

PYKC 5 Nov 2019

E2.1 Digital Electronics

Lecture 10 Slide 1



This lecture is not the same as previous ones. I am not teaching you any new concept, architecture or circuit. Instead, I will go through the entire VERI Lab Experiment in order that you appreciate what I want you to learn in each of the four parts.

I will also point out the various pitfalls that students always make each year, and some of the useful "tricks of the trade".

## **Overview of the four parts in VERI**

Parts	What you will learn?
Part 1	Why Verilog HDL is much better than schematic capture?
Part 2	How to design counters circuits and simple FSM? How to cascade multistage counters in the right way?
Part 3	How does SPI serial interface works? How to use ROM, multiplier and DAC to produce a sinewave of different frequencies? Compare the DAC output and PWM output.
Part 4	How to perfect ADC and DAC in an audio processing system? Create a real-time echo chamber effect.
<ul> <li>VERI is designed</li> <li>Each part is built to</li> <li>Each part has its of</li> <li>Each part has an of</li> <li>VERI is the ONLY description language</li> </ul>	to teach you digital electronics in four steps. upon the previous part. own clearly defined learning outcomes. optional section for those who go faster or want to do more. WAY you learn how to design digital circuits using a hardware age. You will not really learn through lectures and tutorials alone!

VERI is organised in four sequential parts, each build upon the previous parts. Each part has been designed with very clear learning outcomes in mind, and is intended to take a 3-hour supervised laboratory session.

You need to download various files from the Experiment website in order to do this experiment. These files can be found on:

http://www.ee.ic.ac.uk/pcheung/teaching/E2\_experiment/





You MUST use Quartus 16 (known as Quartus Prime) standard edition and NOT Quartus 13 (known as Quartus II) because version 13 does not support Cyclone V FPGA chips. I recommend that you create a shortcut on your desktop for convenience.

Once you started Quartus software, you will eventually see many window panes appearing in the Quartus window:

**Edit window** – You should use this to edit all your source files. The editor is basic, but it is syntax sensitive, so it will highlight Verilog keywords for your.

Message window – This is where you find all the error and warning messages.

**Navigation window** – This shows the hierarchy of your design and provides a quick way of exploring various Verilog files.

**Status window** – This tells you the steps that the Quartus software is taking in order to produce the final design.

**Summary window** – This provides a quick summary of the resources being used by your design – useful to check for overall errors.

**Report window** – This is where you find out details of the compilation results such as timing and pin allocations.

**IP catalog** – This allows you to pick up modules in the component library provided by Altera, such as ROM, multiplier, FIFO etc.

You must also make sure that you have specified the exact FPGA device used on the DE1-SoC board. It is 5CSEMA5F31C6 and it is specified using **> Assignments > Device ....** 



Once you have finished creating your design through hardware compilation, we need to send the bit-stream to the chip via the USB cable. The method of programming is specified in Hardware Setup, and we specify DE1-SoC [USB].

Next we use "Auto Detect" to find out what FPGA chip is connected, and specify that we expect to find the 5CSEMA5 chip family. This is one of many different variants of Cyclone V FPGAs. Each has a different protocol in programming the device.

Then, we have to delete the part of the chip that we are not using – this the ARM part, known as SOCVHPS. If you were to use the ARM processor (e.g. loading it with Linux), we would need this line in.

After that, you must select the "sof" file to send the chip.

Extension	What is it?
.bdf	Block Design File - schematic diagram
.bsf	Block Symbol File - component symbol
.cdf	Chain Description File - ignore this
.do	DO file - Testbench in Modelsim
.mif	Memory Init File - Contents of ROM
.qpf	Quartus Project File - Specify project
.qsf	Quartus Setting File - Modules and pin assignments
.rpt	Report File - text file reporting on various things
.sof	SRAM Oject File - Bitstream file to program FPGA
.v	Verilog source file - your design
uartus uses an nallest design. ere are some o ovided here for	d generates many different files associated with even then of the more commonly used file types and what they mean. It is r your reference and convenience.
	Extension .bdf .bsf .cdf .do .mif .qpf .qsf .rpt .sof .v uartus uses an ballest design. ere are some of poided here for

You can find a full list of file types used by Quartus on the Experiment webpage.





The main goal of Part 1 is to let you get familiar with the entire design process.

Shown here is a summary of all the main steps that you have to go through to create a working design to send to the DE1-SoC.



Part 2 of the Lab is starting to get harder. You will first learn to use Modelsim to simulate an 8-bit counter. On the way, you will also learn how to combine the interactive commands your type in the command window in Modelsim into a DO-file. You then will use this file as the testbench for your circuit.

Next you will extend the 8-bit counter to 16-bit with reset and enable input. You will also combine this with what you have already done in Part 1 to display the counter value as decimal number on the displays.

Next, you will find that the counter is counting too fast – you only see 88888 at the output. You then will add another circuit known as **prescaler**, which is another counter that produces a clock tick once every millisecond. This allows you to see what you see the count value changing.

Then you will create a random number generator and test this on DE1.



Optional challenge is quite hard, but should be very satisfying. Do this only if you have time.



Part 3 and Part 4 of VERI use the analogue I/O card with DE1. The I/O card contains a DAC and a ADC, both 10-bits. These produce an audio output on the right channel of a 3.5mm socket, and one channel of the analogue input from the other socket.

There is also a low-pass filter, which receives a PWM signal and produces an analogue output on the left channel of the audio socket. A 5k ohm potentiometer provides a dc voltage to the other channel of the ADC.

The I/O card is plugged into the 40-way socket on the DE1 board – the socket is the one that is furthest away from the board edge. Beware that you may not have aligned the pins correctly. This will no damage anything. If the I/O board is installed correctly, the GREEN LED will light up when the DE1 board is turned ON.

The communication between the DAC/DAC and the FPGA chip is through serial interface known as SPI (Serial Peripheral Interface). How exactly SPI works will be covered in another lecture later.



Part 3 will introduce you to many different useful digital components. This includes: the SPI interface module spi2dac.v, the PWM module pwm.v, the ROM generator, the multiplier etc. In the end, you will be able to produce at least a fixed frequency sinewave on both channel of the output socket.

## Displaying a binary number as decimal



We now take another example of a relative complex combinational circuit, and see how we can specify our design in Verilog.

The goal is to design a circuit that converts an 8-bit binary number into three x 4-bit binary coded decimal values (i.e. 12 bit).

There is a well-known algorithm called "**shift-and-add-3**" algorithm to do this conversion. For example, if we take 8-bit hexadecimal number 8'hff (i.e. all 1's), it has two hex digits. Once converted to binary coded decimal (BCD) it becomes 255 (3 BCD digits).



Before we examine this algorithm in detail, let us consider the arithmetic operation of shifting left by one bit. This is the same as a  $\times 2$  operation.

If we do it 8 times, then we have multiplied the original number by 256 or  $2^8$ .

Now if you ignore the bottom 8-bit through a truncation process, you effectively divide the number by 256. In other words, we get back to the original number in binary (or in hexadecimal).



Our conversion algorithms works by shift the number left 8 times, but each time make an adjustment (or correction) if it is NOT a valid BCD digit.

Let us consider this example. We can shift the number four time left, and it will give a valid BCD digit of 7.

However, if we shift left again, then 7 becomes hex F, which is NOT valid. Therefore the algorithm demands that 3 is added to 7 (7 is larger or equal to 5) before we do the shift.



The rationale of this algorithm is the following. If the number is 5 or larger, after shift left, we will get 10 or larger, which cannot fit into a BCD digit. Therefore if the number 5 (or larger) we add 3 to it (after shifting is adding 6), which measure we carry forward a 1 to the next BCD digit.



To recap: the basic idea is to shift the binary number left, one bit at a time, into locations reserved for the BCD results. Let us take the example of the binary number 8'h7C. This is being shifted into a 12-bit/3 digital BCD result as shown above.

After 8 shift operations, the three BCD digits contain respectively: hundredth digit = 4'b0001, tens digit = 4'b0010 and ones digit = 4'b0100, thus representing the BCD value of 124.

The key idea behind the algorithm can be understood as follow (see the diagram in the slide):

1.Each time the number is shifted left, it is multiplied by 2 as it is shifted to the BCD locations;

2. The values in the BCD digits are the same as as binary if its value is 9 or lower. However if it is 10 or above it is not correct because for BCD, this should carry over to the next digit. A correction must be made by adding 6 to this digit value.

3. The easiest way to do this is to detect if the value in the BCD digit locations are 5 or above BEFORE the shift (i.e. X2). If it is  $\geq$ 5, then add 3 to the value (i.e. adjust by +6 after the shift).



In order to understand how to we may implement this converter in hardware, you have to understand that shifting in hardware is easy. You just need to connect signals with one bit shift to the left. It DOES NOT need any gates, just wires!

Now we also need to do the adjust module, which simply performs the operation:

if  $(in \ge 5)$  out = in + 3 else out = in

The easiest way to implement such a module is to use a **case statement**. This is set as a tutorial problem in Problem Sheet 1.



The entire full array is shown here. The shade module is the adjust module (which we call: **add3\_ge5**).

As I said in the last slide, the easiest way to implement (specify) **add3\_ge5** is using a case statement.

The BLUE signal path traces what happens to the least significant bit of the original number.



The full array is more complicated than need be. If we propagate the '0's forward in the array of gates, you will find those marked with 'X' will always have its input less than 5. In which, **output = input** in these modules. THIS IS JUST A SET OF FOUR WIRES.

The only remaining **add3\_ge5** modules are those shaped in orange.



After simplification, here are ALL the remaining **add3\_ge5** modules for the 8-bit binary to BCD conversion (bin2bcd8). I have labeled the input ports to **add3\_ge3 wn[3:0]** and the output parts **an[3:0]** where n is 1 to 7.



Assuming that we have designed a module "**add3\_ge5**" to perform the adjustment as required, the converter can be implemented in Verilog by simply "WIRING UP" the various modules together.

The interconnections are specified in the wire statements.

The next block is instantiating 7 add3\_ge5 modules.

The next block of code is to wire the modules together.

Finally the last statements are to connect up the signals from the modules to the output ports.



The final part (Part 4) really brings everything together. The goal is to use the FPGA to implement a real-time speech processing system that perform echo simulation.

To start with, you will implement a DO-NOTHING block. This just tests out the system and takes an analogue input sample, then output it to the earphone. Nevertheless there are details that need to be taken care of. I will go into more details nearer the time in order to explain exactly what's happening.

Finally, the part includes something that CANNOT be done in four lab session, but if you are really keen in going further it is a real challenge.

Bringing everything you have learned in VERI, you can design a voice corruptor – some that that makes changes ones speech in a way that is unrecognizable, but it is still comply understandable.