrices, the discussion here would generalize to the topic of singular value decomposition.

[^63]:    ${ }^{1}$ The source code for this file is at http://heather.cs.ucdavis.edu/~matloff/MiscPLN/R5MinIntro.tex. You can download the file, and copy/paste the text from there.

