Introduction To Embedded Systems Development

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1 Introduction

This book will teach you embedded system design and development, using the LPC1769, an ARM Cortex microcontroller. When compared to normal CPU the LPC is a simple device, with only one processor core and a miniscule amount of memory, yet it still manages to be useful. In fact this simplicity makes it a wonderful learning tool, allowing us to forgo kernel and driver interfaces and work at the lowest level possible, without becoming too complicated to dive headfirst into. We'll exploit the simplicity to teach you about how processors are structured, how to deal with many common protocols and tools, and even how some fundamental parts of an operating system work.

The LPC itself has a number of modules, which encapsulate various features of the processor. We'll be working our way through those modules, explaining why you would use them, how they work, and how to use them. With each module we learn about, we'll present exercises and projects that will help you cement that knowledge, and give you a practical examples of how these devices can be used.

While we'll often be working with electronics, no initial knowledge is required, and we'll give you the resources to learn what you need to know as you go along.

1.1 Resources

This textbook will mainly work to help you build a conceptual framework around these topics, there are other resources that will give you the fine detail.

1.1.1 The LPC 17XX User Manual

The User Manual [\(available here](http://www.nxp.com/documents/user_manual/UM10360.pdf)^{[1](#page-2-2)}) will be the main document this textbook builds on. It contains detailed information on all of the available features of the LPC, and how you can use them.

Though this textbook is meant to give you all of the background that the manual lacks, it is in no way a substitute. There will be a lot we cannot actually go over and to get the most out of this course, after every chapter you read in this textbook, you should read the corresponding chapters in the manual.

A quick warning, this manual is very large, and you should not print it unless it's absolutely necessary.

1.1.2 The LPC1769 Schematic

The schematic [\(available here](http://www.cs.umd.edu/class/fall2012/cmsc498a/manuals/lpcxpresso_lpc1769_schematic.pdf) 2) describes how the pins on the development boards correspond to the pins mentioned in the manual. You'll need to cross reference this schematic with Chapter 8 of the manual every time you need to figure out the physical location of any one particular pin.

1.1.3 The LPC176X Data Sheet

The data sheet [\(available here](http://www.nxp.com/documents/data_sheet/LPC1769_68_67_66_65_64_63.pdf)^{[3](#page-2-4)}) is the last reference document you should keep handy. It contains a lot of information about the limitations of the LPC, the tolerances of various features, and the full set of features the LPC has.

¹http://www.nxp.com/documents/user_manual/UM10360.pdf

²http://www.cs.umd.edu/class/fall2012/cmsc498a/manuals/lpcxpresso_lpc1769_schematic.pdf

 3 http://www.nxp.com/documents/data_sheet/LPC1769_68_67_66_65_64_63.pdf

2 GPIO

2.1 Prequisites

Read the following tutorials to get you up to speed with the electronics:

- [How to read a schematic](https://learn.sparkfun.com/tutorials/how-to-read-a-schematic)
- [How to use a breadboard](https://learn.sparkfun.com/tutorials/how-to-use-a-breadboard/introduction)
- [Understanding LEDs](https://learn.sparkfun.com/tutorials/light-emitting-diodes-leds)
- [Using Pull Up Resistors](https://learn.sparkfun.com/tutorials/pull-up-resistors/introduction)

All these tutorials are Arduino centric, but you can extrapolate to what you need to do for your LPC.

2.2 What is GPIO?

We'll start with the most basic peripheral on the LPC, General Purpose Input Output. GPIO is what lets your microcontroller be something more than a weak auxiliary processor. With it you can interact with the environment, connecting up other devices and turning your microcontroller into something useful.

GPIO has two fundamental operating modes, input and output. Input lets you read the voltage on a pin, to see whether it's held low $(0y)$ or high $(3y)$ and deal with that information programmatically. Output lets you set the voltage on a pin, again either high or low. Every pin on the LPC can be used as a GPIO pin, and can be independently set to act as an input or output.

In this chapter we will show how to complete 2 tasks, reading the state of a button, and making an LED blink. Using what you learn from that, you'll be able to start building more complex devices, and even implementing simple communications protocols.

2.3 GPIO Output

GPIO output is a versatile and powerful tool, especially given that it takes little effort to use and control it. Once you've chosen a pin, the process to use it is straightforward:

- tell the LPC that the pin should be used as an output
- tell the LPC whether the pin should be held low or high.

The interesting part is how exactly you can give the LPC instructions, and what it does in order to carry them out.

2.3.1 Memory Mapped Registers

In order to talk to the GPIO controller, or any other peripheral, we have we have to use *memory mapped registers*. In most computers these low level interfaces are hidden by the kernel, and often only higher level interfaces are available to developers. The LPC however doesn't have a kernel or drivers hiding these interfaces from you. This gives you the ability to work with them directly, without having to deal with the virtual device abstraction a modern OS would impose.

When we try to read or write to a normal chunk of memory, the address and instruction are sent to the memory controller. The memory controller then either retrieves data from memory and places it into a register, or takes data from a register and writes it to somewhere in the memory. However, there are a number of privileged address, and when you try to read from or write to these a different pathway is taken. Here when the memory controller gets the instruction it notices the address is special, and instead of going to the memory module, it'll forward the request to a register that's located in the relevant peripheral.

These registers all have different functions, each of which is detailed in the manual along with the register's memory address. The really important thing to notice is that you're not dealing with a normal piece of memory, and these *memory mapped* registers can act very differently.

Unlike static memory where you can read or write pretty much anywhere, there are a number of these memory mapped registers which you can only read from. Trying to write to any of these will trap your system in a hard fault. Neither can you rely on the assumption that reads are nondestructive. There are some registers which are connected to FIFOs and other structures, and reading from them is the same as popping from that queue, and the next time you read from the same address, you'll get a different value. Even the usual guarantee that a write operation is idempotent is lost. There are registers where a write operation will trigger some change in the peripheral, making the LPC turn an LED on, or send out a signal.

2.3.2 Blinking Lights

So let's start with something simple: Blinking Lights. Connect up an LED to pin P0[9] and ground, making sure to place the proper current limiting resistor in series with it. To actually turn on the LED you have to first tell the LPC that the pin is to be used for output, and then set the state to be on. If you look in the manual you'll see that P0[9]'s direction is controlled by the 9th bit in a register located at 0x2009C000, and that setting it to 1 makes it an output pin. [4](#page-4-0)

((uint32_t *) 0x2009C000) |= (1 << 9); *// Set P0[9] to output*

Then there's another register at 0x2009C014 which controls the state of the pin, so we can turn turn the light on and off by manipulating its 9th bit.

```
((uint32_t *) 0x2009C014) |= (1 << 9); // Turn On
((uint32_t *) 0x2009C014) &= ~(1 << 9); // Turn Off
```
So now you should be able to make the light blink, or by varying the amount of time on and off, let it glow with varying levels of brightness.

2.3.3 An Easier Way

Of course writing out the memory address every time you wish to change a register isn't easy, or readable. You could replace it with a preprocessor macro, but writing those macros would be painful and tedious. Until you notice that the memory locations for these registers are structured, with registers performing related tasks placed close together. In fact, the addresses are chosen so that they can easily map to structs. Finding the base address of a particular block of registers, and defining a suitable structure, will give you easy to use pointers to all the registers in that block. There already exists a library that defines these structs and calculates the proper base addresses.

CMSIS [5](#page-4-1) is a library written by engineers at ARM, and it sets up all these memory addresses as human readable macros for you. To see how it works let's look at the setup for the DAC (Digital to Analog Converter).

⁴See manual page 107

⁵CMSIS: Cortex Microcontroller Software Interface Standard

The DAC is a device which can take a digital value, and turns it into an analog output, it's controlled with only 3 registers, the functions of which we'll look at in a later chapter. In LPC17xx.h you'll find a long list of struct definitions and a series of raw memory addresses. The addresses point to the chunks of memory assigned to each peripheral and the structs show the layout of each of those chunks of memory.

```
/*----- Digital-to-Analog Converter (DAC) ------*/
typedef struct
{
    __IO uint32_t DACR;
    __IO uint32_t DACCTRL;
    __IO uint16_t DACCNTVAL;
} LPC_DAC_TypeDef;
    ...
#define LPC_APB1_BASE (0x40080000UL)
    ...
#define LPC_DAC_BASE (LPC_APB1_BASE + 0x0C000)
    ...
#define LPC_DAC ((LPC_DAC_TypeDef *) LPC_DAC_BASE )
```
The DAC is an APB1^{[6](#page-5-0)} peripheral and so LPC_DAC_BASE is the start of the memory mapped to DAC registers. LPC_DAC_Typedef is a struct that's set up so that when it's aligned to that base address, each of the struct's fields will align with a particular register in the DAC memory space. This whole setup means that you can write to the DAC Control Registers without using a raw memory location.

LPC_DAC->DACCTRL = $/*$ stuff $*/$;

If you look back at the code we wrote for LED manipulation, and refactor it, you'll get something much easier to work with.

```
LPC_GPIO0->FIODIR |= (1 << 9); // Set P0[9] to write
LPC_GPIO0->FIOPIN |= (1 << 9); // Turn LED on
LPC_GPIO0->FIOPIN &= ~(1 << 9); // Turn LED off
```
Having a layer of macros like this also makes it easier to port your code to another platform, since you'll have to only change the macro definitions rather than all the pieces of code which use some registers.

2.3.4 More Registers

If you look closely there's 4 GPIO ports, each controlling up to 32 pins, and each of those blocks has 5 registers. Strictly speaking you only need the FIODIR (set pin direction) and FIOPIN (set or read pin state) registers to control each pin, but there are three others, which allow you to perform operations much faster.

FIOSET and FIOCLR are the two fast output control registers.[7](#page-5-1) Writing a 1 to a bit in FIOSET will enable the corresponding pin, and writing to FIOCLR will disable the pin.

⁶APB stands for Applied Peripheral Bus, there are two in the LPC and some peripherals are connected to each.

⁷The fast in their moniker refers to the fact that using them takes fewer operations than using FIOPIN.

LPC_GPIO0->FIOSET = 1 << 9; *// Turn LED On* LPC_GPIO0->FIOCLR = 1 << 9; *// Turn LED Off*

FIOSET is a good example of how the usual guarantees of memory structure are lost when working with memory mapped data. Writing a 1 to FIOSET will set the corresponding bit in FIOPIN to 1, while writing a 0 will do nothing. In effect 'FIOSET = \dots ' is an alias for 'FIOPIN $| = \dots$ '. However reading from FIOSET is a completely different action, it will return the value from the output state register, a register which stores the current output value for all the pins, regardless of whether they are current being used as such.

Writes to FIOCLR can be similarly thought of as an alias, in this case from 'FIOPIN $k = \sim \ldots$ ' to 'FIOCLR = ...'. But reading from FIOCLR is undefined, there is simply nothing that operation can look at when pointed at FIOCLR.

This idea of memory mapped registers being aliases for more complex commands hints at why these registers are called the "*fast* output control registers". Trying to change the value of a single pin with FIOPIN requires at least 3 operations operations, a read , a bitwise logic operation, and a write. To make the same change using the fast registers requires only a write operation, the rest of the stuff is done in hardware, which is much faster.

FIOMASK is, in effect, a filter for FIOPIN,FIOSET, and FIOCLR. If a bit in FIOMASK is a 1, then none of those registers can cause any change in that pin's state. This means that you can change a subset of the bits very quickly, without having to perform a masking operation every time. By default all of FIOMASK's bits are set to 0, meaning that the other control registers can operate over all bits.

2.4 GPIO Input

Reading a pin uses the same registers we've already used, once a pin's mode is set to input in FIODIR, the corresponding bit in FIOPIN holds the currently read value. In addition to simply reading the registers to figure out the voltage on a pin, we've also got access to interrupts, which will notify your program when the state of a pin changes, while allowing you to do something else in the meantime.

2.4.1 Basics

Reading the current state of a pin in the middle of your code is simple, we take the same two registers as before FIODIR and FIOPIN, and use them slightly differently.

```
LPC_GPIO0->FIODIR &= ~(1 << 9); // Set P0[9] to Input
PinState = (LPC_GPIO0->FIOPIN >> 9) & 1; // Get P0[9] State
```
Here a zero in a particular position in FIODIR means the pin is used for input, and the relevant bit in FIOPIN contains its current state. Writes to FIOPIN,FIOCLR or FIOSET don't affect input pins, and making changes to the value of an input pin is basically a no-op.

Figure 1: A pull-down resistor allows for more predictable connections between logic gates and inputs. With the resistor, when the button is released, the ground will pull the voltage back down to 0v. Without the resistor, a sufficiently isolated pin might stay at 3v even after the button is released, giving an incorrect reading. Additionally, having a large resistor is important since it will only draw a small amount of current when the button is pressed. A small resistor might draw enough to stress the power supply and keep other portions of your device from functioning

Once you have this set up, you can connect up a switch with a pull down resistor to the input pin, and be able to read the state of your button in software.

2.4.2 Bouncing

So if you want to toggle an LED whenever you press a button you might do something like the following.

```
int state, prevstate = 0;
while(1){
    // Get state of P0[9]
    state = (LPC_GPI00->FI0PIN \gg 9) & 1;
    // If there's a change from 0 to 1
    if(!prevstate && state) {
        // Toggle P0[8]
        LPC_GPIO0->FIOPIN \hat{} = 1 << 8;
    }
    prevstate = state;
}
```
When you try that, you'll notice some odd behavior, not only will the LED change when you press the button, but it will occasionally also change when you release the button. Sometimes it'll miss button presses completely, and not change the LED's state.

Figure 2: What button presses actually look like.

This happens because buttons aren't perfect, and instead of getting smooth transitions from connected to disconnected, the transitions are disjointed and shaky, this phenomenon is known as bouncing.^{[8](#page-8-0)} Because the GPIO pins can only read if something is low or high these jitters result in a number of very fast transitions before the voltage stabilizes. Your LPC will see some of these transitions as a separate button presses and toggle the LED accordingly. Most of the time, the bouncing will happen between reads of the pin state but sometimes, a read will happen in the middle of the bouncing and cause anomalous output.

Removing the errors caused by bouncing can be done with hardware or software. In hardware, using a capacitor or a [Schmitt Trigger](http://en.wikipedia.org/wiki/Schmitt_trigger) can sometimes solve the problem. In software, one can check if the input has been stable for a while before acting upon an event. There are also many other [ways to deal with bouncing](http://www.eng.utah.edu/~cs5780/debouncing.pdf) that can be more complex but also more reliable.

2.4.3 Interrupts

There's another way to get input from GPIO pins, this time without having to stop and poll the state of your button waiting for something to happen. With the correct settings the processor can wait for an event in the background while letting your code run in the meantime. When an event occurs, the processor will stop your currently running code, run code to respond to the event, and restart the execution of your main program. This mode of communication is known as *Interrupt Driven I/O*.

These *interrupts* can be very complex and so we're only going to touch on a small subset of their capabilities here. For GPIO specifically, an interrupt can be triggered when the CPU detects a rising or falling edge^{[9](#page-8-1)} on an input pin. Once the edge is detected, the Interrupt Controller performs a context switch. In this case, saving any of the current registers to the stack, and starting the executing of the interrupt handler. Once the interrupt handler is done, the processor will restore the previously saved registers, and continue executing the original code.

The Interrupt Controller uses an internal flag to determine whether or not to perform a context switch, and this flag is not automatically turned off once it has been set. This means that if you don't manually disable the flag, the interrupt handler will execute repeatedly until the flag is disabled.

⁸ Image taken from http://en.wikipedia.org/wiki/File:Bouncy_Switch.png

 $9A$ rising edge is the pin's value changing from a 0 to a 1, and a falling edge is the value changing from a 1 to a 0.

This might not seem useful, but consider a case where you've got a number of interrupts coming in simultaneously. Because the flag isn't automatically disabled, you can handle one incoming interrupt, disable the flag for that particular interrupt, and know that the Interrupt Controller will automatically call another handler to care of the other interrupts.

Interrupt chaining lets you keep your code small, and modular while still being able to handle many quickly incoming events. By only disabling the flag for the particular event you've handled, you are guaranteed to have your interrupt handler called again, so that it can handle a different event.

2.4.3.1 Setting Up an Interrupt Handler Setting up a GPIO interrupt starts with telling the Interrupt Controller or NVIC 10 10 10 that you want certain pins to trigger interrupts. In the struct LPC GPIOINT you'll find the registers IO0IntEnR and IO0IntEnF which define which pins on GPIO Port 0 generate interrupts on a rising and falling edge. 11

Next the specific interrupt handler must be enabled. All the GPIO interrupts are handled by External Interrupt 3, so we'll use a macro to tell the NVIC that it's allowed to call that specific handler.

```
// Enable Rising Edge Interrupt on P0[9]
LPC GPIOINT->IO0IntEnR | = (1 \lt\lt 9);// Enable Falling Edge Interrupt on P0[9]
LPC_GPIOINT->IO0IntEnF |= (1 << 9);
// Turn on External Interrupt 3
NVIC_EnableIRQ(EINT3_IRQn);
```
In order for the interrupt to do anything, the handler has to be defined. The interrupt handlers are found in cr_startup_lpc176x.c in each of your project's source directories, defined as weak aliases to the default interrupt handler. Because they're weakly defined, defining a new function with the same name will make that the handler for the interrupt.

```
// Turn on the LED when the button is pressed
void EINT3_IRQHandler() {
    // If the rising edge interrupt was triggered
    if((LPC_GPIOINT->IO0IntStatR >> 9) & 1){
        // Turn on P0[8]
        LPC_GPIO0->FIOPIN |= 1 << 8;
    }
    // If the falling edge interrupt was triggered
    if((LPC_GPIOINT->IO0IntStatF >> 9) & 1){
        // Turn off P0[8]
        LPC_GPIO0->FIOPIN k = -(1 \leq 8);}
    // Clear the Interrupt on P0[9]
    LPC_GPIOINT->IO0IntClr | = (1 \lt\lt 9);}
```
All the GPIO Interrupts share the same interrupt handler, so it has to check which pins actually triggered the interrupt. You can do this with the GPIO Interrupt Status Registers. IO0IntStatR will have bits set when the relevant pin was triggered by a rising edge, and **IOOIntStatF** does the same for a falling edge.

 $^{10}\rm NVIC$: Nested Vector Interrupt Controller

¹¹These registers are for pins on GPIO Port 0. There are similar registers with IO2 instead for pins on GPIO Port 2. The other GPIO ports don't have support for interrupts, and don't have interrupt registers.

Once you've done the relevant action you can clear a particular pin's interrupt flag by writing a 1 to the bit in IO0IntClr. This means you only have to handle one pin at a time, and as long as you clear that pin's interrupt flag, the handler will be called again to take care of the next triggered pin.

2.5 Project : Bit Banging

Bit banging is the process of implementing a serhttp://www.cs.umd.edu/~rohit/UESDbook.pdfial communication protocol using software instead of dedicated hardware. In this case we're going to be sending data to a shift register, using 3 GPIO pins to control 8 LEDs.

2.5.1 Shift Registers

To control many LEDs with few outputs you need to implement a serial communications protocol, a way of sending data one bit at a time to another entity. In this case we are going to be sending the data to a CD4094B, which has an interface based around 3 input lines, the clock, the data line, and the strobe input.

Figure 3: Shift Register Timing Diagram

The CD4094B has, in effect^{[12](#page-10-1)}, two registers inside of it, a shift register and an output register. The shift register is used to load in data one bit at a time, so whenever the clock signal moves from low to high it'll do two things:

- 1. It will shift the data it contains one bit to the right.
- 2. Now that the lowest bit is empty, it'll read the value on the data line and store it in that least significant bit.

So this way, through 8 clock cycles you can load one byte onto the shift register, starting with the highest first.

The output register is what is actually connected to the external pins, and determine whether each output pin is held low or high. This register waits till it sees a rising edge on the strobe input, and when it does, it'll copy over the values currently in the shift register.

With this, you can load in a byte of data with the clock and data lines, and once you've finished, write it all at once to the output with the strobe line.

2.5.2 Materials

Other than your LPC you'll need the following:

 12 Thinking of them as memory registers is an imperfect abstraction. While it'll serve for anything we need to do, there are much more detailed explanations on the [CD4094B Data Sheet](http://www.ti.com/lit/ds/symlink/cd4094b.pdf) and on [Wikipedia](http://en.wikipedia.org/wiki/Shift_register)

- 1 x [CD4094B 8-Bit Shift Register](http://www.ti.com/lit/ds/symlink/cd4094b.pdf)
- 8 x LEDs
- 8 x [2N2222A NPN Transistor](http://www.fairchildsemi.com/ds/PN/PN2222A.pdf) $^{\rm 13}$ $^{\rm 13}$ $^{\rm 13}$

The shift register in this circuit can channel only 10mA of current, and if you connected it directly to the LEDs you'd get a glow that's barely visible. The transistors act as simple current amplifiers so that you can have brighter LEDs.

2.5.3 Steps

1. Wire up the following circuit

Figure 4: Shift Register Circuit

- 2. Implement the serial protocol needed to write to shift registers, and display a pattern where one LED turns on at a time, moving back and forth across the back of LEDs.
- 3. Use interrupts to detect the state of a button connected to GPIO and switch between two patterns when it is pressed.
- 4. Use fast switching to make the LEDs glow at different brightnesses while displaying some pattern, and having some interrupt based button interaction.
- 5. **Bonus:** Chain up two more shift registers, and use 8 RGB LEDs to display 3 distinct patterns, one in each color channel. [14](#page-11-1)

¹³The 2N2222A and PN2222A are functionally equivalent transistors with different packages, so feel free to use either.

¹⁴The shift register data sheet has a diagram showing how to chain them for more storage.

3 Clocking

Paragraph about clocking being a tradeoff between processing power and power consumptions, and segue into LPC stuff

Clocks in the LPC all derive from one of three oscillators.

Main Oscillator This is the usual clock source for the LPC and runs at 12MHz.

- **Internal RC Oscillator** This is driven by an internal RC circuit at 4MHz, and is the clock the LPC uses when it's reset. Usually software will later switch to the Main Oscillator.
- **Real Time Clock (RTC) Oscillator** This is a 32.768 KHz Oscillator that powers the real time clock. Because of the precisely chosen frequency, the clock can increment a 32 bit counter on every tick, and overflow once a second.

Each of these clocks can be used as a cpu clock, or with the help of a *Phase Locked Loop (PLL)* generate a much faster CPU clock, up to the LPC's limit of 120MHz.

Before we can get into how to set the LPC's clock, there are two components you need to understand: Clock Dividers, and Phase Locked Loops.

3.1 Clock Dividers

Figure 5: Clock Divider Block Diagram

Clock Dividers are an integral part of any complex clocking system. They allow you to turn a single precisely timed clock signal into another running an integer multiple slower.

As shown in Figure [5,](#page-12-2) internally they have two registers, a counter register, and a factor register. The counter register simply counts all the edges in the input signal. And the factor register stores the amount by which the input signal is slowed down, and is usually user modifiable.

During operation, there is an edge detector, which will trigger two events whenever it sees a rising or falling edge on the input signal.

- 1. It will test if the value in the factor register is equal to the value of the counter register.
- 2. It will increment the counter register.

Then, if the equality test was successful it will perform two more actions.

- 3. Reset the counter register to 0.
- 4. Toggle the state of the output signal, switching low to high and vice versa.

If the number in the factor register is N , then the period of the input signal will be multiplied by $N+1$ and the frequency will be divided by the same amount, as shown in Figure [6.](#page-13-1) So, if *Fin* and *Fout* are, respectively, the input and output frequencies then $\frac{F_{in}}{N+1} = F_{out}$

Input Signal:	<u>UUUUUUUUUUUUUUUUUUUU</u>
Factor 0:	
Output	
	Counter of Colorator algorator algorator algorator algorator algorator algorator algorator
Factor 1:	
Output	
Counter	0 101
Factor 4:	
Output	
Counter	2(3) 4101112131410 $\sqrt{2}$ $\left(3\right)$ 12" 2 $\overline{3}$

Figure 6: Clock Divider Timing Example

3.2 Phase Locked Loops

Figure 7: PLL0 Block Diagram

PLLs are clock multipliers, where a clock divider will divide the frequency of an incoming signal, a PLL will multiply it. A PLL is a complex self correcting feedback loop, but in order to use it all one needs to know is that $F_{in} \times 2 \times (M+1) = F_{out}$ where M is the value in the M-Divider's factor register.^{[15](#page-13-2)}

To see how the PLL works, one must first understand that it is a feedback loop, it starts with an approximation of the final output signal and refines it till it's an exact multiple of the input signal.

Within the PLL is a *Current Controlled Oscillator (CCO)* that will generate the final output signal, it starts oscillating at some frequency that's close, but not quite what we want. The output of the CCO is divided by both the *Scaling Divider* and the *M-Divider* to get an approximation for the incoming signal.

If F_{approx} is the frequency of the approximated signal, we know that $F_{approx} = \frac{F_{out}}{2 \times (M+1)}$, because the output signal is divided twice before it becomes the approximation. If we're lucky enough that $F_{approx} = F_{in}$ we can substitute and solve to show that $F_{out} = 2 \times (M + 1) \times F_{in}$, meaning that we've successfully multiplied the frequency of our input signal.

¹⁵The equation accounts for the extra scaling divider that is in the LPC's PLL0. A standard PLL doesn't have this extra divider, with the CCO being connected directly to the M-Divider, and therefore has the equation $F_{in} \times (M+1) = F_{out}$.

		Start	Error	Correction
Case 1	Input CCO Approximation	MMMMMMMMM	No Error	No Correction
Case 2	Input CCO Approximation	MMMMMMMMMMM	Too Fast	Slow Down CC0
Case 3	Input CCO Approximation	mmmmmm	Too Slow	Speed Up CC ₀

But if $F_{approx} \neq F_{in}$ then how does the PLL make $F_{approx} = F_{in}$?

Figure 8: PLL Error Correction

This is done by the *Phase/Frequency Comparator* which can figure out if the approximation is running at a higher or lower frequency than the input, or if the approximation and the input have a different phase. The difference between the approximation and the input constitute an error term. Figure [8,](#page-14-1) shows the ways the error term can be combined with the current state of CCO to create a refined approximation.

This continual process of error checking and correction means the PLL becomes locked to the input signal, and outputs a precisely frequency multiplied version thereof.

3.3 Calculating Clock Speed

Figure 9: Cpu Clock Generation Diagram

Calculating the current clock speed is a matter of following each of the intermediate steps between the three core oscillators, and the final cclk signal that actually clocks the LPC's CPU.

sysclk This can be connected to any of the core oscillators on the LPC, so its frequency can be 12MHz if connected to osc_clk, 4MHz if connected to irc_osc or 32.7 KHz if connected to irc_osc.

pll0clk This is the output of PLL0, and its frequency must be between 275 MHz and 550MHz.

If you let,

 $N =$ The value in the PLL0 N-Divider's factor register

 $M =$ The value in the PLL0 internal clock divider's factor register

then

$$
F_{\text{p110c1k}} = \frac{2 \times (M+1) \times F_{\text{sysc1k}}}{N+1}
$$

pllclk This is the output of the CPU PLL Selector and can be set to either sysclk or pll0clk,

cclk This is your final system clock, and is pllclk divided by the CPU Clock Divider, this can be at most 120 MHz.

If you let

 $D =$ The value in the CPU Clock Divider's Factor Register

then

$$
F_{\text{cclk}} = \frac{F_{\text{pllclk}}}{D+1}
$$

3.4 Working Backwards

If we want to change our LPC's clock speed, we must answer the following questions:

- Which oscillator should be connected to sysclk?
- Should we use the PLL, and if so what should the N-Divider, and M-Divider be set to?
- What should the CPU Clock Divider be set to?

We also must keep in mind the following restrictions:

- The CPU factor register on supports values between 0 and 255.
- The N-Divider's factor register only supports values between 1 and 32
- The M-Divider's factor register only supports values between 6 and 512 and a number of extra values seen on page 38 in the manual.
- If we're using PLL0 it must output a signal between 275MHz and 550MHz.
- The final clock speed cannot be more than 120MHz.

In order to figure out all of the above we should work backwards from our target cclk frequency, F_{cclk} .

First, we should check if we can forgo the PLL entirely. For every oscillator check if $F_{\text{cclk}} \leq F_{oscillator}$. If it is, check if $\frac{F_{oscillator}}{F_{\text{cclk}}}$ is an integer less than 256. If that too is true, then we can do the following:

- 1) Set sysclk to connect to that oscillator.
- 2) Bypass PLL0, connecting sysclk directly to pllclk.
- 3) And set the CPU Clock Divider to $\frac{F_{oscillator}}{F_{cclk}} 1$.

If we couldn't forgo the PLL, then we need to figure out the PLL settings we'll use. If we're using PLL0, we know F_{p11c1k} is less than 275MHz, and 550MHz, and that for some integer *D* between 0 and 255:

$$
F_{\text{pl1clk}} = (D+1) \times F_{\text{cclk}}
$$

The PLL consumes less power when it runs at a lower frequency, so we should choose the smallest D that will fit within the bounds.

Now we know we've got to set the CPU Clock Divider to *D*, but we still have to figure out which oscillator to use, and what to set the N and M dividers to.

We want this to be true:

$$
F_{\text{p11c1k}} = \frac{2 \times (M+1) \times F_{\text{sysc1k}}}{N+1}
$$

We can solve and substitute to get:

$$
\frac{N+1}{M+1} \approx \frac{2 \times F_{oscillator}}{F_{\text{p11c1k}}}
$$

So for each oscillator we can find the best approximation for the right hand side possible using a valid *N* and *M*. Namely with *N* between 1 and 32, and *M* between 6 and 512 (or another value given on page 38 of the manual). We can then choose which combination of oscillator, *N* and *M* will give us the least error.

Namely if we let

$$
R = \frac{2 \times F_{oscillator}}{F_{\text{p11c1k}}}
$$

 $\delta =$ The error for any combination of oscillator, N and M.

then

$$
\delta = \left(\frac{\frac{N+1}{M+1}-R}{R}\right)^2
$$

Once we have the settings that'll minimize the timing error, we can do the following:

- 1) Set sysclk to the proper oscillator.
- 2) Set the N-Divider and M-Divider to *N* and *M*, respectively.
- 3) Connect PLL0 to pllclk.
- 4) And set the CPU Clock Divider to *D*.

3.5 Making The Changes

Now that we know how to choose the various clocking settings we'll use, we need to learn how to apply them. This involves making **absolutely sure** that you perform certain operations in a certain order, checking and double checking the state of the system, and operations that serve only to prove to the LPC you're paying attention.

Clocking can be a dangerous subsystem to play around in, if we input badly chosen settings, we could render our microcontrollers useless. This is why the LPC's clocking subsystem requires people to jump through hoops to change anything.

3.5.1 Registers

Before we dive into the algorithm to change the settings, we should step through the various registers we'll be using and look at their functions.

- **LPC_SC->CLKSRCSEL** The Clock Source selection register, this is what will connect a particular oscillator to sysclk. Look at page 34 of the manual for more details on how to operate it.
- **LPC_SC->CCLKCFG** The CPU Clock Divider's factor register, this controls the divider placed right before CCLK. See manual page 54.
- **LPC_SC->PLL0STAT** The PLL0 Status register, this read-only register makes the currently applied PLL0 settings visible to you. It has the connection status of PLL0 the N-Divider and M-Divider factor registers, and a status bit which tells you if the PLL is synced yet with its input signal. See manual page 39.
- **LPC_SC->PLL0CON** The PLL0 Control register, this register controls whether PLL0 is on, and lets you choose between connecting pllclk directly to sysclk or to PLL0 See manual page 37.
- **LPC_SC->PLL0CFG** The PLL0 Configuration register, this is where you set the factor registers for the N-Divider and M-Divider. See manual page 37.
- LPC SC->PLL0FEED The PLL0 Feed register, you write afeed sequence to this register in quick succession in order to validate changes you've made to the other PLL0 registers, and actually apply them to the PLL. See manual page 40.

3.5.2 Feed Sequence

The PLL registers together form an update-commit system, where every change of PLL settings requires a feed sequence in order to actually be applied. This exists so that random memory accesses won't change the settings on this device, and if you want to break your LPC you've got to actually work to do it.

A feed sequence consists of writing 0xAA and 0x55 to PLL0FEED one after the other, with no other memory operations on any of the other system control registers in between.

```
void PLL0_feed_sequence(){
    LPC_SC->PLLOFEED = 0xAA;
    LPC_SC->PLLOFFED = 0x55;}
```
3.5.3 Update Algorithm

When you have chosen your settings, there's a well specified algorithm 16 which explains what changes you have to make and in what order. It is very important that you don't combine steps, and make sure that you get no interrupts during this process.

1) Check if PLL0 is already connected, if it is disable it with one feed sequence.^{[17](#page-17-1)}

 $^{16}\mathrm{See}$ page 46 in the manual

¹⁷See the appendix for all the macros used herein

```
if(GETBIT(LPC_SC->PLL0STAT,25)){ // If PLL0 is connected
   BITOFF(LPC_SC->PLL0CON,1); // Write disconnect flag
   PLL0_feed_sequence(); // Commit changes
}
```
2) Disable PLL0 with a feed sequence.

```
BITOFF(LPC_SC->PLL0CON,0); // Write disable flag
PLL0_feed_sequence(); // Commit changes
```
3) If you do not plan to use the PLL, set the CPU clock divider to your final value, otherwise set it to 1.

```
// Change CPU Divider
LPC_SC->CCLKSEL = <CPU Clock Divider Value>;
```
4) Write to the Clock Source Selection Control register to change the clock source if needed.

```
// Change sysclk source
LPC_SC->CLKSRCSEL = <Clock Identifier Bits>;
```
- 5) If you are not using the PLL, you are done, otherwise continue.
- 6) Write values to the N-Divider and M-Divider and use a feed sequence to enable them. The dividers can only be updated when PLL0 is disabled.

```
// Write divider values
SETBITS(LPC_SC->PLL0CFG,0,14,<M-Divider Value>);
SETBITS(LPC_SC->PLL0CFG,23,16,<N-Divider Value>);
PLL0_feed_sequence(); // Commit Changes
```
7) Enable PLL0 with one feed sequence.

```
BITON(LPC_SC->PLL0CON,0); // Set Enable Flag
PLL0_feed_sequence(); // Commit Changes
```
8) Set the CPU Clock Divider to its final value. It is critical to do this before connecting PLL0.

LPC_SC->CCLKSEL = <CPU Clock Divider Value>; *// Change Clock Divider*

9) Wait for PLL0 to achieve lock.

Let

$$
F_{pllref} = \frac{F_{\rm sysclk}}{N+1}
$$

If $100kHz \leq F_{pllref} \leq 20MHz$ wait for the PLL to lock.

while(! GETBIT(LPC_SC->PLL0STAT,26)); *// Spin on Lock Flag*

If $F_{pllref} < 100$ kHz wait for $200/F_{pllref}$ seconds.

int i,count = <Number of cycles with current clock speed>; **while**(i++ < count); *// Wait sensible amount of time*

If $20MHz < F_{\text{ellref}}$ wait for $200\mu s$.

int i,count = <Number of cycles with current clock speed>; **while**(i++ < count); *// Wait sensible amount of time*

10) Connect PLL0 with a feed sequence.

```
BITON(LPC_SC->PLL0CON,1); //Set PLL0 Connect Flag
PLL0_feed_sequence(); // Commit Changes
```
3.6 Peripheral Clocks

Many peripherals are timed using the *Peripheral Clock Dividers*. There are four clocks pclk1,pclk2,pclk4 and pclk8, which are clocks derived from cclk, and are 1,2,4,and 8 times as slow as cclk. Namely, they are implemented with clock dividers with fixed factor registers of 0,1,3 and 7.

You can see how to connect them to various paripherals on pages 56 and 57 of the manual.

3.7 Project : Precision Timing

Many aspects of your LPC require very precise clocking. Things like the USB subsystem, the *Analog to Digital Converter (ADC)* and *Direct Memory Access (DMA)* all perform operations that are timed using your internal CPU clock. The accuracy of those features, and more depends on how accurate your clock speed is.

We are going to experiment with changing the clock speed to generate output signals at various frequencies.

3.7.1 Steps

1) Write a script, in any laguage you want, that will calculate clock settings for you. It should, given a target clock speed, give you everything needed to write the clock setup code.

```
$ ./clk-calc --target=120MHz
Base Oscillator : ...
Use PLL : ...
    N-Divider : ...
    M-Divider : ...
CPU Clock Divider : ...
Target Freq: ...
Output Freq: ...
```

```
Error: ...
```

```
$ ./clk-calc --target=200MHz
```
Cannot Compute: Target Clock Too High

- 2) Write a program that repeatedly toggles a GPIO pin, so that you get a square wave.
- 3) Use make lst and the [ARM Cortex M3 Manual](http://infocenter.arm.com/help/index.jsp?topic=/com.arm.doc.ddi0432c/CHDCICDF.html) to find the number of clock cycles between every toggle.
- 4) Change the frequency of your output square wave without changing the loop at all, use only different clocking settings.

3.7.2 Questions

As you progress through the exercise, answer the followign questions:

- 1) What is the lowest CPU clock speed you can achieve with this setup?
- 2) How many clock cycles does your loop take between GPIO toggles?
- 3) What are the options needed to get the maximum speed, 120MHz?
- 4) What is the frequency of your output signal when the CPU clock is 200Hz? 120MHz?
- 5) What clock frequency do you need to get an output signal at 1kHz?

4 Timers

Intro Paragraph for Timers, something about real time systems?

4.1 Basic Description

Timers are peripherals within the LPC that are mainly internal, they use the CPU clock to keep track of time, and make that ability available to the user in a number of ways.

Timers can send out periodic events, make very precise measurements, simply make the time available for your applications, among other things. This means you can start using temporal information in your program, without having to use unwieldy spin loops, and other ill advised hacks.

There are four timers within the LPC, Tim0, Tim1, Tim2 and Tim3. All are identical, but can have options set independently, and can be used without interfering with each other. '

4.2 Speed Settings

Figure 10: Timer Counter Block Diagram

All timers are built around an internal *Timer Counter (TC)*, which is a 32 bit register that is incremented periodically. The rate of change is derived from the current speed of the CPU clock, which peripheral clock you've connected up and what the prescale counter is set to.

There's nothing more to it than that, the prescale counter is a clock divider like so many others, and the Timer counter is a 32 bit register, and as long as nothing else intervenes it's count from 0x00000000 to 0xFFFFFFFF, overflow, and do it all over again.

4.2.1 Powering Devices

Before we can get to choosing the peripheral clock, and setting the prescale register, we need to actually tun on the timer.

On reset most of the LPC's peripherals are off, and aren't being supplied power by the microcontroller. This can save a lot of energy, but a few core peripherals are turned on when the LPC starts, among these GPIO and Tim0 and Tim1. But this means that Tim2 and Tim3 start off, and if you need them you'll have to turn them on. Additionally, if you don't need Tim0 or Tim1, you can turn them off to save some power.

Power control is considered a system feature, and is controlled by register LPC_SC->PCONP. Each bit in that register is assigned to a peripheral, with a 0 meaning unpowered, and a 1 meaning that the peripheral is powered.

BITON(LPC_SC->PCONP,22); *// Turn on Tim2* BITOFF(LPC_SC->PCONP,2); *// Turn off Tim1*

You can find the full table of peripheral to bit mappings on pages 63 and 64 of the manual.

It's probably a good idea to note that you can get some very weird results if you try to work with a peripheral that's off. The whole situation has undefined results, but in practice, writes to registers in unpowered peripherals don't do anything, and reads always return 0.

4.2.2 Choosing a Peripheral Clock

Likewise, choosing a peripheral clock is a system function, and the settings for all the peripherals are on LPC_SC->PCLKSEL0 and LPC_SC->PCLKSEL1.

```
SETBITS(LPC_SC->PCLKSEL0,4,2,0b10); // Set Tim1 to use pclk2
SETBITS(LPC_SC->PCLKSEL1,14,2,0b00); // Set Tim3 to use pclk4
```
You can find the bit assignments, and settings for the peripheral clock selection on pages 56 and 57 of the manual.[18](#page-22-0)

4.2.3 Setting the Prescale Counter

Once you've made sure your chosen timer is on, and is using the peripheral clock you want, you can move onto setting the prescale counter and actually using it to perform timing related tasks.

The prescale counter is basically a 32 bit factor register inside a clock divider. You can set it using the LPC_TIMx->PC register.

LPC_TIM3->PC = 14; *// Divide the incoming clock by 15.*

All of this can be found on page 495 of the manual.

4.2.4 Reset and Enable

Before the timer counter can actually start incrementing, there are two flags within the *Timer Control Register (TCR)*.

The enable flag, when set to one, disables the timer counter's ability to increment.

```
BITON(LPC_TIM2->TCR,0); // Disable the timer counter
BITOFF(LPC_TIM2->TCR,0); // Enable the timer counter
```
The reset flag, when set to one, forces the timer counter's value to zero.

¹⁸The choice of bits to select each clock is somewhat odd, so do look at the manual

```
BITON(LPC_TIM0->TCR,0); // Disable the timer counter
    // The couter's value is stuck wherever we stopped it
BITON(LPC_TIM0->TCR,1); // Reset the timer counter to zero
    // The counter's value is zero
BITOFF(LPC_TIM0->TCR,0); // Enable the timer counter
    // The counter's value is *still* zero, since the reset bit is still on
BITOFF(LPC_TIM0->TCR,1); // Disable the reset flag
    // The counter's value will now start incrementing, since the
    // reset flag is gone.
```
Once you're done with setting those value properly, you can watch your timer counter increment.

Current_Timer_Val = LPC_TIM3->TC;

4.3 Match Registers

Figure 11: Match Register Block Diagram

Within each timer are four match registers, 32 bit registers which can store a specific match value. When the timer counter and match register are equal, certain events can be triggered.

Choosing the match value is just a matter of setting the match register, LPC TIMx->MRx.

```
LPC_TIM0->MR3 = (uint32_t) 4000; // Set match register 3 on timer 0
LPC_TIM2->MR1 = (uint32_t) 453; // Set match register 1 on timer 2
```
4.3.1 Match Settings

There are three events a match can trigger, a reset of the timer counter, disabling the timer counter and sending an interrupt. The disable and reset events work by modifying the timer control register, and using its features to control the timer counter.

The match register's disable event sets the disable flag in the TCR to 1, requiring you to manually toggle it back. On the other hand, the reset event simply toggles the flag in the TCR, setting the timer counter to 0, but allowing it to continue incrementing,

Insert timing diagrams showing outcome of disable and reset flags

To actually change which events are triggered on match requires manipulating the register at LPC_TIMx->MCR. All the flags for all the timer registers are mapped to various bits in the *Match Control Register (MCR)*. This mapping can be found on page 496 on the manual.

```
BITON(LPC_TIM0->MCR,10); // Reset on match for
                        // match register 3 in timer 0
BOTON(LPC_TIM2->MCR,5); // Disable on match for
                        // match register 3 in timer 2
```
4.3.2 Match Interrupts

Like a GPIO interrupt, there are multiple triggers which all trigger an interrupt on the same channet. Each of the match registers can trigger interrupts, and so can other components of each timer.

To tell these interrupts apart, you can use each timer's *Interrupt Register (IR)*. When a component of the timer triggers the interrupt a flag in the IR is flipped and you can use that to determine what's already going on.

4.3.3 External Match Registers

Explanation of how external pins can be modified by match output without CPU intervention.

Explanation of capture register structure and function with example code

4.4 Capture Registers

Figure 12: Capture Register Block Diagram

channel or make sure it's used ear-

4.5 Project : Serial Light Communications

Explanation of project.

4.5.1 Background

explain rise/fall time, timing and synchronization considerations

4.5.2 Materials

- 1 x LED (use the bright blue ones on the strip)
- 1 x Photoresistor
- 1 x [Potentiometer](http://en.wikipedia.org/wiki/Potentiometer)
- 1 x 2kOhm Resistor
- $1 \times TLV2461$ $1 \times TLV2461$ Op-Amp

4.5.3 Steps

Figure 13: LED Output Circuit

Explain voltage dividers and the maths behind it

5 ADC

Intro Paragraph

The ADC is the analog counterpart to the GPIO's input mode. There are 8 ADC channels each of which can read in a single value at a time.

5.1 Theory of Operation

Explain how the ADC works, what is meant by 'bits of precision' and the mapping from D->A->D.

5.2 Basic Usage

HIT1]: This is un-comfortably like a cookbook one can follow blindly

The initial setup of the ADC consists of the following steps:

1) Power the device

Here we set the bit in LPC_SC->PCONP for the ADC, to turn the power on. This is essentially the same as for the Timers and most other peripherals.

```
LPC_SC->PCONP |= 1 << 12; // Manual page 56-57
```
2) Calculate the necessary clock (the ADC can only run at 13MHz) and choose a peripheral clock, and ADC clock divider setting.

The ADC, like most other components is connected to your choice of peripheral clock and then run through its own divider. The ADC is also limited to a 13 MHZ clock, so you must choose values for its internal clock divider, and peripheral clock such that

$$
\frac{F_{\tt pclkx}}{N+1} < 13{\rm MHz}
$$

The easiest way to do this is to simply set your central clock to the 12MHz main oscillator, and the various other dividers to pass that through unchanged.

```
LPC_SC ->CLKSRCSEL = 1; // Select main clock source
LPC_SC ->PLL0CON = 0; // Bypass PLL0, use clock source directly
// Feed the PLL register so the PLL0CON value goes into effect
LPC_SC ->PLL0FEED = 0xAA; // set to 0xAA
LPC_SC ->PLL0FEED = 0x55; // set to 0x55
// Set clock divider to 0 + 1=1
LPC_SC ->CCLKCFG = 0;
```
But you can also use settings that will let your LPC run faster, and get more out the main processor while allowing the ADC to run as fast as possible,

3) Set the peripheral clock

Here we use the undivided clock for the ADC, but as long as the final clock constraints are considered, you can choose any peripheral clock.

```
// Choose undivided peripheral clock for ADC
LPC_SC->PCLKSEL0 k = -(3 \lt 24);
LPC_SC->PCLKSEL0 |= (1 << 24);
```
4) Set the ADC clock divider

The ADC control register, LPC_ADC->ADCR controls a number of functions, but for the moment we'll use it to choose the setting for the ADC clock divider. This is controlled by bits 8 through 16 of the control register and can be set as follows.

```
SETBITS(LPC_ADC->ADCR,8,8,0); // Set clock divider to let the
                              // clock pass unchanged
```
5) Put the pins you'll be using into ADC mode

SETBITS(LPC_PINCON->PINSEL1,14,2,0b01); *// Connect AD0.0 to its pin*

6) Pull the ADC out of power down mode

The ADC also has its own internal power switch, so that you can change settings while conserving power that the conversion circuitry will use.

BITON(LPC_ADC->ADCR,21); *// Turn on ADC internal power*

The simplest method of actually getting data is synchronous collection of values from the ADC. You tell the ADC to go collect some data, wait for it to tell you it's done, and then read out the value.

When trying to synchronously access one of the eight ADC lines, do the following:

1) Select the ADC line you're going to read.

The first eight bits in LPC_ADC->ADCR control which input lines are active, and when collecting data synchronously, you can only have one on at a time.

LPC_ADC->ADCR &= 0xFF; *// Clear first 8 lines* BITON(LPC_ADC->ADCR,0); *// Choose line 0*

2) Start the conversion

There are 3 bits in the ADC control register which let you choose the collection mode, for the moment we'll focus on the first two. Setting bits 24 to 26 in the ADCR to 0 tells the ADC that you want no conversion done, and setting them to 1 tell the ADC to start a conversion immediately.

SETBITS(LPC_ADC->ADCR,24,3,1); *// Start the single conversion*

3) Spin on the done flag

For single conversions you can look at the *General Data Register* which will store the results of the very last conversion. Because a conversion takes 65 of the ADC's clock cycles, where it'll spend time slowly increasing the precision of the value it recovers, you have to wait for it to finish. So you spin on the done flag in the final bit of the data register, which will turn to 1 when the conversion is finally finished.

while((LPC_ADC->ADGDR & (1 << 31)) == 0);

4) Read and parse the output value

And now you can find your converted value in the same register, in bits 5 through 12.

```
value = GETBITS(LPC ADC-> ADGDR, 5, 12);
```
5.3 Interrupts

Explain how to use the ADC interrupt, when it's thrown, what you need to flip, and how to get periodic ADC interrupts

If you want to sample asynchronously, using interrupts you can allow your other code to run during the relatively long sampling process.

The setup for interrupt based ADC use has a few extra steps:

1) Enabling the ADC interrupt in the NVIC.

```
NVIC EnableIRQ(ADC IRQn);
```
2) Setting the correct bits in the ADC Interrupt Enable register.

```
BITON(LPC_ADC->ADINTEN,0);
```
3) Setting up the interrupt handler that will retrieve your value.

Reading from the GDR will turn off the interrupt flag so you don't have to do it manually.

```
void ADC_IRQHandler(void) {
    /** other stuff **/
    // Equivalent to above:
    analog_val = (LPC_ADC ->ADGDR >> 4) & 0x0fff;
    /** other stuff **/
}
```
Then you can start a conversion the same way as before.

SETBITS(LPC_ADC->ADCR,24,3,1); *// Start the single conversion*

65 ADC clock cycles after you do, the interrupt will be thrown and you can retrieve the value.

5.4 Connecting to Timers

Explain how to connect to a timer, trigger on match registers, capture registers, how to use a match interrupt to trigger an ADC interrupt and then get a value while your main loop is chugging along doing its thing.

5.5 Burst Mode

Explain what burst mode is and how to use it.

5.6 DMA

Write when we get to the DMA chapter

5.7 Project : Optical Theramin

A Theramin is a Russian musical instrument, invented in 1919, and is one of the few instruments played without ever being touched. There are two large antennas on a theramin, which can detect the approximate position of a musician's hands, and uses that to modulate the sound it produces. One hand is used to control the pitch, and the other is used to control the volume.

In a standard theramin the detection works by using its antennas and the hands of the artist as opposing plates in a capacitor. By moving their hands around the artist can change the capacitance, and the instrument can use that change to generate different sounds.

isert link to
ouTube video of amin[?]

Figure 15: Hand Sensor

Our version will work of a slightly different principle, we'll use light to sense the distance between the hand and our sensor. We'll place an LED below a photo resistor, so that when there's no hand or other obstacle, the light from the LED will just shine onto the insensitive read of the photoresistor. But when we place our hands above the sensor, the light from the LED will reflect back onto the sensitive portion of the photoresistor and give us something we can sense. The farther away the hand is, the less light will be reflected back, so this setup will give us an approximate measure of distance. We can use those measurements to modulate the output of our DAC which will be plugged into a speaker, giving us a simple optical theramin to play with.

5.7.1 Background

5.7.1.1 Voltage Dividers Voltage Dividers are an essential part of the circuitry for our theramin, and in order to understand how it all works, you must understand how these components work.

Figure 16: Voltage Divider Schematic

A standard voltage diver circuit is made with two resistors placed in series, the voltage at *Vdiv* is the weighted average of the voltages at V_{++} and V_{--} .

$$
V_{div} = (V_{++} - V_{--})\frac{R1}{R1 + R2} + V_{--}
$$

Work through a few examples of voltage divider calculations, explain how this only applies when there's little to no load.

 $\# \# \# \# \#$ Potentiometers $\# \# \# \# \#$

Figure 17: Potentiometer Divider Schematic

Potentiometers are a special type of voltage divider which you can adjust on the fly. A potentiometer is essentially a long resistor, where you can move V_{div} up and down along it. As such, $R1 + R2$ is always going to be constant and the voltage at *VDiv* depends on the position of the potentiometer's dial. The actual value can go all the way from V_{++} , to V_{--} since you can lower either $R1$ or $R2$ down to 0.

5.7.1.1.2 RC-Filters Expand the following

- Signals having frequency components
- DAC producing impure signals because of the difference between transition time and rest time.
- Need to remove the high frequency components so that we only preserve the frequency we want to output

Figure 18: Op-Amp Schematic

5.7.1.2 Op-Amps *Operational Amplifiers (Op-Amps)* are some of the most powerful standard circuit components. They are circuits that run an extremely simple algorithm that can be leveraged in very powerful ways.

The Op-Amps can either push or pull current through their output, and thanks to Ohm's law $V = IR$ this means that they can control the voltage at their output as well.

In figure [18](#page-33-0) the Op-Amp is connected through a resistor to a voltage source. Depending in the state of its inputs it can choose to do one of three things. First among these is to prevent the flow of current either in or out, in this case there will be no current flowing cross the resistor and the voltage at *Vout* will be 1.5v.

Next it could push current outwards, meaning there'll be a current flowing from the output to the 1.5 volt source. Because of Ohm's law, the voltage at the output will increase, if the resistor is large, the voltage will increase very quickly, until it hits the limit of 3 volts imposed by the Op-Amp's power source. Alternatively if the resistor is very small, it'll take a lot of current to increase the voltage a small amount, to the point that the power source simply can't supply more current. If that happens the voltage at the output will hover at whatever value the current available can sustain.

Op-Amp modulate their output voltage with a simple algorithm. If their two inputs are held at the same voltage, the Op-Amp will keep the output stable. If $V_{in+} > V_{in-}$ then the Op-Amp will raise the voltage at its output, and if $V_{in+} < V_{in-}$ then the Op-Amp will lower the voltage.

This property can be exploited to create a number of useful circuits.

Figure 19: Op-Amp Voltage Follower Schematic

5.7.1.2.1 Op-Amp Voltage Follower An Op-Amp voltage follower has a simple function, basically it makes sure that the output voltage is always equal to the input voltage. To understand how this is useful we've got to first understand how loads can change the voltages at a point.

Figure 20: Load Voltage Schematic

In figure [20,](#page-34-0) we've set up two voltage dividers, using the same R_1 and ₂ but the second divider has an extra load resistor applied to it. V_1 will be the same value as before, but because of the existence of the additional load resistor, V_2 will be a different, higher, voltage.

If, instead of connecting the load resistor directly to your divider, you connected the divider to the voltage follower's input, and the load resistor to the output, as with the third setup, the voltages at *V*1, *V*³ and *V*⁴ will all be equal. Here the Op-Amp is compensating for the change in voltage the load would otherwise cause.

What is happening is that when $V_{in+} > V_{in-}$ the Op-Amp will increase the voltage at the output, and that will (being directly connected) increase the voltage at V_{in-} . When the opposite is true and $V_{in+} < V_{in-}$ the Op-Amp will decrease the voltage at output, and *Vin*−. Both of those actions will correct the imbalance, and make $V_{in+} = V_{in-}$ and therefore $V_{in+} = V_{out}$. The Op-Amp will automatically sense changes in the load, and compensate so that the output is always equal to the input.

Figure 21: Op-Amp Voltage Comparator Schematic

5.7.1.2.2 Op-Amp Voltage Comparator This is very similar to the Voltage Follower, but where the voltage follower has a feedback loop, this circuit has no connection at all. This means that when $V_{in+} > V_{in-}$ the output voltage will keep on going up till it hits the maximum voltage the Op-Amp can sustain, in our case usually 3v. Likewise when $V_{in+} < V_{in-}$ the voltage will go down until it hits the lower limit, namely 0v.

So the output is limited to 0 and 3v, and its state will tell whether $V_{in+} > V_{in-}$ or not, comparing the two input values.

Figure 22: Op-Amp Voltage Amplifier Schematic

5.7.1.2.3 Op-Amp Voltage Amplifier The amplifier is based off the same principle, but instead of a direct connection, it uses a voltage divider. The voltage divider formula tells us that the, voltage at $V_{in} = (V_{out} - V_{ref}) \frac{R_2}{R_1 + R_2} + V_{ref}.$

Thanks to it's nature as a feedback loop, the Op-Amp will change V_{out} so that $V_{in} = V_{in+}$. If we assume that $V_{ref} = 0v$, then we can solve the equation for V_{out} and see the following holds:

$$
V_{out} = V_{in+} \frac{R_1 + R_2}{R_2}
$$

So, V_{out} is a straightforward multiple of V_{in+} , and the amount by which it is multiplied, $\frac{R_1+R_2}{R_2}$ is the gain of the amplifier. You can do the same thing for other values of *Vref* to see that it's esentially the midpoint around which you're multiplying. So if $V_{in+} = V_{ref} + 2v$ then $V_{out} = V_{ref} + \text{gain} \cdot 2v$.

5.7.2 Materials

- 2 x LED (use the bright blue ones on the strip)
- 2 x Photoresistor
- 2 x 2 kOhm Resistor
- 1 x 22 Ohm Resistor
- 2 x 10 kOhm Resistor
- 1 x Speaker
- 1 x 1 Microfarad Capacitor
- 1 x Potentiometer (Higher Resistance is better)
- 2 x [TLV2461](http://www.ti.com/lit/ds/symlink/tlv2461-q1.pdf) Op-Amp

5.7.3 Steps

1) Build 2 Hand Sensors, test with oscilloscope.

Figure 23: Hand Sensor Circuit

These are voltage dividers which will use the changing resistance of the photoresistor to change the voltage at the ADC input. Once the circuit is assembled connect the oscilloscope probe to the ADC input and wave your hand above the sensor to get something like figure **??**.

- 2) Connect the hand sensors to your LPC and check if you can receive ADC values with GDB or semi-hosting.
- 3) Get DAC to output sin waves, verify with oscilloscope.

Once you connect your DAC output to the oscilloscope, you should see something like figure [24.](#page-36-0)

Figure 24: DAC Sine Wave Oscilliscope Trace

4) Build Voltage Reference

Figure 25: Voltage Reference Circuit

Since your DAC is limited to values between 0 and 3 volts, we can't amplify the output around 0v, so we're constructing a reference voltage for the amplifier. First we use a voltage divider to get a point at 1.5v and then a voltage follower so that we can attach loads to that point, and have it stay stable.

Once you've constructed this section of the circuit, use the oscilloscope or multimeter to check that the voltage at V_{ref} is 1.5.

5) Build RC Filter

Figure 26: RC Filter Schematic

The RC filter is a component we use to smooth out the jagged edges of the DAC output. At high frequencies the output of your DAC will look like figure [27](#page-37-0) , with an obvious step from one output voltage value to another. When you play this sound you'll be able to hear the high frequency shifts as a seperate tone from the sine wave you're otherwise playing.

Figure 27: High Frequency DAC Sine Wave Oscilliscope Trace

To fix this you can build an RC filter, this circuit will smooth out the wave by forcing it to pour energy into the capacitor and slowing the change in voltage. The exact workings of the filter are outside the scope of this book, but suffice to say that it, when connected, will transform the raw DAC output in figure **??** into the signal seen in figure [28,](#page-37-1) where the signal in yellow is the DAC after it's connected, and the signal in pink is the voltage at *Vsmooth*.

Figure 28: High Frequency RC Filter Oscilliscope Trace

6) Build Amplifier

Figure 29: Speaker Amplifier Circuit

We can use the amplifier to change the volume of the final sound, and the amount of current the Op-Amp will supply.

If you connect up the oscilloscope now (before adding the speaker), and have your DAC output a sine wave, you should be able to modulate output to any gain between 1 and infinity, effectively turning the amp from a voltage follower to a voltage comparator and anywhere in between.

Sample traces are shown in figures [30](#page-38-0) and [31,](#page-38-1) with *Vsmooth* in pink, *Vref* in green, and *Vspeaker* in blue.

Figure 30: Speaker Amplifier with Small Gain

Figure 31: Speaker Amplifier Acting As Comparator

7) Connect Speaker Circuit

Figure 32: Speaker Circuit

Connect up your speaker to the other circuit components, and play some sound.

8) Add Calibration Routines and Basic IO

In order to actively use the theramin you'll have to set up a calibration routine.

The amount of light on the photo-sensors, and therefore the voltage coming into your ADCs can vary wildly depending on the brightness of the room, the relative position of your sensors and LEDs and a number of other environmental factors. So in order to actually play your theramin you'll have to set up a routine to tell your LPC what range of inputs it should expect in the current environment.

This means you'll have to set up a routine where you tell the LPC to record the maximum and minimum voltages for each hand, and scale the output frequency and amplitude accordingly.

9) **Bonus:** Play something recognizable with your newly created instrument.

6 DAC

Intro Paragraph

The DAC is the analog counterpart to GPIO's digital output. Your LPC has one output line, and

6.1 Working Theory

How does the DAC work, nyquist sampling, frequency? also rename this section

6.2 Usage

To set up the DAC one has to connect a particular peripheral clock to the DAC and connect the DAC output to the corrent pin.

```
LPC_SC->PCLKSEL0 |= (1 << 22); // Set pclk to 1
LPC_PINCON->PINSEL1 |= 1 << 21; // Set H[18] to AOUT
```
To use the DAC one has to write to the DAC converter register.

```
SETBITS(LPC_DAC->DACR.6,10,1035); // Set DAC to 1035
```
7 DMA

Finish the intro

Often, when building embedded devices you'll have to deal with large amounts of incoming and outgoing data. Be it reading thousands of samples per second from the ADC as you record sound, reading megabytes per second external memory, or sending large chunks of data over Ethernet quickly.

The simplest way to deal with these large data streams is using the CPU. Loop or timer based mechanisms can move large amounts of data back and forth but at the cost of using CPU time to perform routine memcopy style tasks. Occasionally it's simply not passible to handle data at full speed, or satisfy IO timing requirements using the CPU, and other methods are needed.

This is where *Direct Memory Access (DMA)* comes in, DMA is a special deeply integrated peripheral that can perform memory manipulation tasks independent of the CPU. This means that it's possible to perform routine data gathering tasks faster than the CPU could, without wasting CPU time, and with a level of timing precision that the CPU can't match.

The most basic use of DMA is a memory-to-memory transfer, which will copy data from one block of memory to another. This is analogous to a standard memcpy operation, albeit without blocking. One can also choose to have an interrupt thrown when the copy is completed in order to avoid race conditions.

The next two uses for the DMA controller are memory-to-peripheral tranfers, and peripheral to memory transfers. These data transfers are deeply integrated with the revelant peripherals, and can be used to automatically gather data from an input, or automatically send data to an output.

Consider the output case, where we have a buffer of data we wish to write out to the DAC at specifically timed intervals. The DAC has a programmable countdown timer, which can ask the DMA controller for a new piece of data after an interval. When the DAC timer finishes and a DMA request is sent, the DMA controller will write a word into the DAC output register, and trigger an automatic outut update. At this point, the DAC will reset the timer, and start counting down to another update, and the DMA controller will ready the next byte of data for the DAC. In this way you can fill a buffer with sound data, and write it out at the same rate it was recorded.

For the input case we can look at the ADC. Like the DAC, the ADC has a timer, which can trigger a conversion, and then assert a DMA request when it's done. Here, when the ADC timer counts down, it'll immidiately start a conversion, and reset itself. When the conversion finishes, the DMA controller will read the ADC input register, and store returned data in the predefined output buffer.

For SSP and other serial communications peripherals, the DMA controller can read or write data as fast as it can, sequentially storing or sending large chunks of data.

The final case is a peripheral-to-peripheral transfer which can be useful to forward data from one communications protocol to another without manual intervention.

haven't really used p-to-p transfers

Add detail, in the meantime read pages 607-615 of the manual

7.1 GPDMA in the LPC

The DMA controller in the LPC is called the *General Purpose DMA (GPDMA)* controller. Mostly to reflect the fact that it can work

7.2 DMA with the ADC

ADC DMA Instructions go here. In the meantime read the DMA, and ADC sections of the manual

7.3 DMA with the DAC

DAC DMA Instructions go here. In the meantime read the DMA, and DAC sections of the manual

7.4 DMA and Power Control

Explain how DMA interacts with sleep mode and other power control feature

7.5 Project : Sound Recorder

This project will have you using the LPC to record sound, store it on an SD Card, and play it back on demand. You'll have to use DMA to gather audio samples at precise frequencies, and use external storage to make up for the miniscule amount of onboard storage the LPC has.

7.6 Suggested Steps

- 1) Get DMA and the DAC working to output standard waveforms (sine waves, sawtooth functions, triangle waves), test using the oscilloscope and speaker circuit from the theramin project.
- 2) Use the signal generator, or your DAC output to test reading data in with ADC and DMA
- 3) Make sure you can send and retrieve data with SPI and your SD-Cards.
- 4) Build the Microphone circuit, make sure you get sensible output on both the oscilliscope, and when you read data into your LPC.
- 5) Add some IO LEDs and connect all the parts together so you can record at least 30 seconds of audio.

7.7 Microphone Circuit

7.7.1 Floating Ground

Figure 33: Floating Ground

• Test by making sure the voltage on the 1.5v node is correct.

We're using the Op-Amp to create a voltage source at Vcc/2, since we need a center point for our final signal to go above and below. This center point also needs to be able to sustain a significant current draw for our other components, which is why we use an Op-Amp. If we simply used a voltage divider, the resistance of the divider itself would keep the other components from drawing the current they need, you could try making the resistors small, but then the current traveling through the divider itself would be problematic.

7.7.2 Microphone Assembly

Figure 34: Mic Assembly

- The electrolytic capacitors (the big cylindrical ones) are polarized, the negative side has a stripe on it.
- Make sure that when you connect the oscilloscope to this (ground to 1.5v, probe to Coarse Out) you see a sound signal in the millivolt range.
- Electret Mics also have a polarity, the negative pad on the bottom has a little bronze dash etched onto the PCB next to it. It's the only asymmetry on the bottom of the mic.

These microphones are basically air pressure sensitive transistors, and with the 2k resistor, it acts like a transistor amplifier, with the input replaced by air pressure. The capacitor and the 10 resistor together form a high pass filter, what this does is filter out the DC element of the air pressure, and any components too low in frequency to hear.

7.7.3 Filters

Figure 35: Filter Stage

• Test by connecting the oscilliscope to this portion of the circuit (ground to 1.5v, and probe to Fine Mic Out), and making sure you see a signal in the millivolt range.

This is two low pass filters in series, which will significantly reduce the amplitude of high frequency noise.

7.7.4 Offset Voltage

• Test the voltage at the Offset Voltage point, and make sure that it moves up and down when you fiddle with the potentiometer.

Figure 36: Offset Voltage

The output from the microphone and the filter aren't perfectly centered on the 1.5v value so this offset voltage lets you tweak the final centering of the signal to fix that. In the final output, this pot will move the signal up and down on the oscilloscope.

7.7.5 Amplifier

Figure 37: Microphone Amplifier

• Make sure the output changes scale with the new potentiometer, and moves up and down with the offset potentiometer, and that you can get output signals that have a 3v amplitude.

Here we use an Op-Amp amplifier to scale the microphone's signal to levels which our LPC can read.

8 Appendix

8.1 Development Environment Setup

Executables destined for a microcontroller are somewhat different than executables you intend to simply run. Where a normal executable will be loaded by an OS, we need to make files that can be flashed, bit for bit, onto the microcontroller's memory and run. To greatly simplify, an OS takes an executable file, loads it into memory, looks for a pointer to first instruction to execute in the executable's metadata, and starts a process with the program counter set to that location. We have no OS to offload a lot of this work to, we must build files that can be moved directly onto the LPC's memory, and run on a bare machine.

Our executables will have the loaders and initializers that an OS would normally provide, in addition to our actual programs. In effect creating a minimal OS that exists only to load our app, and get out of the way. These loaders are provided by our toolchain, and allow us to concentrate on the main application

To do this we will be using a toolchain building applications for arm-none-eabi , which means:

- **arm** The CPU architecture we're building our applications for, this determines the instructions and registers we have available, among other things.
- **none** This is the operating system we're building for, in this case we're building applications that run on the bare metal, with no intervening OS.
- **eabi** The Embedded Application Binary Interface, this defines the conventions for data types, file formats, stack usage, register usage, and function parameter passing. Having a standard specification for this ^{[19](#page-47-2)} means different libraries can be compatible with each other and your code.

This toolchain comes with a number of tools, :

- **as** The GNU assembler, which takes the assembly our compiler outputs, and gives us ARM compatible machine code.
- **gcc** The GNU compiler collection which will take our C code, and turn it into assembly.
- **gdb** The GNU debugger, you can use this to view your program as it's executing, and see what its internal state is.
- **ld** The GNU linker, which takes compiled object code and can correlate all the symbols and link it into a single executable.

There's also a number of less important tools from [Gnu Binutils](http://www.gnu.org/software/binutils/) that we'll be using, all targeted to arm-none-eabi.

8.1.1 Install Steps

8.1.1.1 LPCXpresso In order to save building these tools ourselves, and to get the tools which will allow us to flash the LPC through the proprietary LPC-Link board, we'll have to download and install LPCXpresso, a set of tools and Eclipse based IDE for working with the LPC.

To download the tools, visit the [LPCXpresso Website](http://lpcxpresso.code-red-tech.com/LPCXpresso/) and register an account. Once you have confirmed your email, you will be able to visit the [download page.](http://lpcxpresso.code-red-tech.com/LPCXpresso/Downloads) There, download the latest version of LPCXpresso 5

¹⁹The ARM EABI Standard : http://infocenter.arm.com/help/topic/com.arm.doc.ihi0042e/IHI0042E_aapcs.pdf

for your operating system and follow the provided installation instructions. Take note of the installation directory, we'll use it later.

At this point, you can launch LPCXpresso and use it, if you would like. It is recommended to launch it at least once and go through the software activation step (no payment necessary, you are just verifying your email again), as this upgrades the proprietary tools to handle a larger file size.

8.1.1.2 [Git](http://git-scm.com/downloads) You also need to install the Git version control system.

Git is a similar tool to SVN or CVS and allows you to store your code's history and synchronize development with other programmers.

We use git to manage our build system.

8.1.1.3 [CMake](http://www.cmake.org/) Next, install the CMake build system, make sure you install the very latest version, a number of distros do now have the proper binaries in their repositories, and you might have to build it yourself. Windows Instruction and networks of the structure of the st

CMake is a meta-build system. Instead of the standard UNIX make, which directly compiles source code into the desired output format, CMake instead generates one of many different types of build systems,which are then used to perform the actual compilation. This is useful for allowing different developers to use their preferred development environment (POSIX, Visual Studio, Eclipse, etc. . .). In our example, we will use CMake to generate Makefiles.

8.1.1.4 MinGW If you're on windows, you should also install [MinGW.](http://www.mingw.org)

MinGW or Minimalist GNU for Windows is a package containing all of the usual GNU tools, and GNU binutils, specifically targeted to compile windows binaries.

8.1.1.5 UMD LPC Build The UMD LPC Build system was created in order to let us use the LPC with our own choice of editors for coding, and a standard gdb session to debug, instead of having to deal with a clumsy, hard to use proprietary IDE.

To install, start by cloning the repository from github, while in whatever directory you wish to work out of:

git clone https://github.com/rohit507/UMD-LPC1769-Build.git

This will have git retrieve the latest version of the build system from github. We've also created a few submodules: separate repositories storing single projects that let you manage them with git at the project level instead of having to makes changes to the entire workspace whenever you wish to change a project. To initialize these, run the following:

```
# Get submodules
cd GNU-LPC-Core/
git submodule init
git submodule update
```
Next we need to run the setup script and set a few internal variables so CMake will know where it should look for the **arm-none-eabi** toolchain, and proprietary tools. Here the DLPCXPRESSO DIR variable should be set to the root directory under which all of the necessary files can be found, this should contain the lpcxpresso $\frac{\text{Ad} \text{dim} \text{U}}{\text{dom} \text{det}}$

ndows Instr
ns,and path changes.

footnote

```
# Run setup script
cd _setup
cmake . -DLPCXPRESSO_DIR=<LPCXpresso Root Dir>
```
During setup, our script copied over the CMSIS library, this is a library that provides some very basic LPC functionality, but we still need to unzip it.

On Linux or OS X:

```
# Extract CMSIS library .zip
cd ..
unzip CMSISv2p00_LPC17xx.zip -d CMSISv2p00_LPC17xx
```
On Windows: Add instructions here

Now from within the CMSIS directory, we'll use CMake to generate a build system.

On Linux or OS X:

```
# Build CMSIS
cd CMSISv2p00_LPC17xx/
cmake . -G "Unix Makefiles"
make
```
On Windows: Add instruction

Now we can build our skeleton project, which is the bare template all your projects will be build from. On Linux or OS X:

```
# Build CMSIS
cd ../Skeleton
cmake . -G "Unix Makefiles"
make
```
On Windows:

Add instructions here

At this point all your installation steps are done and we can go over basic usage, and workflow.

8.1.2 Available Targets

The default build target builds a .axf file with debug settings.

The following targets are provided:

lst Generate a .lst file, which includes an overview of all the sections and symbols in your output, along with a complete disassembly.

²⁰On Linux or OSX it should look like /usr/local/lpcxpresso_5.1.2_2065/lpcxpresso/. On Windows it should look like TODO: Insert actual path here.

- **hex and bin** Generate alternate formats of the normal .axf output, which may be useful if you want to use other tools.
- **boot** Boot the LPC-Link board. This is required before flashing or debugging the microcontroller using the LPC-Link board.
- **flash** Write the currently built output to the microcontroller. The microcontroller immediately starts running after the flash is complete.
- **flash-halt** The same as flash, but the microcontroller is left in a stopped state.
- **gdb** Launch a gdb session. Complete debug information is also setup. Note that it is not always simple to restart the current program. run (r) will work for some programs, and sometimes, using jump (i) to jump to the entry point of the current image (found with info files) works. However, none of these completely reset the chip - the only command that will is the load command, however that will flash the entire image to the chip again, which may take a while for projects with large compiled binaries. Also note that some circuits (particularly when interfacing with other chips that have their own state) may require completely unplugging the circuit between program runs.

If using makefiles, these are directly accessible as make lst, etc. (run in the root directory of the desired project). If you are using a build system based around IDE project files (such as Eclipse or Visual Studio), the targets should be accessible from a menu.

8.1.3 Typical Workflows

These sections detail how to do many common tasks. They are written under the assumption that a Makefile build system is being used, but the calls to make can all be substituted with the appropriate action in any other chosen build system.

8.1.3.1 Start a new project (by forking): First go [here](https://github.com/rohit507/GNU-LPC-Skeleton) and fork a new project.

In your terminal:

```
git submodule add http://github.com/yourname/newprojectname
git add .gitmodules
git commit -m "Added a submodule for NewProjectName"
git submodule update
cd NewProjectName
YOUR_EDITOR CMakeLists.txt
# Edit CMakeLists.txt, and change the project name.
cmake . -G "Unix Makfiles"
make
```
8.1.3.2 Start a new project (by copying): In your terminal:

```
cp -R Skeleton/ NewProjectName
cd NewProjectName
YOUR_EDITOR CMakeLists.txt
# Edit CMakeLists.txt, and change the project name.
cmake . -G "Unix Makfiles"
make
```
8.1.3.3 Work on an existing project In your terminal:

```
# Open source files in editor and make + save changes
make
```
8.1.3.4 Enabling semihosting for a project Open a project's CMakeLists.txt file and uncomment the following line:

```
set(SEMIHOSTING_ENABLED True)
```
Then just rebuild the project, and semihosting messages will be viewable in gdb.

8.1.3.5 Adding compiler options All compiler options are configured in Platform/LPC1769_project_default.cmake. To add compiler options to a single project, you can use CMake's add_definitions command in that project's CMakeLists.txt.

In your CMakeLists.txt:

```
add_definitions(
  -Wall
  -Werror
  -02# Add more compiler flags here
)
```
8.1.3.6 Add a source file to a project When adding source files to a project, remember to open that projects CMakeLists.txt and add the file to the SOURCES variable.

In your CMakeLists.txt:

```
set(SOURCES
  src/cr_startup_lpc176x.c
  src/project.c
  # Add more source files here
)
```
8.1.3.7 Flash a program In your terminal:

```
# Plug in the LPC1769
make
make boot
make flash
```
8.1.3.8 Debug a program In your terminal :

```
# Plug in the LPC1769
make
make boot
make gdb
```
In GDB:

```
b main
load
c
# Make changes to the program
make
load
c
```
For more information on how use GDB refer to a [simple tutorial](http://www.cs.cmu.edu/~gilpin/tutorial/) or the [full user manual.](https://sourceware.org/gdb/current/onlinedocs/gdb/)

8.2 Standard Macros

*/**

```
* BitMacros.h
 */
#ifndef BITMACROS_H_
#define BITMACROS_H_
   // Some macros to speed bit twiddling
   #define BITVALUE(var) (1 << (var))
   #define BITON(var,bit) var |= (1 << bit)
   #define BITOFF(var,bit) var &= ~(1 << bit)
   #define BITTOG(var,bit) var ^= (1 << bit)
   // MASK(3,4) = 0x78
   #define MASK(start,len) (~(~0 << (len)) << start)
   #define INVMASK(start,len) (~MASK(start,len))
   // OR with things to set all 1s
   #define ORMASK(start,len,val) (MASK(start,len) & ((val) << (start)))
   // AND with things to set all 0s
   #define ANDMASK(start,len,val) (INVMASK(start,len) | ((val) << (start)))
   // Shift a set of bits over and mask it to that length
   // same as or mask with slightly different ordering
   #define SHIFTMASK(val,start,len) ORMASK(start,len,val)
   // SETBITS does two write passes, first pushing all
   // the 1s from val, and then all the 0s. This should
   // keep from re-setting any bits that are already or
   // writing 1 to any unset bits.
   #define SETBITS(var,start,len,val)\
       do { \
            var | = ORMASK(start, len, val); \
           var &= ANDMASK(start,len,val); \
       } while (0)
   #define GETBIT(var,start) (((var) & (1 << (start))) >> (start))
   #define GETBITS(var,start,len) (((var) & MASK(start,len)) >> start)
   #define LOWBYTE(var) ((char) ((var) & 0xFF))
   #define HIGHBYTE(var) ((char) (((ushort_t)(var)) >> 8) & 0xFF)
#endif /* BITMACROS_H_ */
```