Scheduling: Introduction

By now low-level **mechanisms** of running processes (e.g., context switching) should be clear; if they are not, go back a chapter or two, and read the description of how that stuff works again. However, we have yet to understand the high-level **policies** that an OS scheduler employs. We will now do just that, presenting a series of **scheduling policies** (sometimes called **disciplines**) that various smart and hard-working people have developed over the years.

The origins of scheduling, in fact, predate computer systems; early approaches were taken from the field of operations management and applied to computers. This reality should be no surprise: assembly lines and many other human endeavors also require scheduling, and many of the same concerns exist therein, including a laser-like desire for efficiency. And thus, our problem:

THE CRUX: HOW TO DEVELOP SCHEDULING POLICY

How should we develop a basic framework for thinking about scheduling policies? What are the key assumptions? What metrics are important? What basic approaches have been used in the earliest of computer systems?

7.1 Workload Assumptions

Before getting into the range of possible policies, let us first make a number of simplifying assumptions about the processes running in the system, sometimes collectively called the **workload**. Determining the workload is a critical part of building policies, and the more you know about workload, the more fine-tuned your policy can be.

The workload assumptions we make here are mostly unrealistic, but that is alright (for now), because we will relax them as we go, and eventually develop what we will refer to as ... (*dramatic pause*) ...

a fully-operational scheduling discipline¹.

We will make the following assumptions about the processes, sometimes called **jobs**, that are running in the system:

- 1. Each job runs for the same amount of time.
- 2. All jobs arrive at the same time.
- 3. Once started, each job runs to completion.
- 4. All jobs only use the CPU (i.e., they perform no I/O)
- 5. The run-time of each job is known.

We said many of these assumptions were unrealistic, but just as some animals are more equal than others in Orwell's *Animal Farm* [O45], some assumptions are more unrealistic than others in this chapter. In particular, it might bother you that the run-time of each job is known: this would make the scheduler omniscient, which, although it would be great (probably), is not likely to happen anytime soon.

7.2 Scheduling Metrics

Beyond making workload assumptions, we also need one more thing to enable us to compare different scheduling policies: a **scheduling metric**. A metric is just something that we use to *measure* something, and there are a number of different metrics that make sense in scheduling.

For now, however, let us also simplify our life by simply having a single metric: **turnaround time**. The turnaround time of a job is defined as the time at which the job completes minus the time at which the job arrived in the system. More formally, the turnaround time $T_{turnaround}$ is:

$$T_{turnaround} = T_{completion} - T_{arrival} \tag{7.1}$$

Because we have assumed that all jobs arrive at the same time, for now $T_{arrival}=0$ and hence $T_{turnaround}=T_{completion}$. This fact will change as we relax the aforementioned assumptions.

You should note that turnaround time is a **performance** metric, which will be our primary focus this chapter. Another metric of interest is **fairness**, as measured (for example) by **Jain's Fairness Index** [J91]. Performance and fairness are often at odds in scheduling; a scheduler, for example, may optimize performance but at the cost of preventing a few jobs from running, thus decreasing fairness. This conundrum shows us that life isn't always perfect.

7.3 First In, First Out (FIFO)

The most basic algorithm we can implement is known as First In, First Out (FIFO) scheduling or sometimes First Come, First Served (FCFS).

¹Said in the same way you would say "A fully-operational Death Star."

FIFO has a number of positive properties: it is clearly simple and thus easy to implement. And, given our assumptions, it works pretty well.

Let's do a quick example together. Imagine three jobs arrive in the system, A, B, and C, at roughly the same time ($T_{arrival} = 0$). Because FIFO has to put some job first, let's assume that while they all arrived simultaneously, A arrived just a hair before B which arrived just a hair before C. Assume also that each job runs for 10 seconds. What will the **average turnaround time** be for these jobs?

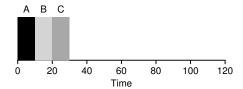


Figure 7.1: FIFO Simple Example

From Figure 7.1, you can see that A finished at 10, B at 20, and C at 30. Thus, the average turnaround time for the three jobs is simply $\frac{10+20+30}{3} = 20$. Computing turnaround time is as easy as that.

Now let's relax one of our assumptions. In particular, let's relax assumption 1, and thus no longer assume that each job runs for the same amount of time. How does FIFO perform now? What kind of workload could you construct to make FIFO perform poorly?

(think about this before reading on ... keep thinking ... got it?!)

Presumably you've figured this out by now, but just in case, let's do an example to show how jobs of different lengths can lead to trouble for FIFO scheduling. In particular, let's again assume three jobs (A, B, and C), but this time A runs for 100 seconds while B and C run for 10 each.

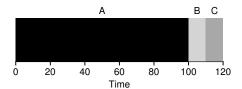


Figure 7.2: Why FIFO Is Not That Great

As you can see in Figure 7.2, Job A runs first for the full 100 seconds before B or C even get a chance to run. Thus, the average turnaround time for the system is high: a painful 110 seconds ($\frac{100+110+120}{3} = 110$).

This problem is generally referred to as the **convoy effect** [B+79], where a number of relatively-short potential consumers of a resource get queued behind a heavyweight resource consumer. This scheduling scenario might remind you of a single line at a grocery store and what you feel like when

TIP: THE PRINCIPLE OF SIF

Shortest Job First represents a general scheduling principle that can be applied to any system where the perceived turnaround time per customer (or, in our case, a job) matters. Think of any line you have waited in: if the establishment in question cares about customer satisfaction, it is likely they have taken SJF into account. For example, grocery stores commonly have a "ten-items-or-less" line to ensure that shoppers with only a few things to purchase don't get stuck behind the family preparing for some upcoming nuclear winter.

you see the person in front of you with three carts full of provisions and a checkbook out; it's going to be a while².

So what should we do? How can we develop a better algorithm to deal with our new reality of jobs that run for different amounts of time? Think about it first; then read on.

7.4 Shortest Job First (SJF)

It turns out that a very simple approach solves this problem; in fact it is an idea stolen from operations research [C54,PV56] and applied to scheduling of jobs in computer systems. This new scheduling discipline is known as **Shortest Job First (SJF)**, and the name should be easy to remember because it describes the policy quite completely: it runs the shortest job first, then the next shortest, and so on.

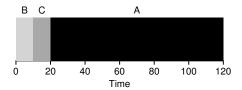


Figure 7.3: SJF Simple Example

Let's take our example above but with SJF as our scheduling policy. Figure 7.3 shows the results of running A, B, and C. Hopefully the diagram makes it clear why SJF performs much better with regards to average turnaround time. Simply by running B and C before A, SJF reduces average turnaround from 110 seconds to 50 ($\frac{10+20+120}{3}=50$), more than a factor of two improvement.

In fact, given our assumptions about jobs all arriving at the same time, we could prove that SJF is indeed an **optimal** scheduling algorithm. How-

²Recommended action in this case: either quickly switch to a different line, or take a long, deep, and relaxing breath. That's right, breathe in, breathe out. It will be OK, don't worry.

ASIDE: PREEMPTIVE SCHEDULERS

In the old days of batch computing, a number of **non-preemptive** schedulers were developed; such systems would run each job to completion before considering whether to run a new job. Virtually all modern schedulers are **preemptive**, and quite willing to stop one process from running in order to run another. This implies that the scheduler employs the mechanisms we learned about previously; in particular, the scheduler can perform a **context switch**, stopping one running process temporarily and resuming (or starting) another.

ever, you are in a systems class, not theory or operations research; no proofs are allowed.

Thus we arrive upon a good approach to scheduling with SJF, but our assumptions are still fairly unrealistic. Let's relax another. In particular, we can target assumption 2, and now assume that jobs can arrive at any time instead of all at once. What problems does this lead to?

(Another pause to think ... are you thinking? Come on, you can do it)

Here we can illustrate the problem again with an example. This time, assume A arrives at t=0 and needs to run for 100 seconds, whereas B and C arrive at t=10 and each need to run for 10 seconds. With pure SJF, we'd get the schedule seen in Figure 7.4.

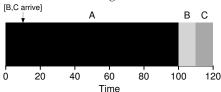


Figure 7.4: SJF With Late Arrivals From B and C

As you can see from the figure, even though B and C arrived shortly after A, they still are forced to wait until A has completed, and thus suffer the same convoy problem. Average turnaround time for these three jobs is 103.33 seconds $(\frac{100+(110-10)+(120-10)}{2})$. What can a scheduler do?

7.5 Shortest Time-to-Completion First (STCF)

To address this concern, we need to relax assumption 3 (that jobs must run to completion), so let's do that. We also need some machinery within the scheduler itself. As you might have guessed, given our previous discussion about timer interrupts and context switching, the scheduler can certainly do something else when B and C arrive: it can **preempt** job A and decide to run another job, perhaps continuing A later. SJF by our definition is a **non-preemptive** scheduler, and thus suffers from the problems described above.

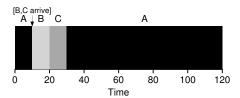


Figure 7.5: STCF Simple Example

Fortunately, there is a scheduler which does exactly that: add preemption to SJF, known as the **Shortest Time-to-Completion First** (**STCF**) or **Preemptive Shortest Job First** (**PSJF**) scheduler [CK68]. Any time a new job enters the system, the STCF scheduler determines which of the remaining jobs (including the new job) has the least time left, and schedules that one. Thus, in our example, STCF would preempt A and run B and C to completion; only when they are finished would A's remaining time be scheduled. Figure 7.5 shows an example.

The result is a much-improved average turnaround time: 50 seconds $(\frac{(120-0)+(20-10)+(30-10)}{3})$. And as before, given our new assumptions, STCF is provably optimal; given that SJF is optimal if all jobs arrive at the same time, you should probably be able to see the intuition behind the optimality of STCF.

7.6 A New Metric: Response Time

Thus, if we knew job lengths, and that jobs only used the CPU, and our only metric was turnaround time, STCF would be a great policy. In fact, for a number of early batch computing systems, these types of scheduling algorithms made some sense. However, the introduction of time-shared machines changed all that. Now users would sit at a terminal and demand interactive performance from the system as well. And thus, a new metric was born: **response time**.

Response time is defined as the time from when the job arrives in a system to the first time it is scheduled. More formally:

$$T_{response} = T_{firstrun} - T_{arrival} \tag{7.2}$$

For example, if we had the schedule above (with A arriving at time 0, and B and C at time 10), the response time of each job is as follows: 0 for job A, 0 for B, and 10 for C (average: 3.33).

As you might be thinking, STCF and related disciplines are not particularly good for response time. If three jobs arrive at the same time, for example, the third job has to wait for the previous two jobs to run *in their entirety* before being scheduled just once. While great for turnaround time, this approach is quite bad for response time and interactivity. Indeed, imagine sitting at a terminal, typing, and having to wait 10 seconds

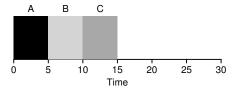


Figure 7.6: SJF Again (Bad for Response Time)

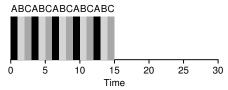


Figure 7.7: Round Robin (Good for Response Time)

to see a response from the system just because some other job got scheduled in front of yours: not too pleasant.

Thus, we are left with another problem: how can we build a scheduler that is sensitive to response time?

7.7 Round Robin

To solve this problem, we will introduce a new scheduling algorithm, classically referred to as **Round-Robin** (**RR**) scheduling [K64]. The basic idea is simple: instead of running jobs to completion, RR runs a job for a **time slice** (sometimes called a **scheduling quantum**) and then switches to the next job in the run queue. It repeatedly does so until the jobs are finished. For this reason, RR is sometimes called **time-slicing**. Note that the length of a time slice must be a multiple of the timer-interrupt period; thus if the timer interrupts every 10 milliseconds, the time slice could be 10, 20, or any other multiple of 10 ms.

To understand RR in more detail, let's look at an example. Assume three jobs A, B, and C arrive at the same time in the system, and that they each wish to run for 5 seconds. An SJF scheduler runs each job to completion before running another (Figure 7.6). In contrast, RR with a time-slice of 1 second would cycle through the jobs quickly (Figure 7.7).

The average response time of RR is: $\frac{0+1+2}{3} = 1$; for SJF, average response time is: $\frac{0+5+10}{3} = 5$.

As you can see, the length of the time slice is critical for RR. The shorter it is, the better the performance of RR under the response-time metric. However, making the time slice too short is problematic: suddenly the cost of context switching will dominate overall performance. Thus, deciding on the length of the time slice presents a trade-off to a system designer, making it long enough to **amortize** the cost of switching without making it so long that the system is no longer responsive.

TIP: AMORTIZATION CAN REDUCE COSTS

The general technique of **amortization** is commonly used in systems when there is a fixed cost to some operation. By incurring that cost less often (i.e., by performing the operation fewer times), the total cost to the system is reduced. For example, if the time slice is set to 10 ms, and the context-switch cost is 1 ms, roughly 10% of time is spent context switching and is thus wasted. If we want to *amortize* this cost, we can increase the time slice, e.g., to 100 ms. In this case, less than 1% of time is spent context switching, and thus the cost of time-slicing has been amortized.

Note that the cost of context switching does not arise solely from the OS actions of saving and restoring a few registers. When programs run, they build up a great deal of state in CPU caches, TLBs, branch predictors, and other on-chip hardware. Switching to another job causes this state to be flushed and new state relevant to the currently-running job to be brought in, which may exact a noticeable performance cost [MB91].

RR, with a reasonable time slice, is thus an excellent scheduler if response time is our only metric. But what about our old friend turnaround time? Let's look at our example above again. A, B, and C, each with running times of 5 seconds, arrive at the same time, and RR is the scheduler with a (long) 1-second time slice. We can see from the picture above that A finishes at 13, B at 14, and C at 15, for an average of 14. Pretty awful!

It is not surprising, then, that RR is indeed one of the *worst* policies if turnaround time is our metric. Intuitively, this should make sense: what RR is doing is stretching out each job as long as it can, by only running each job for a short bit before moving to the next. Because turnaround time only cares about when jobs finish, RR is nearly pessimal, even worse than simple FIFO in many cases.

More generally, any policy (such as RR) that is **fair**, i.e., that evenly divides the CPU among active processes on a small time scale, will perform poorly on metrics such as turnaround time. Indeed, this is an inherent trade-off: if you are willing to be unfair, you can run shorter jobs to completion, but at the cost of response time; if you instead value fairness, response time is lowered, but at the cost of turnaround time. This type of **trade-off** is common in systems; you can't have your cake and eat it too³.

We have developed two types of schedulers. The first type (SJF, STCF) optimizes turnaround time, but is bad for response time. The second type (RR) optimizes response time but is bad for turnaround. And we still have two assumptions which need to be relaxed: assumption 4 (that jobs do no I/O), and assumption 5 (that the run-time of each job is known). Let's tackle those assumptions next.

³A saying that confuses people, because it should be "You can't *keep* your cake and eat it too" (which is kind of obvious, no?). Amazingly, there is a wikipedia page about this saying; even more amazingly, it is kind of fun to read [W15]. As they say in Italian, you can't *Avere la botte piena e la moglie ubriaca*.

TIP: OVERLAP ENABLES HIGHER UTILIZATION

When possible, **overlap** operations to maximize the utilization of systems. Overlap is useful in many different domains, including when performing disk I/O or sending messages to remote machines; in either case, starting the operation and then switching to other work is a good idea, and improves the overall utilization and efficiency of the system.

7.8 Incorporating I/O

First we will relax assumption 4 — of course all programs perform I/O. Imagine a program that didn't take any input: it would produce the same output each time. Imagine one without output: it is the proverbial tree falling in the forest, with no one to see it; it doesn't matter that it ran.

A scheduler clearly has a decision to make when a job initiates an I/O request, because the currently-running job won't be using the CPU during the I/O; it is **blocked** waiting for I/O completion. If the I/O is sent to a hard disk drive, the process might be blocked for a few milliseconds or longer, depending on the current I/O load of the drive. Thus, the scheduler should probably schedule another job on the CPU at that time.

The scheduler also has to make a decision when the I/O completes. When that occurs, an interrupt is raised, and the OS runs and moves the process that issued the I/O from blocked back to the ready state. Of course, it could even decide to run the job at that point. How should the OS treat each job?

To understand this issue better, let us assume we have two jobs, A and B, which each need 50 ms of CPU time. However, there is one obvious difference: A runs for 10 ms and then issues an I/O request (assume here that I/Os each take 10 ms), whereas B simply uses the CPU for 50 ms and performs no I/O. The scheduler runs A first, then B after (Figure 7.8).

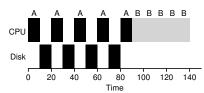


Figure 7.8: **Poor Use of Resources**

Assume we are trying to build a STCF scheduler. How should such a scheduler account for the fact that A is broken up into 5 10-ms sub-jobs, whereas B is just a single 50-ms CPU demand? Clearly, just running one job and then the other without considering how to take I/O into account makes little sense.

A common approach is to treat each 10-ms sub-job of A as an independent job. Thus, when the system starts, its choice is whether to schedule

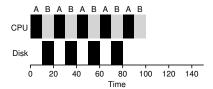


Figure 7.9: Overlap Allows Better Use of Resources

a 10-ms A or a 50-ms B. With STCF, the choice is clear: choose the shorter one, in this case A. Then, when the first sub-job of A has completed, only B is left, and it begins running. Then a new sub-job of A is submitted, and it preempts B and runs for 10 ms. Doing so allows for **overlap**, with the CPU being used by one process while waiting for the I/O of another process to complete; the system is thus better utilized (see Figure 7.9).

And thus we see how a scheduler might incorporate I/O. By treating each CPU burst as a job, the scheduler makes sure processes that are "interactive" get run frequently. While those interactive jobs are performing I/O, other CPU-intensive jobs run, thus better utilizing the processor.

7.9 No More Oracle

With a basic approach to I/O in place, we come to our final assumption: that the scheduler knows the length of each job. As we said before, this is likely the worst assumption we could make. In fact, in a general-purpose OS (like the ones we care about), the OS usually knows very little about the length of each job. Thus, how can we build an approach that behaves like SJF/STCF without such *a priori* knowledge? Further, how can we incorporate some of the ideas we have seen with the RR scheduler so that response time is also quite good?

7.10 Summary

We have introduced the basic ideas behind scheduling and developed two families of approaches. The first runs the shortest job remaining and thus optimizes turnaround time; the second alternates between all jobs and thus optimizes response time. Both are bad where the other is good, alas, an inherent trade-off common in systems. We have also seen how we might incorporate I/O into the picture, but have still not solved the problem of the fundamental inability of the OS to see into the future. Shortly, we will see how to overcome this problem, by building a scheduler that uses the recent past to predict the future. This scheduler is known as the **multi-level feedback queue**, and it is the topic of the next chapter.

References

[B+79] "The Convoy Phenomenon"

M. Blasgen, J. Gray, M. Mitoma, T. Price

ACM Operating Systems Review, 13:2, April 1979

Perhaps the first reference to convoys, which occurs in databases as well as the OS.

[C54] "Priority Assignment in Waiting Line Problems"

A. Cobham

Journal of Operations Research, 2:70, pages 70-76, 1954

The pioneering paper on using an SJF approach in scheduling the repair of machines.

[K64] "Analysis of a Time-Shared Processor"

Leonard Kleinrock

Naval Research Logistics Quarterly, 11:1, pages 59-73, March 1964

May be the first reference to the round-robin scheduling algorithm; certainly one of the first analyses of said approach to scheduling a time-shared system.

[CK68] "Computer Scheduling Methods and their Countermeasures"

Edward G. Coffman and Leonard Kleinrock

AFIPS '68 (Spring), April 1968

An excellent early introduction to and analysis of a number of basic scheduling disciplines.

[J91] "The Art of Computer Systems Performance Analysis:

Techniques for Experimental Design, Measurement, Simulation, and Modeling"

R. Jain

Interscience, New York, April 1991

The standard text on computer systems measurement. A great reference for your library, for sure.

[O45] "Animal Farm"

George Orwell

Secker and Warburg (London), 1945

A great but depressing allegorical book about power and its corruptions. Some say it is a critique of Stalin and the pre-WWII Stalin era in the U.S.S.R; we say it's a critique of pigs.

[PV56] "Machine Repair as a Priority Waiting-Line Problem"

Thomas E. Phipps Jr. and W. R. Van Voorhis

Operations Research, 4:1, pages 76–86, February 1956

Follow-on work that generalizes the SJF approach to machine repair from Cobham's original work; also postulates the utility of an STCF approach in such an environment. Specifically, "There are certain types of repair work, ... involving much dismantling and covering the floor with nuts and bolts, which certainly should not be interrupted once undertaken; in other cases it would be inadvisable to continue work on a long job if one or more short ones became available (p.81)."

[MB91] "The effect of context switches on cache performance"

Jeffrey C. Mogul and Anita Borg

ASPLOS, 1991

A nice study on how cache performance can be affected by context switching; less of an issue in today's systems where processors issue billions of instructions per second but context-switches still happen in the millisecond time range.

[W15] "You can't have your cake and eat it"

http://en.wikipedia.org/wiki/You_can't_have_your_cake_and_eat_it

Wikipedia, as of December 2015

The best part of this page is reading all the similar idioms from other languages. In Tamil, you can't "have both the moustache and drink the soup."

Homework

This program, scheduler.py, allows you to see how different schedulers perform under scheduling metrics such as response time, turnaround time, and total wait time. See the README for details.

Questions

- 1. Compute the response time and turnaround time when running three jobs of length 200 with the SJF and FIFO schedulers.
- 2. Now do the same but with jobs of different lengths: 100, 200, and 300.
- 3. Now do the same, but also with the RR scheduler and a time-slice of 1.
- 4. For what types of workloads does SJF deliver the same turnaround times as FIFO?
- 5. For what types of workloads and quantum lengths does SJF deliver the same response times as RR?
- 6. What happens to response time with SJF as job lengths increase? Can you use the simulator to demonstrate the trend?
- 7. What happens to response time with RR as quantum lengths increase? Can you write an equation that gives the worst-case response time, given *N* jobs?

Scheduling: The Multi-Level Feedback Queue

In this chapter, we'll tackle the problem of developing one of the most well-known approaches to scheduling, known as the **Multi-level Feedback Queue (MLFQ)**. The Multi-level Feedback Queue (MLFQ) scheduler was first described by Corbato et al. in 1962 [C+62] in a system known as the Compatible Time-Sharing System (CTSS), and this work, along with later work on Multics, led the ACM to award Corbato its highest honor, the **Turing Award**. The scheduler has subsequently been refined throughout the years to the implementations you will encounter in some modern systems.

The fundamental problem MLFQ tries to address is two-fold. First, it would like to optimize *turnaround time*, which, as we saw in the previous note, is done by running shorter jobs first; unfortunately, the OS doesn't generally know how long a job will run for, exactly the knowledge that algorithms like SJF (or STCF) require. Second, MLFQ would like to make a system feel responsive to interactive users (i.e., users sitting and staring at the screen, waiting for a process to finish), and thus minimize *response time*; unfortunately, algorithms like Round Robin reduce response time but are terrible for turnaround time. Thus, our problem: given that we in general do not know anything about a process, how can we build a scheduler to achieve these goals? How can the scheduler learn, as the system runs, the characteristics of the jobs it is running, and thus make better scheduling decisions?

THE CRUX:

HOW TO SCHEDULE WITHOUT PERFECT KNOWLEDGE? How can we design a scheduler that both minimizes response time for interactive jobs while also minimizing turnaround time without *a priori* knowledge of job length?

TIP: LEARN FROM HISTORY

The multi-level feedback queue is an excellent example of a system that learns from the past to predict the future. Such approaches are common in operating systems (and many other places in Computer Science, including hardware branch predictors and caching algorithms). Such approaches work when jobs have phases of behavior and are thus predictable; of course, one must be careful with such techniques, as they can easily be wrong and drive a system to make worse decisions than they would have with no knowledge at all.

8.1 MLFQ: Basic Rules

To build such a scheduler, in this chapter we will describe the basic algorithms behind a multi-level feedback queue; although the specifics of many implemented MLFQs differ [E95], most approaches are similar.

In our treatment, the MLFQ has a number of distinct **queues**, each assigned a different **priority level**. At any given time, a job that is ready to run is on a single queue. MLFQ uses priorities to decide which job should run at a given time: a job with higher priority (i.e., a job on a higher queue) is chosen to run.

Of course, more than one job may be on a given queue, and thus have the *same* priority. In this case, we will just use round-robin scheduling among those jobs.

Thus, the key to MLFQ scheduling lies in how the scheduler sets priorities. Rather than giving a fixed priority to each job, MLFQ *varies* the priority of a job based on its *observed behavior*. If, for example, a job repeatedly relinquishes the CPU while waiting for input from the keyboard, MLFQ will keep its priority high, as this is how an interactive process might behave. If, instead, a job uses the CPU intensively for long periods of time, MLFQ will reduce its priority. In this way, MLFQ will try to *learn* about processes as they run, and thus use the *history* of the job to predict its *future* behavior.

Thus, we arrive at the first two basic rules for MLFQ:

- **Rule 1:** If Priority(A) > Priority(B), A runs (B doesn't).
- **Rule 2:** If Priority(A) = Priority(B), A & B run in RR.

If we were to put forth a picture of what the queues might look like at a given instant, we might see something like the following (Figure 8.1). In the figure, two jobs (A and B) are at the highest priority level, while job C is in the middle and Job D is at the lowest priority. Given our current knowledge of how MLFQ works, the scheduler would just alternate time slices between A and B because they are the highest priority jobs in the system; poor jobs C and D would never even get to run — an outrage!

Of course, just showing a static snapshot of some queues does not really give you an idea of how MLFQ works. What we need is to under-

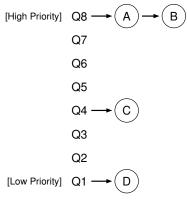


Figure 8.1: MLFQ Example

stand how job priority *changes* over time. And that, in a surprise only to those who are reading a chapter from this book for the first time, is exactly what we will do next.

8.2 Attempt #1: How To Change Priority

We now must decide how MLFQ is going to change the priority level of a job (and thus which queue it is on) over the lifetime of a job. To do this, we must keep in mind our workload: a mix of interactive jobs that are short-running (and may frequently relinquish the CPU), and some longer-running "CPU-bound" jobs that need a lot of CPU time but where response time isn't important. Here is our first attempt at a priority-adjustment algorithm:

- Rule 3: When a job enters the system, it is placed at the highest priority (the topmost queue).
- Rule 4a: If a job uses up an entire time slice while running, its priority is *reduced* (i.e., it moves down one queue).
- Rule 4b: If a job gives up the CPU before the time slice is up, it stays at the *same* priority level.

Example 1: A Single Long-Running Job

Let's look at some examples. First, we'll look at what happens when there has been a long running job in the system. Figure 8.2 shows what happens to this job over time in a three-queue scheduler.

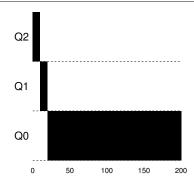


Figure 8.2: Long-running Job Over Time

As you can see in the example, the job enters at the highest priority (Q2). After a single time-slice of 10 ms, the scheduler reduces the job's priority by one, and thus the job is on Q1. After running at Q1 for a time slice, the job is finally lowered to the lowest priority in the system (Q0), where it remains. Pretty simple, no?

Example 2: Along Came A Short Job

Now let's look at a more complicated example, and hopefully see how MLFQ tries to approximate SJF. In this example, there are two jobs: A, which is a long-running CPU-intensive job, and B, which is a short-running interactive job. Assume A has been running for some time, and then B arrives. What will happen? Will MLFQ approximate SJF for B?

Figure 8.3 plots the results of this scenario. A (shown in black) is running along in the lowest-priority queue (as would any long-running CPU-intensive jobs); B (shown in gray) arrives at time T=100, and thus is

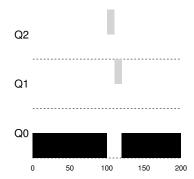


Figure 8.3: Along Came An Interactive Job

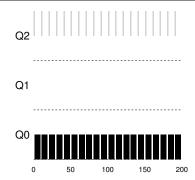


Figure 8.4: A Mixed I/O-intensive and CPU-intensive Workload

inserted into the highest queue; as its run-time is short (only 20 ms), B completes before reaching the bottom queue, in two time slices; then A resumes running (at low priority).

From this example, you can hopefully understand one of the major goals of the algorithm: because it doesn't *know* whether a job will be a short job or a long-running job, it first *assumes* it might be a short job, thus giving the job high priority. If it actually is a short job, it will run quickly and complete; if it is not a short job, it will slowly move down the queues, and thus soon prove itself to be a long-running more batch-like process. In this manner, MLFQ approximates SJF.

Example 3: What About I/O?

Let's now look at an example with some I/O. As Rule 4b states above, if a process gives up the processor before using up its time slice, we keep it at the same priority level. The intent of this rule is simple: if an interactive job, for example, is doing a lot of I/O (say by waiting for user input from the keyboard or mouse), it will relinquish the CPU before its time slice is complete; in such case, we don't wish to penalize the job and thus simply keep it at the same level.

Figure 8.4 shows an example of how this works, with an interactive job B (shown in gray) that needs the CPU only for 1 ms before performing an I/O competing for the CPU with a long-running batch job A (shown in black). The MLFQ approach keeps B at the highest priority because B keeps releasing the CPU; if B is an interactive job, MLFQ further achieves its goal of running interactive jobs quickly.

Problems With Our Current MLFQ

We thus have a basic MLFQ. It seems to do a fairly good job, sharing the CPU fairly between long-running jobs, and letting short or I/O-intensive interactive jobs run quickly. Unfortunately, the approach we have developed thus far contains serious flaws. Can you think of any?

(This is where you pause and think as deviously as you can)

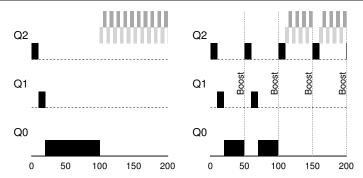


Figure 8.5: Without (Left) and With (Right) Priority Boost

First, there is the problem of **starvation**: if there are "too many" interactive jobs in the system, they will combine to consume *all* CPU time, and thus long-running jobs will *never* receive any CPU time (they **starve**). We'd like to make some progress on these jobs even in this scenario.

Second, a smart user could rewrite their program to game the scheduler. Gaming the scheduler generally refers to the idea of doing something sneaky to trick the scheduler into giving you more than your fair share of the resource. The algorithm we have described is susceptible to the following attack: before the time slice is over, issue an I/O operation (to some file you don't care about) and thus relinquish the CPU; doing so allows you to remain in the same queue, and thus gain a higher percentage of CPU time. When done right (e.g., by running for 99% of a time slice before relinquishing the CPU), a job could nearly monopolize the CPU.

Finally, a program may *change its behavior* over time; what was CPU-bound may transition to a phase of interactivity. With our current approach, such a job would be out of luck and not be treated like the other interactive jobs in the system.

8.3 Attempt #2: The Priority Boost

Let's try to change the rules and see if we can avoid the problem of starvation. What could we do in order to guarantee that CPU-bound jobs will make some progress (even if it is not much?).

The simple idea here is to periodically **boost** the priority of all the jobs in system. There are many ways to achieve this, but let's just do something simple: throw them all in the topmost queue; hence, a new rule:

• **Rule 5:** After some time period *S*, move all the jobs in the system to the topmost queue.

Our new rule solves two problems at once. First, processes are guaranteed not to starve: by sitting in the top queue, a job will share the CPU

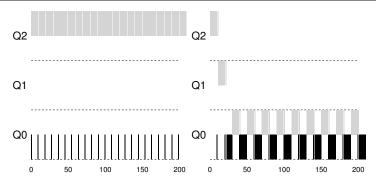


Figure 8.6: Without (Left) and With (Right) Gaming Tolerance

with other high-priority jobs in a round-robin fashion, and thus eventually receive service. Second, if a CPU-bound job has become interactive, the scheduler treats it properly once it has received the priority boost.

Let's see an example. In this scenario, we just show the behavior of a long-running job when competing for the CPU with two short-running interactive jobs. Two graphs are shown in Figure 8.5 (page 6). On the left, there is no priority boost, and thus the long-running job gets starved once the two short jobs arrive; on the right, there is a priority boost every 50 ms (which is likely too small of a value, but used here for the example), and thus we at least guarantee that the long-running job will make some progress, getting boosted to the highest priority every 50 ms and thus getting to run periodically.

Of course, the addition of the time period S leads to the obvious question: what should S be set to? John Ousterhout, a well-regarded systems researcher [O11], used to call such values in systems **voo-doo constants**, because they seemed to require some form of black magic to set them correctly. Unfortunately, S has that flavor. If it is set too high, long-running jobs could starve; too low, and interactive jobs may not get a proper share of the CPU.

8.4 Attempt #3: Better Accounting

We now have one more problem to solve: how to prevent gaming of our scheduler? The real culprit here, as you might have guessed, are Rules 4a and 4b, which let a job retain its priority by relinquishing the CPU before the time slice expires. So what should we do?

The solution here is to perform better **accounting** of CPU time at each level of the MLFQ. Instead of forgetting how much of a time slice a process used at a given level, the scheduler should keep track; once a process has used its allotment, it is demoted to the next priority queue. Whether

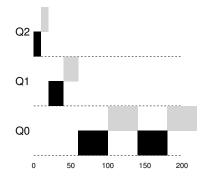


Figure 8.7: Lower Priority, Longer Quanta

it uses the time slice in one long burst or many small ones does not matter. We thus rewrite Rules 4a and 4b to the following single rule:

• Rule 4: Once a job uses up its time allotment at a given level (regardless of how many times it has given up the CPU), its priority is reduced (i.e., it moves down one queue).

Let's look at an example. Figure 8.6 (page 7) shows what happens when a workload tries to game the scheduler with the old Rules 4a and 4b (on the left) as well the new anti-gaming Rule 4. Without any protection from gaming, a process can issue an I/O just before a time slice ends and thus dominate CPU time. With such protections in place, regardless of the I/O behavior of the process, it slowly moves down the queues, and thus cannot gain an unfair share of the CPU.

8.5 Tuning MLFQ And Other Issues

A few other issues arise with MLFQ scheduling. One big question is how to **parameterize** such a scheduler. For example, how many queues should there be? How big should the time slice be per queue? How often should priority be boosted in order to avoid starvation and account for changes in behavior? There are no easy answers to these questions, and thus only some experience with workloads and subsequent tuning of the scheduler will lead to a satisfactory balance.

For example, most MLFQ variants allow for varying time-slice length across different queues. The high-priority queues are usually given short time slices; they are comprised of interactive jobs, after all, and thus quickly alternating between them makes sense (e.g., 10 or fewer milliseconds). The low-priority queues, in contrast, contain long-running jobs that are CPU-bound; hence, longer time slices work well (e.g., 100s of ms). Figure 8.7 shows an example in which two long-running jobs run for 10 ms at the highest queue, 20 in the middle, and 40 at the lowest.

we call it Ousterhout's Law.

TIP: AVOID VOO-DOO CONSTANTS (OUSTERHOUT'S LAW) Avoiding voo-doo constants is a good idea whenever possible. Unfortunately, as in the example above, it is often difficult. One could try to make the system learn a good value, but that too is not straightforward. The frequent result: a configuration file filled with default parameter values that a seasoned administrator can tweak when something isn't quite working correctly. As you can imagine, these are often left unmodified, and thus we are left to hope that the defaults work well in the field. This tip brought to you by our old OS professor, John Ousterhout, and hence

The Solaris MLFQ implementation — the Time-Sharing scheduling class, or TS — is particularly easy to configure; it provides a set of tables that determine exactly how the priority of a process is altered throughout its lifetime, how long each time slice is, and how often to boost the priority of a job [AD00]; an administrator can muck with this table in order to make the scheduler behave in different ways. Default values for the table are 60 queues, with slowly increasing time-slice lengths from 20 milliseconds (highest priority) to a few hundred milliseconds (lowest), and priorities boosted around every 1 second or so.

Other MLFQ schedulers don't use a table or the exact rules described in this chapter; rather they adjust priorities using mathematical formulae. For example, the FreeBSD scheduler (version 4.3) uses a formula to calculate the current priority level of a job, basing it on how much CPU the process has used [LM+89]; in addition, usage is decayed over time, providing the desired priority boost in a different manner than described herein. See Epema's paper for an excellent overview of such **decay-usage** algorithms and their properties [E95].

Finally, many schedulers have a few other features that you might encounter. For example, some schedulers reserve the highest priority levels for operating system work; thus typical user jobs can never obtain the highest levels of priority in the system. Some systems also allow some user **advice** to help set priorities; for example, by using the command-line utility nice you can increase or decrease the priority of a job (somewhat) and thus increase or decrease its chances of running at any given time. See the man page for more.

8.6 MLFQ: Summary

We have described a scheduling approach known as the Multi-Level Feedback Queue (MLFQ). Hopefully you can now see why it is called that: it has *multiple levels* of queues, and uses *feedback* to determine the priority of a given job. History is its guide: pay attention to how jobs behave over time and treat them accordingly.

TIP: USE ADVICE WHERE POSSIBLE

As the operating system rarely knows what is best for each and every process of the system, it is often useful to provide interfaces to allow users or administrators to provide some **hints** to the OS. We often call such hints **advice**, as the OS need not necessarily pay attention to it, but rather might take the advice into account in order to make a better decision. Such hints are useful in many parts of the OS, including the scheduler (e.g., with nice), memory manager (e.g., madvise), and file system (e.g., informed prefetching and caching [P+95]).

The refined set of MLFQ rules, spread throughout the chapter, are reproduced here for your viewing pleasure:

- **Rule 1:** If Priority(A) > Priority(B), A runs (B doesn't).
- **Rule 2:** If Priority(A) = Priority(B), A & B run in RR.
- Rule 3: When a job enters the system, it is placed at the highest priority (the topmost queue).
- Rule 4: Once a job uses up its time allotment at a given level (regardless of how many times it has given up the CPU), its priority is reduced (i.e., it moves down one queue).
- **Rule 5:** After some time period \hat{S} , move all the jobs in the system to the topmost queue.

MLFQ is interesting for the following reason: instead of demanding *a priori* knowledge of the nature of a job, it observes the execution of a job and prioritizes it accordingly. In this way, it manages to achieve the best of both worlds: it can deliver excellent overall performance (similar to SJF/STCF) for short-running interactive jobs, and is fair and makes progress for long-running CPU-intensive workloads. For this reason, many systems, including BSD UNIX derivatives [LM+89, B86], Solaris [M06], and Windows NT and subsequent Windows operating systems [CS97] use a form of MLFQ as their base scheduler.

THE MULTI-LEVEL FEEDBACK QUEUE

References

[AD00] "Multilevel Feedback Queue Scheduling in Solaris"

Andrea Arpaci-Dusseau

Available: http://www.cs.wisc.edu/~remzi/solaris-notes.pdf

A great short set of notes by one of the authors on the details of the Solaris scheduler. OK, we are probably biased in this description, but the notes are pretty darn good.

[B86] "The Design of the UNIX Operating System"

M.J. Bach

Prentice-Hall, 1986

One of the classic old books on how a real UNIX operating system is built; a definite must-read for kernel hackers.

[C+62] "An Experimental Time-Sharing System"

F. J. Corbato, M. M. Daggett, R. C. Daley

IFIPS 1962

A bit hard to read, but the source of many of the first ideas in multi-level feedback scheduling. Much of this later went into Multics, which one could argue was the most influential operating system of all time.

[CS97] "Inside Windows NT"

Helen Custer and David A. Solomon

Microsoft Press, 1997

The NT book, if you want to learn about something other than UNIX. Of course, why would you? OK, we're kidding; you might actually work for Microsoft some day you know.

[E95] "An Analysis of Decay-Usage Scheduling in Multiprocessors"

D.H.J. Epema

SIGMETRICS '95

A nice paper on the state of the art of scheduling back in the mid 1990s, including a good overview of the basic approach behind decay-usage schedulers.

[LM+89] "The Design and Implementation of the 4.3BSD UNIX Operating System"

S.J. Leffler, M.K. McKusick, M.J. Karels, J.S. Quarterman

Addison-Wesley, 1989

Another OS classic, written by four of the main people behind BSD. The later versions of this book, while more up to date, don't quite match the beauty of this one.

[M06] "Solaris Internals: Solaris 10 and OpenSolaris Kernel Architecture"

Richard McDougall

Prentice-Hall, 2006

A good book about Solaris and how it works.

[O11] "John Ousterhout's Home Page"

John Ousterhout

Available: http://www.stanford.edu/~ouster/

The home page of the famous Professor Ousterhout. The two co-authors of this book had the pleasure of taking graduate operating systems from Ousterhout while in graduate school; indeed, this is where the two co-authors got to know each other, eventually leading to marriage, kids, and even this book. Thus, you really can blame Ousterhout for this entire mess you're in.

[P+95] "Informed Prefetching and Caching"

R.H. Patterson, G.A. Gibson, E. Ginting, D. Stodolsky, J. Zelenka

SOSP '95

A fun paper about some very cool ideas in file systems, including how applications can give the OS advice about what files it is accessing and how it plans to access them.

Homework

This program, mlfq.py, allows you to see how the MLFQ scheduler presented in this chapter behaves. See the README for details.

Questions

- 1. Run a few randomly-generated problems with just two jobs and two queues; compute the MLFQ execution trace for each. Make your life easier by limiting the length of each job and turning off I/Os.
- 2. How would you run the scheduler to reproduce each of the examples in the chapter?
- 3. How would you configure the scheduler parameters to behave just like a round-robin scheduler?
- 4. Craft a workload with two jobs and scheduler parameters so that one job takes advantage of the older Rules 4a and 4b (turned on with the -S flag) to game the scheduler and obtain 99% of the CPU over a particular time interval.
- 5. Given a system with a quantum length of 10 ms in its highest queue, how often would you have to boost jobs back to the highest priority level (with the -B flag) in order to guarantee that a single long-running (and potentially-starving) job gets at least 5% of the CPU?
- 6. One question that arises in scheduling is which end of a queue to add a job that just finished I/O; the −I flag changes this behavior for this scheduling simulator. Play around with some workloads and see if you can see the effect of this flag.

Real Time Scheduling

Introduction

Priority based scheduling enables us to give better service to certain processes. In our discussion of multi-queue scheduling, priority was adjusted based on whether a task was more interactive or compute intensive. But most schedulers enable us to give any process any desired priority. Isn't that good enough?

Priority scheduling is inherently a *best effort* approach. If our task is competing with other high priority tasks, it may not get as much time as it requires. Sometimes best effort isn't good enough:

- During reentry, the space shuttle is aerodynamically unstable. It is not actually being kept under control by the quick reflexes of the well-trained pilots, but rather by guidance computers that are collecting attitude and acceleration input and adjusting numerous spoilers hundreds of times per second.
- Scientific and military satellites may receive precious and irreplaceable sensor data at extremely high speeds. If it takes us too long to receive, process, and store one data frame, the next data frame may be lost.
- More mundanely, but also important, many manufacturing processes are run by computers nowadays. An assembly line needs to move at a particular speed, with each step being performed at a particular time. Performing the action too late results in a flawed or useless product.
- Even more commonly, playing media, like video or audio, has real time requirements. Sound must be produced at a certain rate and frames must be displayed frequently enough or the media becomes uncomfortable to deal with.

There are many computer controlled applications where delays in critical processing can have undesirable, or even disastrous consequences.

What are Real-Time Systems

A real-time system is one whose correctness depends on timing as well as functionality.

When we discussed more traditional scheduling algorithms, the metrics we looked at were *turn-around time* (or throughput), *fairness*, and mean *response time*. But real-time systems have very different requirements, characterized by different metrics:

- timeliness ... how closely does it meet its timing requirements (e.g. ms/day of accumulated tardiness)
- predictability ... how much deviation is there in delivered timeliness

And we introduce a few new concepts:

- feasibility ... whether or not it is possible to meet the requirements for a particular task set
- *hard real-time* ... there are strong requirements that specified tasks be run a specified intervals (or within a specified response time). Failure to meet this requirement (perhaps by as little as a fraction of a microsecond) may result in system failure.
- *soft real-time* ... we may want to provide very good (e.g. microseconds) response time, the only consequences of missing a deadline are degraded performance or recoverable failures.

It sounds like real-time scheduling is more critical and difficult than traditional time-sharing, and in many ways it is. But real-time systems may have a few characteristics that make scheduling easier:

• We may actually know how long each task will take to run. This enables much more intelligent scheduling.



- *Starvation* (of low priority tasks) may be acceptable. The space shuttle absolutely must sense attitude and acceleration and adjust spolier positions once per millisecond. But it probably doesn't matter if we update the navigational display once per millisecond or once every ten seconds. Telemetry transmission is probably somewhere in-between. Understanding the relative criticality of each task gives us the freedom to intelligently shed less critical work in times of high demand.
- The work-load may be relatively fixed. Normally high utilization implies long queuing delays, as bursty traffic creates long lines. But if the incoming traffic rate is relatively constant, it is possible to simultaneously achieve high utilization and good response time.

Real-Time Scheduling Algorithms

In the simplest real-time systems, where the tasks and their execution times are all known, there might not even be a scheduler. One task might simply call (or yield to) the next. This model makes a great deal of sense in a system where the tasks form a producer/consumer pipeline (e.g. MPEG frame receipt, protocol decoding, image decompression, display).

In more complex real-time system, with a larger (but still fixed) number of tasks that do not function in a strictly pipeline fashion, it may be possible to do *static* scheduling. Based on the list of tasks to be run, and the expected completion time for each, we can define (at design or build time) a fixed schedule that will ensure timely execution of all tasks.

But for many real-time systems, the work-load changes from moment to moment, based on external events. These require *dynamic* scheduling. For *dynamic* scheduling algorithms, there are two key questions:

- 1. how they choose the next (ready) task to run
 - shortest job first
 - static priority ... highest priority ready task
 - soonest start-time deadline first (ASAP)
 - soonest completion-time deadline first (slack time)
- 2. how they handle overload (infeasible requirements)
 - best effort
 - o periodicity adjustments ... run lower priority tasks less often.
 - work shedding ... stop running lower priority tasks entirely.

Preemption may also be a different issue in real-time systems. In ordinary time-sharing, preemption is a means of improving mean response time by breaking up the execution of long-running, compute-intensive tasks. A second advantage of preemptive scheduling, particularly important in a general purpose timesharing system, is that it prevents a buggy (infinite loop) program from taking over the CPU. The trade-off, between improved response time and increased overhead (for the added context switches), almost always favors preemptive scheduling. This may not be true for real-time systems:

- preempting a running task will almost surely cause it to miss its completion deadline.
- since we so often know what the expected execution time for a task will be, we can schedule accordingly and should have little need for preemption.
- embedded and real-time systems run fewer and simpler tasks than general purpose time systems, and the code is often much better tested ... so infinite loop bugs are extremely rare.

For the least demanding real time tasks, a sufficiently lightly loaded system might be reasonably successful in meeting its deadlines. However, this is achieved simply because the frequency at which the task is run happens to be high enough to meet its real time requirements, not because the scheduler is aware of such requirements. A lightly loaded machine running a traditional scheduler can often display a video to a user's satisfaction, not because the scheduler "knows" that a frame must be rendered by a certain deadline, but simply because the machine has enough cycles and a low enough work load to render the frame before the deadline has arrived.

Real-Time and Linux

Linux was not designed to be an embedded or real-time operating system, but many tasks that were once-considered embedded applications now require the capabilities (e.g. file systems, network protocols) of a general purpose operating system. As these requirements have increased and processors have gotten faster, increasingly many embedded and real-time applications have moved to Linux.

To support these applications Linux now supports a real-time scheduler, which can be enabled with sched setscheduler(2), and is described in this Linux Journal Article. This real-time scheduler does not provide quite the same level of response-time guarantees that more traditional Real-Time-OSs do, but they are adequate for many soft real-time applications.

What about Windows? Conventional wisdom states that Windows is not well suited for real time needs, offering no native real time scheduler and too many ways in which desired real time deadlines might be missed. Windows favors general purpose throughput over meeting deadlines, as a rule. With sufficiently low load, a Windows system may, nonetheless, provide fast enough service for some soft real time requirements, such as playing music or video. One should be careful in relying on Windows for critical real time operations, however, as it is not designed for that purpose.