

Operating Systems

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Introduction

Operating systems provide an interface for computer users that permits them to gain access without needing to understand how the computer works. The software needed to achieve this is complex and this course introduces students to some of the details of design and implementation.

Aims

This course unit introduces students to the principles of operating system design and to the prevailing techniques for their implementation. The course unit assumes that students are already familiar with the structure of a user-program after it has been converted into an executable form, and that they have a rudimentary understanding of the performance trade-offs inherent in the choice of algorithms and data structures. Pertinent features of the hardware-software interface are described, and emphasis is placed on the concurrent nature of operating system activities. Two concrete examples of operating systems are used to illustrate how principles and techniques are deployed in practice.

Additional reading

Operating system concepts (8th edition)	Silberschatz, Abraham and Peter Baer Galvin and Greg Gagne	2009
Operating system concepts with Java (8th edition)	Silberschatz, Abraham and Peter Baer Galvin and Greg Gagne	2010

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1 Recap of COMP12111 & COMP15111

There material in both the *Fundamentals of Computer Architecture* and the *Fundamentals of Computer Engineering* courses in the first year provides a good base for the course this year. The following re-visits that material and builds upon it.

1.1 Datapath and Control

From the point of view of the CPU all data is of a fixed size, the length of one word. Each word is usually moved around the architecture of the computer in a bit-parallel manner, that is to say that there are at least the same number of wires in a bus between any two components in the system as there are bits in the word.

The individual operations on each bit inside a word when the CPU performs an operation on the whole word are usually identical. This results in a very regular datapath with lots of duplicated (usually by as many times as the word length) hardware logic.

Control logic is derived after the datapath has been conceived. It governs which operation is performed at what time, and is different for each instruction in the instruction set.

A typical example of control logic might be to control the enable pin on a binary adder. The datapath will direct bits to the adder all the time, but the control logic will determine if the result is sent forward.

1.2 The MU0 Instruction Set Architecture

The MU0 is a very simple 16 bit word architecture, and as a result, the instruction set is also very simple. Each instruction can address one memory location, and consists of four bits for the instruction (allowing sixteen instructions to be coded) and twelve bits for the memory address as illustrated in Figure 1. Since we can only store twelve bits of memory address in the instruction, and the architecture is very simple, the system has 2^{12} words of memory, which is equivalent to 8 kB.

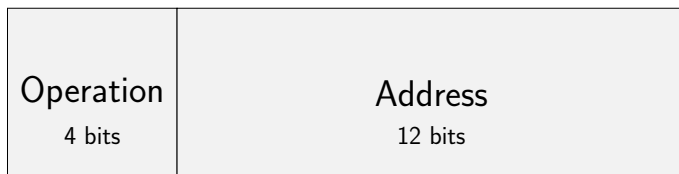


Figure 1: A generic MU0 instruction

The MU0 has two programmer visible registers, the Program Counter and the Accumulator. The Program Counter stores the address in memory of the next instruction to be executed, thus being twelve bits long. The Accumulator is sixteen bits long, and stores the result of the last arithmetic operation.

The instructions that the MU0 understands are listed in Table 1.

1.3 Maintaining Processor State

If the execution cycle of the MU0 was somehow disrupted, say because of an interrupt call, it would be handy to save the state of the processor before switching to a different task (e.g. running the interrupt handler).

The way to do this is to save the registers in memory, doing the other task, and then reloading them when it's time to resume execution of the program.

Op Code	Mnemonic	Description
0	LDA $[op]$	$[op] \rightarrow Acc$
1	STO $[op]$	$Acc \rightarrow [op]$
2	ADD $[op]$	$Acc = Acc + [op]$
3	SUB $[op]$	$Acc = Acc - [op]$
4	JMP $[op]$	$PC = S$
5	JGE $[op]$	If $Acc \geq 0$ then $PC = S$
6	JNE $[op]$	If $Acc \neq 0$ then $PC = S$
7	STP	Stop

Table 1: The MU0 instruction set

1.4 The Fetch Execute Cycle

The fetch-execute cycle describes how a CPU executes instructions. First, the next instruction is fetched from memory (at the address pointed to by the PC), then the instruction is executed. Since some instructions access memory (such as load and store), and we can only do one memory access per clock cycle, one fetch-execute cycle takes two clock cycles, one for fetching, and one for execution.

1.4.1 Fetching instructions

Fetching is an operation that is the same for all instructions. First memory addressed by the PC is read and stored into the Instruction Register (IR). This is a 16 bit internal register that isn't visible to programmers. Once this has occurred, the PC is incremented. This means that the RAM must be able to send a word directly to the instruction register, so a datapath must be in place to allow this.

1.4.2 Executing instructions

It is obvious that different instructions will have different paths of execution within the processor, and will have different effects on components within the system.

JMP In order to execute the JMP instruction, the last twelve bits are read from the instruction register and transferred over to the PC. This means that there must be a datapath from the bottom twelve bits of the IR to the PC.

STA When STA is executed, the bottom twelve bits in the IR are used direct the contents of the accumulator to a location in memory. To do this, we need a datapath from the bottom twelve bits of the IR to the part of the RAM that takes addresses, and from the PC to the part of the RAM that takes data.

ADD To perform the ADD instruction, we need to fetch the bottom twelve bits of the IR and send it to the RAM. The result should be fed into the adder along with the contents of the accumulator. The result of the calculation should be sent to the accumulator. To do this, we need datapaths from the accumulator to the ALU, the RAM to the ALU and finally from the ALU to the accumulator.

Control Signals whenever two separate components within the system interact. For example, every time the CPU loads a word from the RAM, a control signal must be sent to say 'load', and every time the ADD command it executed, the ALU must be sent a control signal to say 'add' as opposed to subtract or shift.

Timing Timing is very important when executing the instructions. If the result of a load from RAM hasn't yet returned, but the control signal to the ALU to add is sent, then the wrong result will almost certainly occur! In order for everything to run smoothly, the critical path for each operation must be worked out, and time allowed for signals to propagate through even the longest critical path.

1.4.3 Deriving the datapaths from the operation of instruction

In order to produce a working processor, we need to look at all the instructions that can be executed by the processor, and examine what datapaths and control signals they require to work. Only when we have this information can we begin to actually design the hardware on the CPU.

Note that data going to one destination can only go to one source, so if you want multiple components to be able to send data to one other component, then you must use a multiplexer with control signals in order to achieve this.

1.5 Control Signals

The purpose of control signals is to make each component within the CPU function as intended for each specific instruction. Control signals include:

- Enable write for registers
- Enable write for memory
- Enable read for memory
- Multiplexer input select
- ALU actions (add, subtract, bypass)

Sometimes, one component (such as the ALU) may have control inputs that can be represented by more than two states (add, bypass, subtract). If this is the case, then multiple wires (a bus) is used to specify its action.

The first lab in the course shows the control signals sent for each instruction, the solution for which is shown in Tables 2 and 3.

En_IR	1	(enable write to IR)
En_PC	0	(enable write to PC)
En_ACC	X	(enable write to ACC)
byp	X	(ALU action: bypass)
add	X	(ALU action: add)
sub	X	(ALU action: subtract)
Ren	1	(RAM action: Read)
Wen	0	(RAM action: Write)
addr_Mux	1	(RAM address = PC, otherwise IR.S)

Table 2: The control signals in the MU0 fetch phase

2 What does an Operating System do?

The job of an operating system varies from system to system but on general, it is responsible for managing the resources of the system (including dealing with concurrency, security etc) and abstracting the implementation of the system from the running programs (such as what exact components are being utilised).

	lda	sta	add	sub	stp	jump	no jump
En_IR	0	0	0	0	0	0	0
En_PC	0	0	0	0	0	1	0
En_ACC	1	0	1	1	0	0	0
byp	1	X	0	0	X	X	X
add	0	X	1	0	X	X	X
sub	0	X	0	1	X	X	X
Ren	1	0	1	1	0	1	0
Wen	0	1	0	0	0	0	0
addr_Mux	0	0	0	0	X	0	1

Table 3: The control signals in the MU0 execute phase

2.1 Processes

A process is a program that is currently running on the system. It consists of a Thread (a set of instructions to be executed) and address space (a set of memory locations that can be accessed by the thread). In most systems, multi-threading is used to allow each process own multiple threads, and therefore execute in parallel.

Nearly all systems have many processes running at any one time, on a Linux system, use `htop` or `ps aux` to see what processes are running.

2.2 Address Space

Address space (aka memory space) is a term used to speak about a section of memory. This could be the whole memory available to the system, the memory that a specific program has access to etc.

When a program starts, it assumes that it does so from memory address 0. On a single process system this is okay, however this presents a problem on systems where multiple processes run concurrently, since no two processes can share the same memory space.

Sometimes, operating systems may even running pause programs, move them out of memory (onto secondary memory such as hard drive) and later on swap it back in at a different place in memory.

In both cases, a technique called *Relocation* is used to make every running program able to safely assume that it has sole use of memory.

In order to facilitate relocation, operating systems abstract away the implementation of the hardware, and instead provide a virtual machine for each program. This enables programs to behave as though they have the whole system to themselves, and it also lets the operating system easily stop programs interfering with each other (such as providing disjoint memory spaces for each program).

2.3 Modes of operation

It is often necessary to prevent some programs from executing some operations, such as manipulating memory, or allocating CPU time. In order to achieve this, operating systems nearly always implement different ‘modes’ of operation that processes can run under. The two most common modes are *user* and *system*.

All the processes owned by the operating system will run under system mode, which is very permissive and lets programs perform operations with the potential for misuse. Programs that the user might run are usually executing under user mode. User mode is less permissive, and restricts certain operations, yet the restricted operations aren’t usually required for normal programs.

2.3.1 System calls

If there was an eventuality where a program needed to perform privileged operation that wasn't permitted under its current mode, then it can use a system call to achieve the same result. The premise is that the operating system will provide a 'gatekeeper' function that will perform the requested operation, but only after the parameters have undergone checks to ensure that the application isn't behaving badly. The execution of the user program will (of course) pause while the system call is running.

Lots of functions in languages that you already know might just be wrappers around system calls, albeit often with slightly more functionality. The course notes make a good example; `fread` in C uses the UNIX system call `read`.

3 Engineering an Operating System

Like a lot of things in Computer Science, operating systems can often be conceptualised as being built up of several layers. As you inspect further and further into the OS, looking deeper into each layer as you go, the level of abstraction decreases.

The outermost layer could be seen to be the UI, which is obviously a very abstracted way of thinking about a computer. The kernel is probably the lowest layer, since the details of how the hardware is managed is, also fairly obviously, a very low level of abstraction.

Different operating systems can contain a different number of layers of abstraction so that they are best suited to their purpose. Some operating systems will contain little more than a microkernel, which will have the bare minimum of logic required to keep a computer running.

Some components of an operating system are monolithic, most notably the Linux kernel. A monolithic component is easier to design (especially for a kernel) since there is less inter-process communication to worry about, and trap falls like race conditions can be more easily avoided when everything is running on a single thread.

I feel obliged to point out that monolithic, in this instance, refers to the fact that the program is running under *one* process that is in system mode. It doesn't mean that the code is all in one big file, or even in one big project, since many monolithic projects employ some degree of modularity in the development process at least.

See this StackOverflow answer for an explanation (and follow the link to the Linus V Tanenbaum showdown on the topic):
<http://stackoverflow.com/questions/1806585/why-is-linux-called-a-monolithic-kernel>

3.1 Managing processes

When operating systems were first developed, they usually ran one process at a time. This was obviously limiting for the users, since only one person could do stuff at a time. In light of this, timesharing was developed, which is a way for the computer to be used by multiple users at once, even here, users would all share one CPU and one PC though.

Now, processes execute within their own virtual CPU, while the real CPU switches back and forth giving time to each task. This is good, since the order with which the CPU gives its time to processes can be used to effectively make some processes run faster than others.

We have already learnt that a process is made up of a thread and some address space, but we've not been over how a process is created. In order to do this, an existing process must create new child process.

(Processes also have register values and external interfaces too, but they're not as important)

Processes can be killed by themselves (with a variety of different exit codes), by other processes (using something like the `kill` command), or in some systems, by the parent process terminating.

The lifecycle of a process, and the states it occupies can be represented in a state machine like diagram, as shown in Figure 2.

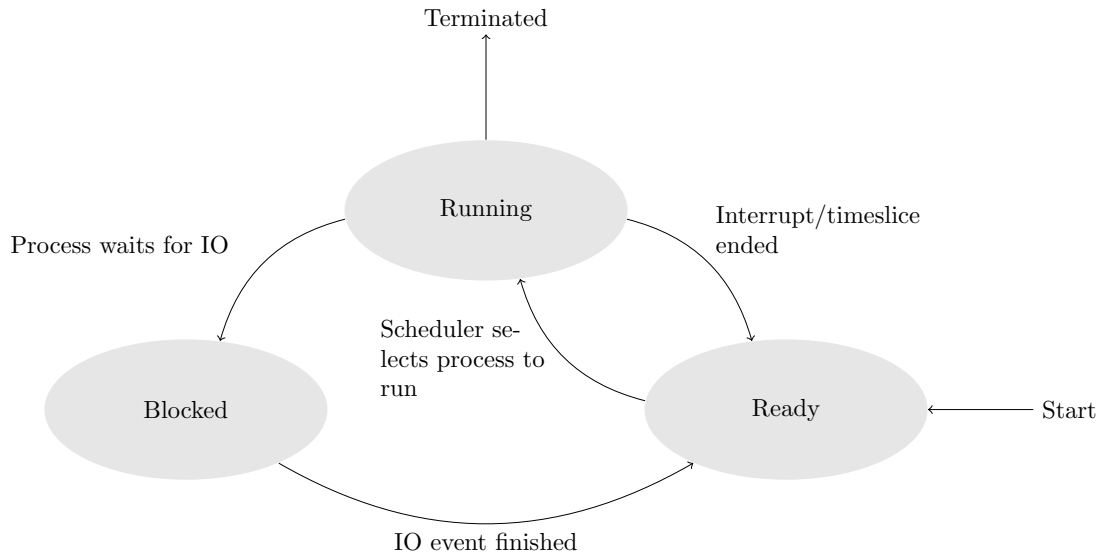


Figure 2: The lifecycle of a process

3.1.1 Scheduling

One particularly troublesome section of the lifecycle of a process is going from the *ready* state to the *running* state (and vice versa). In order to do this, a process to be resumed must be chosen from a pool of processes waiting for CPU time, and the registers, IO operations etc must be loaded and made ready.

In order to do this, a PCB (Process Control Block) table is maintained, which contains all the necessary information for each process to be paused and resumed. This includes (among other things):

- Process ID
- Parent ID
- Saved registers
- Memory, IO management information
- CPU scheduling info

On some operating systems, timeslices are given to processes, where on others, they are assigned to threads. Henceforth, processes can often be made more efficient by using threads in order to maximise the use of their timeslice (switching to non-blocked threads whenever there is a block).

In order to handle the complicated tasks of scheduling, operating systems have a dedicated component as part of the process manager to do this. It's main job is decide which process should run next on what core of the CPU, while minimising the average wait and turnaround times for processes to execute.

Processes alternate between CPU (expensive computation) and IO (blocked on IO) bursts. Processes that have long CPU bursts are said to be CPU bound, while processes that use lots of IO are said to be IO bound. Processes can change their characteristics while running (if a web browser is using lots of cached content, it might be IO bound, but if it's rendering 4K video with JavaScript, it's probably CPU bound).

Here are some scheduling algorithms:

First Come First Serve (FCFS)

This is when processes get all of the CPU time until they are finished, or they block. It's easy to implement, since all you need is a queue of processes:



Figure 3: What an FCFS implementation might behave like

FCFS is a **non-preemptive** algorithm, since processes are allowed to run until they finish with the CPU and start doing IO stuff, or exit.

Round Robin

The round robin algorithm is a **preemptive** algorithm. Each process is given a time slice when it starts processing on the CPU, and if it hasn't finished by the end of its time slice, then the CPU is given to another process. It's still a very simple algorithm, and it's much more efficient than FCFS.

'Time slice' is also known as 'Time Quantum' and its length is very important. Smaller time slices are affected greater by the time it takes to perform a context switch. The best solution is to try and decrease the cost of context switching and increase the time slice length a little (though not too much, otherwise you end up with pseudo-FCFS)



Figure 4: What a Round Robin implementation might behave like

Shortest job first

We could just run whichever process will have the shortest CPU burst first. The problem with this one, is it can be hard to tell how long processes will use the CPU for. If we can pull this off though, the average turnaround time and wait time are very low. This would be a **non-preemptive** approach, since when a process starts to run, it is left running until it finishes its CPU burst.

You can also **make SJF preemptive**. In order to do this, when a new process is added to the queue, if it has a lower CPU burst than the remaining time on the currently running process, then context switch and run the new one first. This is called **STRF** (Shortest Time Remaining First).

One problem with SJF, is that if a process has a very long CPU burst, then it may be starved of CPU time by the scheduler, since other processes, with a smaller initial burst will be allowed in first.

Priority Queue

We could have multiple queues, each with a different priority. Processes in higher priority queues could have longer time slices, or could finish all their CPU bursts before allowing other processes to start with theirs. We could still end up starving low priority processes using (particularly) the latter method, however, if we had the option to *dynamically move processes between queues*, then we could solve this problem too.

Can you work out the actual amount of CPU/IO/idle that the processes have? It's in the \LaTeX source if you want to find out!



Figure 5: If process A and B were in the high priority queue, process C was in normal priority, while D and E were in low priority, the scheduler may behave like this.

Dynamically changing which queue processes are in sounds hard, but it's not too bad. Every time a process uses up all of its timeslice, then move it down a queue, every time it finishes its CPU burst before its timeslice expires, move it up a queue. In this manner, IO bound applications will tend to execute faster, and there will be less idle time.

In earlier versions of Linux¹, there were three queues, but each used a different strategy; the highest priority queue was FIFO, the next queue used a round robin technique and the lowest priority queue used a method where the timeslice differs based on the process priority.

The low priority queue is most interesting; each process has a quantum associated with it, which is reduced by one for each clock tick it has on the CPU. The timeslice it gets the process priority plus the remaining quantum divided by two. When all of the processes in the queue run out of quanta, then the quantum of all the processes in the queue is recomputed.

Unfortunately, the old Linux way of scheduling didn't scale well, since it had an $O(n)$ runtime, so when you started up a lot of processes, the algorithm's performance would degrade. Now, Linux only uses algorithms with a $O(1)$ runtime, so that the number of processes doesn't impact on the CPU speed.

The above algorithms are targeted at desktop PC's, and 'normal' operating systems, but for applications that are a little different, such as real time systems, then different algorithms are used (maybe one that picks the process with a short deadline for computation, or the smallest deadline - completion time).

3.1.2 Context switching

A context switch occurs whenever the CPU pauses execution for one process and resumes it for another. It's sometimes an expensive operation, and depending on the processor, can take anywhere from one micro-second to one nano-second.

It is important to realise that not all context switches are equal even on the same machine. If the processor is switching between two threads of the same process, it won't have to switch in most of the information about memory, since the threads will use the same memory space. Because of this, the memory cache might also be more efficient due to locality of reference.

This isn't really a good thing, since your IO programs (such as a backup utility), may not be the ones you want to run the fastest.

It is important to remember that there are plenty of things that do need to be included in the context switch when two threads of the same process are swapping, including registers (including the PC), and the stack.

¹ If you're ahead of schedule with your revision, take a look at:

- [http://en.wikipedia.org/wiki/O\(n\)_scheduler](http://en.wikipedia.org/wiki/O(n)_scheduler)
- [http://en.wikipedia.org/wiki/O\(1\)_scheduler](http://en.wikipedia.org/wiki/O(1)_scheduler)
- http://en.wikipedia.org/wiki/Completely_Fair_Scheduler

4 Synchronisation

When two threads or processes have access to the same data, there is a danger that they could interfere with each other by using the data as a medium. This issue is especially acute when there are two threads in the same process, since they occupy the same memory space.

Race conditions can often occur as a result of shared access to data, and when synchronisation isn't (properly) used to regulate execution. A race condition is when one or more processes/threads execute in parallel, but the outcome depends on which finishes (or gets to a critical section) first. Since a thread is not guaranteed to always run at the same speed (due to clock frequencies changing, hardware being busy etc), race conditions are unpredictable, and are often hard to debug.

A critical section is a block of code that uses data that is shared between threads, consequently, only one thread should be 'in' the critical section at once.

Data inconsistency can occur when threads are executing in parallel, but they have different values for what should be the same data. This can occur when two or more threads try and update (or read from) a variable at once. Although reading and writing to a variable is often written as one instruction in most reasonably high level programming languages, further down the software stack, it could comprise of many CPU instructions. If these were to interleave, then strange things could happen.

For example, if I had one thread adding 10 to an integer (initialised to 0), and another adding 6 this could happen:

```
1 Thread A: Read variable
2 Thread B: Read variable
3 Thread A: Add 10
4 Thread B: Add 6
5 Thread A: Write variable
6 Thread B: Write variable
```

Now the value of the variable will be 6, since Thread A's write has been overwritten by Thread B.

The solution to this, is to have only one thread in a critical section at once (writing to the variable would be considered a critical section in this instance). This idea is called **Mutual Exclusion**, since only one thread can be active at a time. It is often achieved in the form of a mutex lock, which a thread will occupy when it enters a critical section, and release when it exits.

4.1 Semaphores

The famous computer scientist Dijkstra formulated a lightweight way to ensure that only a specific number of processes would enter the critical section at any one time, using only one integer variable.

If a process wanted to enter its critical section, it would wait until the Semaphore variable (S) was greater than zero. When it was, then the process would decrement S and then enter its critical section. When it leaves its critical section, it will increment S to indicate that the 'space' is free. See Listing 1 for an implementation.

There are numerous errors in Listing 1 (although it's fine as an example), it wouldn't work very well in production, can you spot them?

```
1 public abstract class Semaphore {
2     private int S = 0;
3
4     public Semaphore(int numThreads) {
5         S = numThreads;
6     }
7
8     // P = Procure the critical section
9     public synchronised void P() {
10         while(S <= 0);
11         S--;
12     }
13
14     // V = Vacate the critical section
```

```

15     public synchronised void V() {
16         S++;
17     }
18 }

```

Listing 1: A Java implementation of a Semaphore

Using an integer instead of a boolean is good, since it allows for a configurable number of processes to be working concurrently in the critical section, not just one. However, for most applications, only one thread can be in the critical section at once for the aforementioned reasons.

Although it's easy to implement a semaphore that uses a while loop to wait until another process is ready, this is very wasteful of CPU time (it's also known as a busy-wait). A better way to do it is to halt execution of the thread altogether until another semaphore space is ready. An example of this is given in Listing 2.

```

1  public abstract class Semaphore {
2      private int S = 0;
3
4      public Semaphore(int numThreads) {
5          S = numThreads;
6      }
7
8      // P = Procure the critical section
9      public synchronised void P() {
10         while(S <= 0) {
11             try { wait(); }
12             catch (InterruptedException e) { e.printStackTrace(); }
13         }
14         S--;
15     }
16
17     // V = Vacate the critical section
18     public synchronised void V() {
19         S++;
20         if(S > 0) notifyAll();
21     }
22 }

```

Listing 2: A Java implementation of a Semaphore that doesn't use busy loops

4.2 Deadlock

Deadlock is when more than one process is waiting for something that can only be provided by a process that is also in deadlock. It's a complicated thing for an operating system to even detect, never mind solve.

5 Java Threads

For some reason, the University decides to teach us about Java threads in an Operating Systems course, not in the Java course. Hmmmm...

The most basic way to run code concurrently in Java is probably to make a subclass of `java.lang.Thread`, and call `start()`, or make your class implement the `Runnable` interface.

```

1  class BitcoinMiner extends Thread {
2      public void run() {
3          // Mine many bitcoins...
4      }
5  }
6
7  ...
8

```

```
9  BitcoinMiner miner = new BitcoinMiner (...);
10 miner.start();
```

Listing 3: Extending the Thread class

```
1  class DogecoinMiner implements Runnable {
2      public void run() {
3          // very mine..
4          // such fun
5      }
6  }
7
8  ...
9
10 new Thread(new DogecoinMiner()).start();
```

Listing 4: Using the runnable interface

Threads provide three notable methods:

sleep(millis)

Sleeps the thread for an amount of milliseconds. Basically, it says *I'm done with my timeslice, and please don't give me another one for at least n milliseconds*.

wait()

Stop executing the thread until somebody calls **notify()**. *I'm done with my timeslice. Don't give me another timeslice until someone calls notify()*.

notify()

Wakes up one thread that is sleeping.

notifyAll()

Wakes up *all* the threads that are sleeping.

5.1 The synchronized keyword

In Java, every object comes with its very own mutually exclusive lock. When the object is locked, threads other than the one that locked the object cannot use it until the object is unlocked by that thread, and therefore only one thread can hold the lock at any one time.

Methods can also be made to come with a built in lock, in order to do so, you must declare them as **synchronized**. The lock must be obtained before the method will start (even though it may have been called).

Locks are automatically obtained and relinquished (kind of), so synchronization is important to get right when you're writing your methods and objects, but less so when you're using them.

You can, in fact, synchronize any block of code, by wrapping it in a **synchronized(expression)...** block. The expression is an object whose lock will be held while the block of code between the curly brackets is executed.

You can see in Listing 2, I've used a **while** loop to make the thread wait. This is because we might not always obtain the lock on the first go. Likewise, I also used **notifyAll()** in case a thread doesn't wake up properly or doesn't receive the wakeup.

6 Memory Management

Random Access Memory (RAM) is one of the most important resources for a computer, and making efficient and safe use of it is important for a well running Operating System.

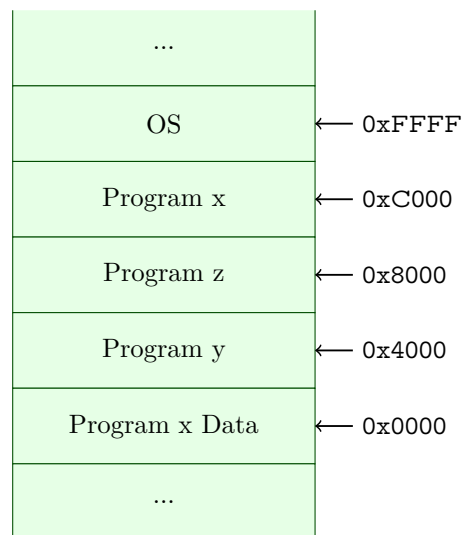
6.1 Uniprogramming

In a Von Neumann architecture RAM is used to store both programs and data. If an older OS wanted to run a program, it would load the program into RAM, move the PC to the start of the program in RAM and start executing. This method of executing programs is called **Uniprogramming**. It seems rather simple, but it has the following drawbacks:

- We don't know in advance what the base address (the start address) of the program will be. This could be problematic if the program has hardcoded addresses in it.
- If the memory (synonymous with RAM) is too small, then we can't execute the program.

6.2 Multiprogramming

Multiprogramming breaks the main memory down into fixed partitions (also called *frames*) that can each hold (sections) of a program. This allows multiple programs to be loaded (and executed) at once.



The method of assigning programs frames allows programs to be swapped in and out of memory based on when they are needed. This strategy of partitioning the main memory has its drawbacks though; if a program can write to any part of the RAM, then malicious or badly made programs could overwrite parts of other programs, which could change what the program does, or change the data it's using.

We still haven't solved the problem that some programs may have hardcoded memory addresses in them. The solution to this is to have a 'loader' which changes all the memory addresses in the program to accommodate the fact that the program isn't really running from 0x0000. It does this by adding n to the memory addresses, where n is the 'real' base address of the program.

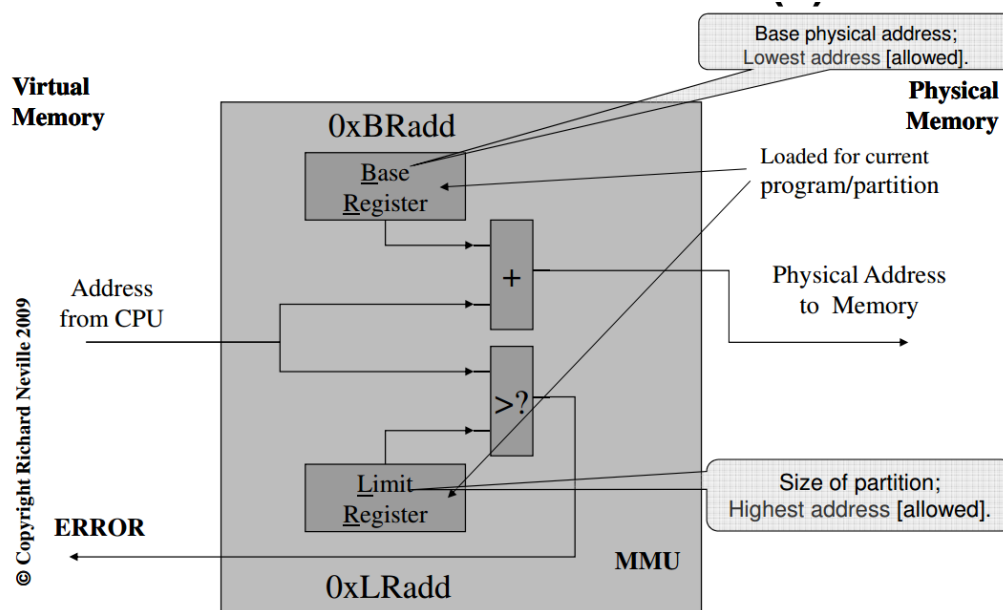
In order to stop programs from accessing memory outside their frame, a limit can be implemented when the loader edits the memory addresses, which ensures that the program cannot see or mutate data outside its frame. The limit is initialised to be the length in bytes of the program.

All of the address translation goes on in the **MMU**, which is convenient, since we can easily implement logic such as this in fast hardware, as shown in Figure 6.

Since programs that use hardcoded addresses in memory aren't *actually* using those addresses, they are referred to as '**virtual addresses**'. When a virtual address has been translated into the actual address and put into memory, it is called a '**physical address**'.

Why don't we need to worry about malicious programs editing data in frames with lower memory addresses?

The MMU (Memory Management Unit) is a piece of hardware that sits in between the processor and the main memory.



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Figure 6: How the MMU could translate addresses for execution. This diagram is taken from the course notes, ©Richard Neville 2009

6.2.1 Swapping

I've alluded to the fact that a computer could have more programs running than are currently in memory, but this section makes it explicit. When there isn't enough main memory to hold all of the running programs, some of them can be transferred onto disk for a short time until they need to be executed again.

There are strategies for keeping the 'right' programs in memory, and to mitigate the delays that the IO brings. Variable sized partitions are also used to make sure that as little space in the RAM is wasted as possible.

The memory can still get very fragmented though; programs allocate memory dynamically, and so the required size of frames could be unknown...

6.3 Virtual Memory

Virtual memory is there in order to allow the processor to address a larger range of address spaces (than there is in the physical memory), and to support the operating system in managing processes.

The MMU uses a **page table** (which is maintained by the operating system) that is used to translate between physical and logical addresses. The translation must be *very* fast since every read and write to the memory will go through it.

There are two techniques used to implement virtual memory; *paged* virtual memory and *segmented* virtual memory. Paged memory is split into fixed sized chunks called, you guessed it, pages. Segmented memory is split into chunks of a variable size. As you may have been able to tell from exploits into programs like Windows Task Manager, paging is the most commonly used method of implementing virtual memory.

6.4 Virtual paged memory

The number of pages used in the system is given by the formula:

$$\#pages = \frac{\text{Address Space (bits)}}{\text{Page size (bits)}}$$

If you want to find how many page frames (each one is the same size as a page) there are in the physical memory, then just substitute the Address Space in the above equation for the size of the physical memory.

When using paged memory, the processor gives two numbers when it wants to access a memory cell, the page number, and the offset. These numbers are usually contiguously melded together into one long number, where the first $\log_2(n)$ bits (where n is the number of pages in the virtual memory, given by 2^x , where x is the number of bits in a machine word) is the page number, and the rest of the bits is the offset. This is depicted in Figure 7.



Figure 7: The format of an address to paged memory

When the MMU gets a paged address, it follows these steps:

1. Look up the page base address in the page table (using the page number), see if the page is in the physical memory or not.
2. If the page is in memory, generate a physical address (page-base address + offset), and request that from the RAM.
3. If it's not, then abort the memory access, and the operating system will load the page into memory (this is a **page fault**).

The page table contains one row for each page. The columns will include:

Resident flag

If the page currently in virtual memory?

Used flag

Has the page been used?

Dirty flag

Has the page been written to while it's been in physical memory (so it needs to be copied back in full when it's put back onto the disk).

Physical address

The base address of the page in RAM.

Disk address

The base address of the page on backing store.

A **Page Fault** is when a page is on the backing store, but is needed by a process. If this happens:

1. The current processes is halted.
2. The OS will load the page into physical memory.
3. While the page is loading, other processes will have CPU time.
4. Once the page has been loaded into RAM, the memory reference will be retried.

6.4.1 Page replacement algorithms

After a while (or not a while if you've not got much RAM), all of the page frames will be occupied in memory. This means we need to swap one page for another in the main memory next time we have a page fault. We need an algorithm to determine which page to swap. The LRU (Last Recently Used) and FIFO (First In First Out) algorithms are good for this.

If we make bad choices about which pages to reject from memory, then our performance can seriously degrade. The following algorithms try and find the best page to swap out whenever we've got a page fault on our hands. An ideal solution is to find the page that will be used furthest in the future and swap that one out, but since we don't know in advance what memory programs will want to use (programs are dynamic), we can't do this.

First In First Out (FIFO)

Identifies the oldest page in memory and gets rid of that. This is risky, since old pages can be critical pages (such as the operating system for example, that's going to be loaded into RAM pretty early on isn't it!).

Second Chance Algorithm (SCA)

This algorithm is similar to FIFO, except that the oldest page with the fewest number of accesses since the last pass of the algorithm is removed. It's implemented by keeping a count of page accesses, that is reset on every pass of the algorithm.

Last Recently Used (LRU)

The LRU algorithm works on the principle that if a page has been used recently, then it's probably going to be used again soon. The easiest implementation is to have a timestamp for every page (32/64 bit counter) that is updated whenever the page is accessed. This can be slow when implemented in software, so it's often added as a counter in the Translation Lookaside Buffer (TLB), which is expensive in terms of hardware, but is fast.

The TLB is a cache that's used to improve the virtual address translation speed. Most PC's have one.

Not Recently Used (NRU)

The NRU algorithm is a dumb replacement for LRU. It keeps two extra bits in the page table; one called the 'referenced bit' that is set to 1 when the page is referenced and the other, called the 'modified bit' that is set to 1 when the page is written to. At fixed intervals, all of these bits will be reset to 0 (so all pages don't get 'recently used' eventually).

When a page needs to be removed from memory, the OS first looks for ones that have two 0's (they've been neither read or written to recently). Then it looks for ones that have only been written to recently (referenced pages are more important than ones that are merely modified), if it still can't find a page, then ones that have only their referenced bit set, and if not then it'll just choose one with both bits set.

These algorithms are worth it, because each memory access will probably take around $100ns$, while a read from a hard drive can take $4ms$ ($40,000\times$ longer). Even an SSD isn't that much better, taking $16\mu s$ ($160\times$ slower).

6.4.2 Pre-paging

If the pages from a process are brought back into the main memory, the algorithms above would re-load the pages one-by one as they were required. A more sensible algorithm might keep track of which pages belonged to which processes, and load them in bulk before restarting the process. A half-way solution would be to keep track of groups of pages that are commonly used together when the process is running, and load that group in bulk.

In practice, the page in memory which holds the page table may be paged out, however, when a page is running, its page tables *must be in memory*.

6.4.3 Swapping pages

If we do want to swap a page in physical memory, we first need to decide if we are to write it back to the disk. This is done by inspecting the dirty bit in the page table, and is called a **write-back**. An alternative strategy is **write-through**, where you write to RAM and disk every time, (so you don't need to copy the whole page on every swap), however, this is silly, since you spend ages waiting for disk IO.

6.4.4 Shared memory

Processes often need to share memory (e.g. unix pipes or library code). If each process only had access to its own memory, then this wouldn't be able to occur. To allow for this, page tables can be set up let virtual addresses be mapped to the same real pages. This is organised by the OS since allowing processes to edit their own page tables is an, umm, *slight* security issue.

It's a shame that we can't easily protect different areas of memory according to their function; the stack area, the data area and the code area. Unfortunately, we can't do this easily (unless we're using a Harvard Architecture or something).

6.5 Virtual segmented memory

The idea behind segmented virtual memory is similar to that of paged virtual memory, except that the segments (nearly synonymous with pages) are of a variable size. The motivation behind segmentation is about supporting the system as a whole by allowing multiple address spaces, rather than enabling us to address a larger memory space (like it is with paging), though that is a side effect of segmented memory too.

However, unlike a page, a segment is a variably sized block of virtual address space, and each segment has attributes that define how it is used, these include access rights (which users can use the segment), and usage rights (what operations can be performed on the segment, read/write/execute). Also unlike pages, programmers can actually *segments*!

Really though, in most respects, paged and segmented memory is really quite similar:

- Both segmentation and paging ensures that programs don't interfere with each other by writing to (or reading from) each others memory. This protects against malicious and badly made programs; if a program could write to the OS's memory, then the OS could be taken over by a malicious program!
- Both act like a mini address space, starting from address 0, and ending at address n , where n is the length of the page in bytes.
- The address of a segment is made up of a segment number and an offset, just like the address of a page is made up of a page number and offset.
- A table of segments/pages is kept by the OS that keeps informational about the segment/-page. The only difference for segments is that an additional field for the segment size is added.

Figure 8 shows how pages and segments are mapped from number- offset combinations onto actual address space locations.

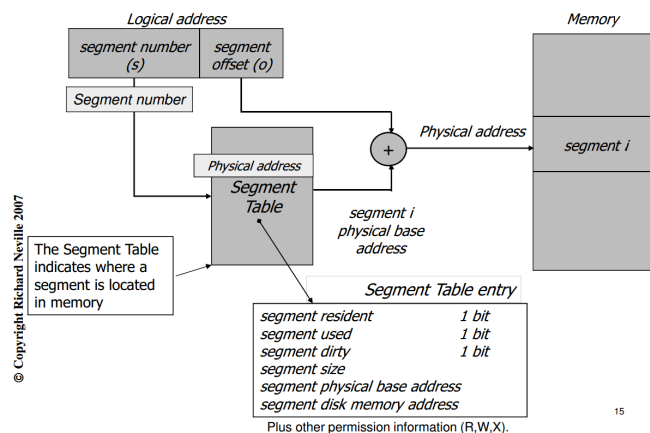


Figure 8: Richard's diagram of **virtual address mapping** for segmented memory (it's really similar for paged memory too)

A **segment fault** occurs if the OS tries to get a segment that is not in memory. If this occurs, then the page must be loaded from the hard drive into memory before it can be accessed. This isn't necessarily a *bad* thing, but it will take time to move the segment from the hard drive onto memory, and potentially free up space in memory for it first.

The position of segments in memory is harder to manage than that of pages, since segments come in different sizes. The OS needs to decide where to put segments in the memory, and what

to do with fragmented memory. Fragmentation of the memory space builds up over time as segments of different sizes are moved between main memory and the backing store. This is called **External Fragmentation**.

The OS compacts fragmented memory by shuffling segments to fill the holes, however, this is expensive since whole fragments must be copied (potentially lots of data). Of course, it's better to avoid fragmentation in the first place?

When the OS is placing segments in memory, it can use algorithms such as best fit (scan all the holes in the memory, and find the closest fit), or first fit (shove the segment in the first hold that's big enough). The first tends to produce lots of small holes (which is more efficient), while the second is faster. All of these are a lot more intensive than the LRU and FIFO algorithms we saw with paging.

The above points can make segments a lot less efficient than pages (especially since some segments can be just a few bytes, which incurs a lot of overhead for the amount of data being used), but the extra power that they give to programmers makes them a good trade off.

7 File systems

A file is a collection of related information, usually found on secondary storage. The structure of a file differs based on what it is used for, and what created it, but all files do have some stuff in common:

- Attributes, such as the size of the file, which user owns it, when was it updated, etc
- The operations you can do on it, usually reading, writing, executing.
- Some operating systems associate types with files, for example, the ubiquitous .txt file type.

When a file is opened, it is given a SFID or UFID (short for System/User File Identifier). On Unix file systems, these are called file descriptors and on Windows, they're called file handles. The kernel only deals with file descriptors, it doesn't understand file names.

SFID's have been opened by the system, UFID's have been opened by user processes

If the kernel doesn't understand file names, then how do we work with it to do IO then? The **open** system call will take a file name and return an UFID, **read** will take a UFID and return data, and **write** will take data and a UFID and write it to the file.

Different parts of the operating system will handle these processes. A file naming service may handle the **open** command, while the storage service and disk drivers will probably be used for **read** and **write**.,

Directories are converted to SFID's by the naming service. On the disk, they are pretty similar to files, but are just treated differently. File system hierarchies were first conceived, only one directory was permitted per level (so it was like a linked list of directories), or there could be one per user. Now complex trees can be created by nesting many layers of directories inside each other, each with many branches.

If you were to draw a tree of the file system, then the leaf nodes would be files, and other nodes would be directories.

7.1 File names

As we know, file names have to be converted to file identifiers before they can be used by the kernel. How then, do we turn a complex string into a single integer? All file paths can be split into one or more chunks (usually by a slash), each representing a file or directory, but there are two differing types, absolute and relative. Absolute paths have their left most 'chunk' as the root directory, while relative paths implicitly start with the working directory.

'Chunk' isn't an official term here...

Each process has a working directory that is set when the process is started (but can change), and multiple (can be zero) UFID's associated with it.

Each chunk must both identify a directory, and contain the next chunk. In both ways, the right most chunk is the file or directory name we want.

7.2 Processes and files

Each file has a size, permissions, an owner, a group that it belongs to, and most importantly, where to find it on the disk. All of this data is stored in the File Control Block (FCB).

The OS keeps a table of open files in memory, indexed by the UFID. Each entry contains the file attributes and the number of processes reading and writing to it. `open` creates a new entry in the table, while `close` removes one (and writes the attributes to the disk if they've changed). Some operating systems will vary this slightly, maybe having a separate table per process.

A buffer cache will often be utilised to improve the speed of reading and writing to files that are being frequently accessed.

7.3 How are hard disks organised?

Disks are usually structured in the following way:

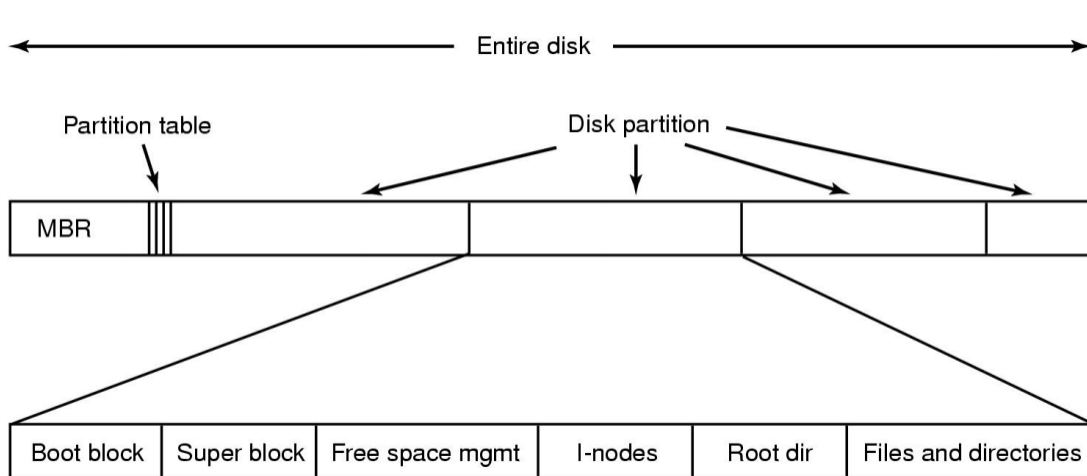


Figure 9: MOS2 fig 6.11 - How a disk may be structured

At the very start of the disk, there will be a Master Boot Record, which is where the BIOS of the computer looks first to load the Operating System, enabling the bootloader to be transferred to the main memory. Then there will be a partition table, outlining the locations, attributes and sizes of the various partitions on the disk.

After this, there will be a number (possibly one) of partitions. Each partition will have its own internal structure, as shown in the diagram.

7.3.1 The structure of files on a disk

There are a number of ways to store a file on disk:

- **Contiguous blocks**

Each file is split into a number of contiguous blocks. Here, a file at its most basic level, is just a reference to a start block, and a block count.

This is a really simple, and very fast solution, but it can lead to lots of fragmentation over time.

- **List of blocks**

We could just make the file a pointer to the first block on the disk, and then each subsequent block in the file is pointed to by the current block in a linked list format.

This is good for avoiding fragmentation, since you can stick blocks pretty much anywhere, but it's bad for access speed, since you might have to spend lots of time waiting for the hard disk to find blocks.

- **List of blocks with separate table (FAT)**

Here, we could maintain a separate table of block locations. This is good, since it means that we can store which blocks are free in the table, except it's bad since we need to keep the table in memory.

The table will consume $\frac{\text{hard-disk-size}}{\text{block-size}}$ bytes of memory, which can be quite a lot (100MB).

- **Use index nodes**

An index node (inode) is a datastructure that is used to represent a file or directory. Each inode contains the attributes of a file, as well as disk block locations for it. The inodes usually contain eleven to fifteen block numbers:

- The first eight-twelve are the first blocks of the file
- The last three are the blocks that contain block numbers for the rest of the file.

Inodes mean that you only have to have a table in RAM that is proportional to the size of the number of files you have open. You just load in the inode for every file you open. Inodes are stored in a separate disk area, shown in figure 9.

7.3.2 The structure of directories on a disk

All directories have a name, which is a fixed size on some systems, but unlimited on others. To achieve an unlimited size, you can either have a variable sized field in the block where the directory is stored, or have a fixed size pointer to a heap of strings which will contain the file name.

The directory will also have attributes, which will be stored in the entry in some file formats such as FAT, but will be stored in a block and pointed at by other formats (such as with inodes).

7.3.3 Other issues with file organisation

If you've taken the databases course (COMP23111), then you're probably familiar with the devils of concurrency. If multiple processes want to use a file, then we need some kind of policy to ensure that it doesn't get corrupted by multiple writes, or the processes don't get dodgy data when they're reading and another process is writing. To avoid this, we can let only one process write to a file, or have many processes reading a file. Some applications (such as databases) could have specific protocols if they're extra clever.

In order to increase performance, it's a good idea to use a cache in between the programs and the disk. This way, the turnaround time to access frequently used data can be lowered. The efficiency of such caches depends on the usage patterns and the cache strategies used.

One other way to improve performance is to write different blocks of a file to different disks. This requires the disks to be in a RAID (Redundant Array of Independent Disks) format, but can increase IO speeds.

Since lots of (potentially very large) datastructures can be used on the disks, there is a very real chance that they can get corrupted. Consistency checking tools such as `fsck` can be used to everything's correct. We can also make use of partitions (so that if a corruption occurs, it's probably only on part of the disk), and mirroring (have the same data written to multiple disks in case one goes boom, this also requires a RAID setup).

7.4 Unifying the virtual memory and storage

If the virtual memory manager and the file manager both copy data between the main memory and the disk, then we could potentially combine them. They have pretty synonymous operations:

File operation	VM Operation
Open	Map a file onto virtual address space
Read	Access virtual address. Page fault causes disk read
Write	Access virtual address, page rejection causes disk write
Close	Unmap pages back to disk.

This is good because files can be manipulated quicker in RAM, however it can have bad implications if there are different access patterns between main memory and secondary backing storage that the manager will have to contend with (so you have to choose an algorithm that deals with both well).