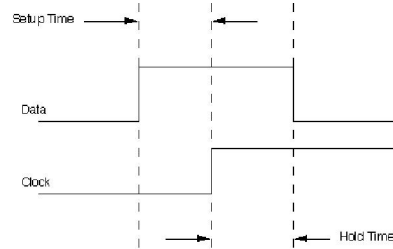


- 1) Explain about setup time and hold time, what will happen if there is setup time and hold time violation, how to overcome this?

For Synchronous flip-flops, we have special requirements for the inputs with respect to clock signal input there are
Setup Time: Minimum time Period during which data must be stable before the clock makes a valid transition. E.g. for a positive edge triggered flip-flop having a setup time of 2ns so input data should be Stable for 2ns before the clock makes a valid transaction from zero to one

Hold Time: Minimum time period during which data must be stable after the clock has made a valid transition. E.g. for a posedge triggered flip-flop, with a hold time of 1 ns. Input Data (i.e. R and S in the case of RS flip-flop) should be stable for at least 1 ns after clock has made transition from 0 to 1

Hold time is the amount of time after the clock edge that same input signal has to be held before changing it to make sure it is sensed properly at the clock edge. Whenever there are setup and hold time violations in any flip-flop, it enters a state where its output is unpredictable: this state is known as metastable state (quasi stable state); at the end of metastable state, the flip-flop settles down to either '1' or '0'. This whole process is known as metastability



- 2) What is difference between latch and flip-flop?

The main difference between latch and FF is that latches are level sensitive while FF is edge sensitive. They both require the use of clock signal and are used in sequential logic. For a latch, the output tracks the input when the clock signal is high, so as long as the clock is logic 1, the output can change if the input also changes. FF on the other hand, will store the input only when there is a rising/falling edge of the clock. Latches are sensitive to glitches on enable pin, whereas flip-flop is immune to glitches. Latches take fewer gates (also less power) to implement than flip-flops. Latches are faster than flip-flops

- 3) Given only two xor gates one must function as buffer and another as inverter?

Tie one of xor gates input to 1 it will act as inverter.

Tie one of xor gates input to 0 it will act as buffer.

- 4) Difference between Mealy and Moore state machine?

A) Mealy and Moore models are the basic models of state machines. A state machine which uses only Entry Actions, so that its output depends on the state, is called a Moore model. A state machine which uses only Input Actions, so that the output depends on the state and also on inputs, is called a Mealy model. The models selected will influence a design but there are no general indications as to which model is better. Choice of a model depends on the application, execution means (for instance, hardware systems are usually best realized as Moore models) and personal preferences of a designer or programmer

B) Mealy machine has outputs that depend on the state and input (thus, the FSM has the output written on edges)

Moore machine has outputs that depend on state only (thus, the FSM has the output written in the state itself).

Advantage and Disadvantage

In Mealy as the output variable is a function both input and state, changes of state of the state variables will be delayed with respect to changes of signal level in the input variables, there are possibilities of glitches appearing in the output variables. Moore overcomes glitches as output dependent on only states and not the input signal level.

All of the concepts can be applied to Moore-model state machines because any Moore state machine can be implemented as a Mealy state machine, although the converse is not true.

Moore machine: the outputs are properties of states themselves... which means that you get the output after the machine reaches a particular state, or to get some output your machine has to be taken to a state which provides you the output. The outputs are held until you go to some other state Mealy machine:

Mealy machines give you outputs instantly, that is immediately upon receiving input, but the output is not held after that clock cycle.

- 5) Difference between one hot and binary encoding?

Common classifications used to describe the state encoding of an FSM are Binary (or highly encoded) and One hot.

A binary-encoded FSM design only requires as many flip-flops as are needed to uniquely encode the number of states in the state machine. The actual number of flip-flops required is equal to the ceiling of the log-base-2 of the number of states in the FSM. A one hot FSM design requires a flip-flop for each state in the design and only one flip-flop (the flip-flop representing the current or "hot" state) is set at a time in a one hot FSM design. For a state machine with 9- 16 states, a binary FSM only requires 4 flip-flops while a one hot FSM requires a flip-flop for each state in the design

FPGA vendors frequently recommend using a one hot state encoding style because flip-flops are plentiful in an FPGA and the combinational logic required to implement a one hot FSM design is typically smaller than most binary encoding styles. Since FPGA performance is typically related to the combinational logic size of the FPGA design, one hot FSMs typically run faster than a binary encoded FSM with larger combinational logic blocks

- 6) How to achieve 180 degree exact phase shift?

Never tell using inverter

a) DCM an inbuilt resource in most of FPGA can be configured to get 180 degree phase shift.

b) BUFGDS that is differential signaling buffers which are also inbuilt resource of most of FPGA can be used.

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- 7) **What is significance of RAS and CAS in SDRAM?**
SDRAM receives its address command in two address words. It uses a multiplex scheme to save input pins. The first address word is latched into the DRAM chip with the row address strobe (RAS). Following the RAS command is the column address strobe (CAS) for latching the second address word. Shortly after the RAS and CAS strobes, the stored data is valid for reading.
- 8) **Tell some of applications of buffer?**
a) They are used to introduce small delays.
b) They are used to eliminate cross talk caused due to inter electrode capacitance due to close routing.
c) They are used to support high fan-out, e.g.: bufg
- 9) **Give two ways of converting a two input NAND gate to an inverter?**
a) Short the 2 inputs of the nand gate and apply the single input to it.
b) Connect the output to one of the input and the other to the input signal.
- 10) **Why is most interrupts active low?**
This answers why most signals are active low
if you consider the transistor level of a module, active low means the capacitor in the output terminal gets charged or discharged based on low to high and high to low transition respectively. When it goes from high to low it depends on the pull down resistor that pulls it down and it is relatively easy for the output capacitance to discharge rather than charging. Hence people prefer using active low signals.
- 11) **Design a four-input NAND gate using only two-input NAND gates.**
Basically, you can tie the inputs of a NAND gate together to get an inverter.
- 12) **What will happen if contents of register are shifter left, right?**
It is well known that in left shift all bits will be shifted left and LSB will be appended with 0 and in right shift all bits will be shifted right and MSB will be appended with 0 this is a straightforward answer
What is expected is in a left shift value gets Multiplied by 2 e.g.: consider 0000_1110=14 a left shift will make it 0001_110=28, it the same fashion right shift will Divide the value by 2.
- 13) **Given the following FIFO and rules, how deep does the FIFO need to be to prevent underflow or overflow?**
RULES:
1) $\text{frequency}(\text{clk_A}) = \text{frequency}(\text{clk_B}) / 4$
2) $\text{period}(\text{en_B}) = \text{period}(\text{clk_A}) * 100$
3) $\text{duty cycle}(\text{en_B}) = 25\%$
Assume $\text{clk_B} = 100\text{MHz}$ (10ns)
From (1), $\text{clk_A} = 25\text{MHz}$ (40ns)
From (2), $\text{period}(\text{en_B}) = 40\text{ns} * 100 = 4000\text{ns}$, but we only output for 1000ns, due to (3), so 3000ns of the enable we are doing no output work. Therefore, FIFO size = $3000\text{ns}/40\text{ns} = 75$ entries
- 14) **Differences between D-Latch and D flip-flop?**
D-latch is level sensitive where as flip-flop is edge sensitive. Flip-flops are made up of latches.
- 15) **What is a multiplexer?**
Is a combinational circuit that selects binary information from one of many input lines and directs it to a single output line. ($2^n \Rightarrow n$). Where n is selection line.
- 16) **What are set up time & hold time constraints? What do they signify? Which one is critical for estimating maximum clock frequency of a circuit?**
Set up time is the amount of time the data should be stable before the application of the clock signal, where as the hold time is the amount of time the data should be stable after the application of the clock. Setup time signifies maximum delay constraints; hold time is for minimum delay constraints. Setup time is critical for establishing the maximum clock frequency.
- 17) **How can you convert an SR Flip-flop to a JK Flip-flop?**
By giving the feedback we can convert, i.e. $Q \Rightarrow S$ and $Q' \Rightarrow R$. Hence the S and R inputs will act as J and K respectively.
- 18) **How can you convert the JK Flip-flop to a D Flip-flop?**
By connecting the J input to the K through the inverter.
- 19) **How do you detect if two 8-bit signals are same?**
XOR each bits of A with B (for e.g. A [0] xor B [0]) and so on. The o/p of 8 xor gates is then given as i/p to an 8-i/p nor gate. if o/p is 1 then A=B.
- 20) **Convert D-FF into divide by 2. (not latch) What is the max clock frequency the circuit can handle, given the following information?**
 $T_{\text{setup}} = 6\text{ns}$ $T_{\text{hold}} = 2\text{ns}$ $T_{\text{propagation}} = 10\text{ns}$
Circuit: Connect Qbar to D and apply the clk at clk of DFF and take the O/P at Q. It gives freq/2. Max. Freq of operation: $1/(\text{propagation delay} + \text{setup time}) = 1/16\text{ns} = 62.5\text{ MHz}$
- 21) **7 bit ring counter's initial state is 0100010. After how many clock cycles will it return to the initial state?**
6 cycles
- 22) **Design all the gates (NOT, AND, OR, NAND, NOR, XOR, XNOR) using 2:1 Multiplexer?**
Using 2:1 Mux, (2 inputs, 1 output and a select line)
a) NOT: Give the input at the select line and connect I0 to 1 & I1 to 0. So if A is 1, we will get I1 that is 0 at the O/P.
b) AND: Give input A at the select line and 0 to I0 and B to I1. O/p is A & B
c) OR: Give input A at the select line and 1 to I1 and B to I0. O/p will be A | B
d) NAND: AND + NOT implementations together
e) NOR: OR + NOT implementations together
f) XOR: A at the select line B at I0 and ~B at I1. ~B can be obtained from (a)
g) XNOR: A at the select line B at I1 and ~B at I0

23) Design a circuit that calculates the square of a number?

It should not use any multiplier circuits. It should use Multiplexers and other logic?

$$1^2=0+1=1$$

$$2^2=1+3=4$$

$$3^2=4+5=9$$

$$4^2=9+7=16$$

$$5^2=16+9=25$$

See a pattern yet? To get the next square, all you have to do is add the next odd number to the previous square that you found. See how 1,3,5,7 and finally 9 are added. Wouldn't this be a possible solution to your question since it only will use a counter, multiplexer and a couple of adders? It seems it would take n clock cycles to calculate square of n.

24) N number of XNOR gates is connected in series such that the N inputs (A0, A1, A2.....) are given in the following way: A0 & A1 to first XNOR gate and A2 & O/P of First XNOR to second XNOR gate and so on..... Nth XNOR gates output is final output. How does this circuit work? Explain in detail?

If N=Odd, the circuit acts as even parity detector, i.e. the output will 1 if there are even number of 1's in the N input...This could also be called as odd parity generator since with this additional 1 as output the total number of 1's will be ODD. If N=Even, just the opposite, it will be Odd parity detector or Even Parity Generator.

25) What is Race-around problem? How can you rectify it?

The clock pulse that remains in the 1 state while both J and K are equal to 1 will cause the output to complement again and repeat complementing until the pulse goes back to 0, this is called the race around problem. To avoid this undesirable operation, the clock pulse must have a time duration that is shorter than the propagation delay time of the F-F, this is restrictive so the alternative is master-slave or edge-triggered construction.

26) An assembly line has 3 fail safe sensors and one emergency shutdown switch. The line should keep moving unless any of the following conditions arise:

- (i) If the emergency switch is pressed
- (ii) If the sensor1 and sensor2 are activated at the same time.
- (iii) If sensor 2 and sensor3 are activated at the same time.
- (iv) If all the sensors are activated at the same time

suppose a combinational circuit for above case is to be implemented only with NAND Gates. How many minimum number of 2 input NAND gates are required?

No of 2-input NAND Gates required = 6 you can try the whole implementation.

27) How will you implement a Full subtractor from a Full adder?

All the bits of subtrahend should be connected to the xor gate. Other input to the xor being one. The input carry bit to the full adder should be made 1. Then the full adder works like a full subtract

28) What is difference between setup and hold time. The interviewer was looking for one specific reason, and its really a good answer too..The hint is hold time doesn't depend on clock, why is it so...?

Setup violations are related to two edges of clock, i mean you can vary the clock frequency to correct setup violation. But for hold time, you are only concerned with one edge and do not basically depend on clock frequency.

29) In a 3-bit Johnson's counter what are the unused states?

$2(\text{power } n)-2n$ is the one used to find the unused states in Johnson counter.

So for a 3-bit counter it is $8-6=2$. Unused states=2. the two unused states are 010 and 101

30) What is difference between RAM and FIFO?

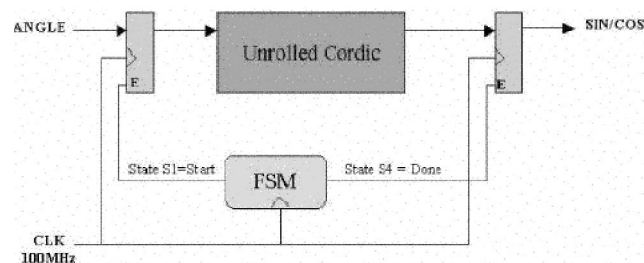
FIFO does not have address lines

Ram is used for storage purpose where as FIFO is used for synchronization purpose i.e. when two peripherals are working in different clock domains then we will go for FIFO.

31) What are multi-cycle paths?

Multi-cycle paths are paths between registers that take more than one clock cycle to become stable.

For ex. analyzing the design shown in fig below shows that the output SIN/COS requires 4 clock-cycles after the input ANGLE is latched in. This means that the combinatorial block (the Unrolled Cordic) can take up to 4 clock periods (25MHz) to propagate its result. Place and Route tools are capable of fixing multi-cycle paths problem.



32) Consider two similar processors, one with a clock skew of 100ps and other with a clock skew of 50ps. Which one is likely to have more power? Why?

Clock skew of 50ps is more likely to have clock power. This is because it is likely that low-skew processor has better designed clock tree with more powerful and number of buffers and overheads to make skew better.

33) Is it possible to reduce clock skew to zero? Explain your answer?

Even though there are clock layout strategies (H-tree) that can in theory reduce clock skew to zero by having the same path length from each flip-flop from the pll, process variations in R and C across the chip will cause clock skew as well as a pure H-Tree scheme is not practical (consumes too much area).

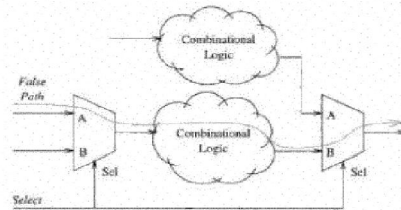
Digital Electronics Solved Questions

- 34) The circle can rotate clockwise and back. Use minimum hardware to build a circuit to indicate the direction of rotating?

2 sensors are required to find out the direction of rotating. They are placed like at the drawing. One of the m is connected to the data input of D flip-flop, and a second one - to the clock input. If the circle rotates the way clock sensor sees the light first while D input (second sensor) is zero - the output of the flip-flop equals zero, and if D input sensor "fires" first - the output of the flip-flop becomes high.

- 35) What is false path? How it determine in ckt? What the effect of false path in ckt?

By timing all the paths in the circuit the timing analyzer can determine all the critical paths in the circuit. However, the circuit may have false paths, which are the paths in the circuit which are never exercised during normal circuit operation for any set of inputs.



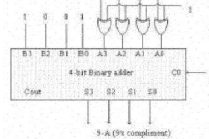
An example of a false path is shown in figure below. The path going from the input A of the first MUX through the combinational logic out through the B input of the second MUX is a false path. This path can never be activated since if the A input of the first MUX is activated, then Sel line will also select the A input of the second MUX.

STA (Static Timing Analysis) tools are able to identify simple false paths; however they are not able to identify all the false paths and sometimes report false paths as critical paths. Removal of false paths makes circuit testable and its timing performance predictable (sometimes faster)

- 36) You have two counters counting upto 16, built from negedge DFF , First circuit is synchronous and second is "ripple" (cascading), Which circuit has a less propagation delay? Why?

The synchronous counter will have lesser delay as the input to each flop is readily available before the clock edge. Whereas the cascade counter will take long time as the output of one flop is used as clock to the other. So the delay will be propagating. For E.g.: 16 state counter = 4 bit counter = 4 Flip flops Let 10ns be the delay of each flop The worst case delay of ripple counter = $10 \times 4 = 40\text{ns}$ The delay of synchronous counter = 10ns only.(Delay of 1 flop)

- 37) Design a circuit for finding the 9's compliment of a BCD number using 4-bit binary adder and some external logic gates?



9's compliment is nothing but subtracting the given no from 9. So using a 4 bit binary adder we can just subtract the given binary no from 1001 (i.e. 9). Here we can use the 2's complement method addition.

- 38) Difference between Synchronous and Asynchronous reset?

Synchronous reset logic will synthesize to smaller flip-flops, particularly if the reset is gated with the logic generating the d-input. But in such a case, the combinational logic gate count grows, so the overall gate count savings may not be that significant. The clock works as a filter for small reset glitches; however, if these glitches occur near the active clock edge, the Flip-flop could go metastable. In some designs, the reset must be generated by a set of internal conditions. A synchronous reset is recommended for these types of designs because it will filter the logic equation glitches between clocks.

Disadvantages of synchronous reset:

Problem with synchronous resets is that the synthesis tool cannot easily distinguish the reset signal from any other data signal. Synchronous resets may need a pulse stretcher to guarantee a reset pulse width wide enough to ensure reset is present during an active edge of the clock. If you have a gated clock to save power, the clock may be disabled coincident with the assertion of reset. Only an asynchronous reset will work in this situation, as the reset might be removed prior to the resumption of the clock. Designs that are pushing the limit for data path timing, cannot afford to have added gates and additional net delays in the data path due to logic inserted to handle synchronous resets.

Asynchronous reset:

The biggest problem with asynchronous resets is the reset release, also called reset removal. Using an asynchronous reset, the designer is guaranteed not to have the reset added to the data path. Another advantage favoring asynchronous resets is that the circuit can be reset with or without a clock present.

Disadvantages of asynchronous reset: ensure that the release of the reset can occur within one clock period. If the release of the reset occurred on or near a clock edge such that the flip-flops went metastable.

- 39) Implement the following circuits:

- 3 input NAND gate using min no of 2 input NAND Gates
- 3 input NOR gate using min no of 2 input NOR Gates
- 3 input XNOR gate using min no of 2 input XNOR Gates

Assuming 3 inputs A,B,C?

3 input NAND Connect:

- A and B to the first NAND gate

b) Output of first Nand gate is given to the two inputs of the second NAND gate (this basically realizes the inverter functionality)

- Output of second NAND gate is given to the input of the third NAND gate, whose other input is C

((A NAND B) NAND (A NAND B)) NAND C Thus, can be implemented using '3' 2-input NAND gates. I guess this is the minimum number of gates that need to be used.

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3 input NOR: Same as above just interchange NAND with NOR ((A NOR B) NOR (A NOR B)) NOR C

3 input XNOR: Same as above except the inputs for the second XNOR gate, Output of the first XNOR gate is one of the inputs and connect the second input to ground or logical '0'

((A XNOR B) XNOR 0)) XNOR C

40) What is slack?

'Slack' is the amount of time you have that is measured from when an event 'actually happens' and when it 'must happen'. The term 'actually happens' can also be taken as being a predicted time for when the event will 'actually happen'. When something 'must happen' can also be called a 'deadline' so another definition of slack would be the time from when something 'actually happens' (call this Tact) until the deadline (call this Tdead).

Slack = Tdead - Tact.

Negative slack implies that the 'actually happen' time is later than the 'deadline' time...in other words it's too late and a timing violation....you have a timing problem that needs some attention.

41) Difference between heap and stack?

The Stack is more or less responsible for keeping track of what's executing in our code (or what's been "called"). The Heap is more or less responsible for keeping track of our objects (our data, well... most of it - we'll get to that later.).

Think of the Stack as a series of boxes stacked one on top of the next. We keep track of what's going on in our application by stacking another box on top every time we call a method (called a Frame). We can only use what's in the top box on the stack. When we're done with the top box (the method is done executing) we throw it away and proceed to use the stuff in the previous box on the top of the stack. The Heap is similar except that its purpose is to hold information (not keep track of execution most of the time) so anything in our Heap can be accessed at any time. With the Heap, there are no constraints as to what can be accessed like in the stack. The Heap is like the heap of clean laundry on our bed that we have not taken the time to put away yet - we can grab what we need quickly. The Stack is like the stack of shoe boxes in the closet where we have to take off the top one to get to the one underneath it.

42) What is Difference between write back and write through cache?

A caching method in which modifications to data in the cache aren't copied to the cache source until absolutely necessary. Write-back caching is available on many microprocessors, including all Intel processors since the 80486. With these microprocessors, data modifications to data stored in the L1 cache aren't copied to main memory until absolutely necessary. In contrast, a write-through cache performs all write operations in parallel -- data is written to main memory and the L1 cache simultaneously. Write-back caching yields somewhat better performance than write-through caching because it reduces the number of write operations to main memory. With this performance improvement comes a slight risk that data may be lost if the system crashes.

A write-back cache is also called a copy-back cache.

43) What is skew, what are problems associated with it and how to minimize it?

In circuit design, clock skew is a phenomenon in synchronous circuits in which the clock signal (sent from the clock circuit) arrives at different components at different times.

This is typically due to two causes. The first is a material flaw, which causes a signal to travel faster or slower than expected. The second is distance: if the signal has to travel the entire length of a circuit, it will likely (depending on the circuit's size) arrive at different parts of the circuit at different times. Clock skew can cause harm in two ways. Suppose that a logic path travels through combinational logic from a source flip-flop to a destination flip-flop. If the destination flip-flop receives the clock tick later than the source flip-flop, and if the logic path delay is short enough, then the data signal might arrive at the destination flip-flop before the clock tick, destroying there the previous data that should have been clocked through. This is called a hold violation because the previous data is not held long enough at the destination flip-flop to be properly clocked through. If the destination flip-flop receives the clock tick earlier than the source flip-flop, then the data signal has that much less time to reach the destination flip-flop before the next clock tick. If it fails to do so, a setup violation occurs, so-called because the new data was not set up and stable before the next clock tick arrived. A hold violation is more serious than a setup violation because it cannot be fixed by increasing the clock period.

Clock skew, if done right, can also benefit a circuit. It can be intentionally introduced to decrease the clock period at which the circuit will operate correctly, and/or to increase the setup or hold safety margins. The optimal set of clock delays is determined by a linear program, in which a setup and a hold constraint appear for each logic path. In this linear program, zero clocks skew is merely a feasible point.

Clock skew can be minimized by proper routing of clock signal (clock distribution tree) or putting variable delay buffer so that all clock inputs arrive at the same time

44) Difference between Synchronous, Asynchronous & I synchronous communication?

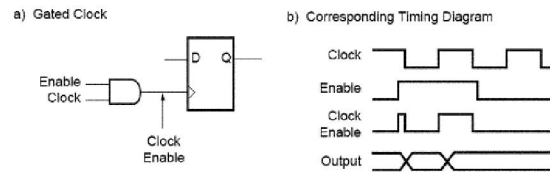
Sending data encoded into your signal requires that the sender and receiver are both using the same encoding/decoding method, and know where to look in the signal to find data. Asynchronous systems do not send separate information to indicate the encoding or clocking information. The receiver must decide the clocking of the signal on its own. This means that the receiver must decide where to look in the signal stream to find ones and zeroes, and decide for itself where each individual bit stops and starts. This information is not in the data in the signal sent from transmitting unit.

Synchronous systems negotiate the connection at the data-link level before communication begins. Basic synchronous systems will synchronize two clocks before transmission, and reset their numeric counters for errors etc. More advanced systems may negotiate things like error correction and compression.

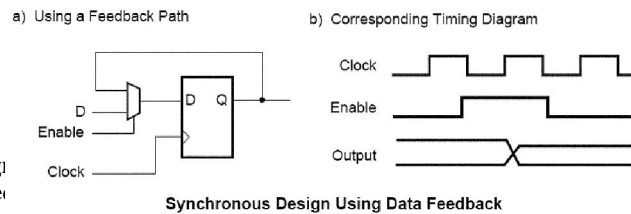
Time-dependent. it refers to processes where data must be delivered within certain time constraints. For example, Multimedia stream require an isochronous transport mechanism to ensure that data is delivered as fast as it is displayed and to ensure that the audio is synchronized with the video.

45) What is glitch? What causes it (explain with waveform)? How to overcome it?

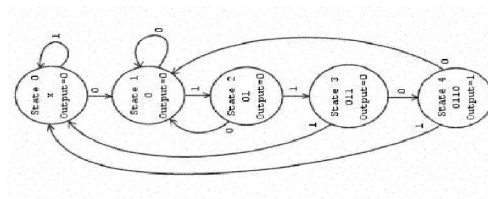
The gated clock 's corresponding timing diagram shows that this implementation can lead to clock glitches ,Which can cause flip-flop to clock at the wrong time



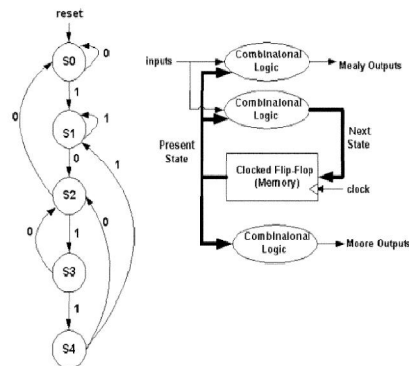
The following figure shows a synchronous alternative to the gated clock using a data path. The flip-flop is clocked at every clock cycle and the data path is controlled by an enable. When the enable is Low, the multiplexer feeds the output of the register back on itself. When the enable is High, new data is fed to the flip-flop and the register changes its state



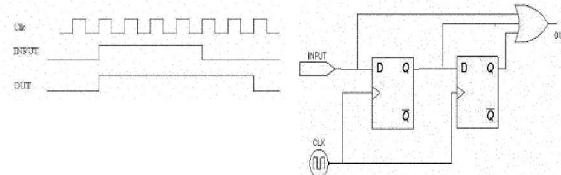
46) Draw the state diagram in more than one set



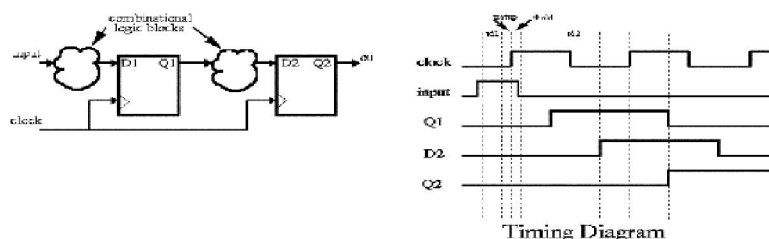
47) Design a FSM (Finite State Machine) to detect a sequence 10110?



48) Give the circuit to extend the falling edge of the input by 2 clock pulses?

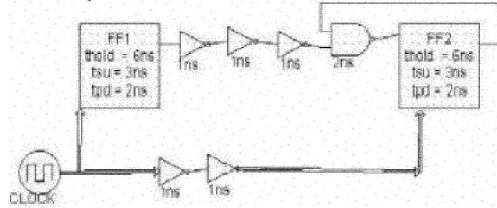


49) Draw timing diagrams for following circuit?



Digital Electronics Solved Questions

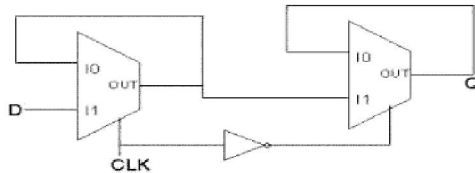
- 50) For the Circuit Shown below, what is the Maximum Frequency of Operation? Are there any hold time violations for FF2? If yes, how do you modify the circuit to avoid them?



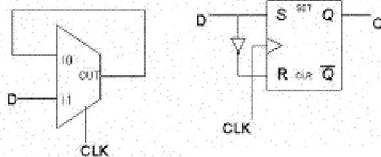
The minimum time period = $3 + 2 + (1 + 1 + 1) = 8\text{ns}$ Maximum Frequency = $1/8\text{ns} = 125\text{MHz}$.

And there is a hold time violation in the circuit, because of feedback, if you observe, $t_{cq2} + \text{AND gate delay}$ is less than t_{hold2} . To avoid this we need to use even number of inverters (buffers). Here we need to use 2 inverters each with a delay of 1ns. Then the hold time value exactly meets.

- 51) How to implement a Master Slave flip flop using a 2 to 1 Mux?



- 52) Design a D-latch using (a) using 2:1 Mux (b) from S-R Latch?



- 53) What are different ways multiply & Divide?

Binary Division by Repeated Subtraction

- § Set quotient to zero
- § Repeat while dividend is greater than or equal to divisor
- § Subtract divisor from dividend
- § Add 1 to quotient
- § End of repeat block
- § Quotient is correct, dividend is remainder
- § STOP

Binary Division by Shift and Subtract

- § Basically the reverse of the multiply by shift and add.
- § **Set** quotient to 0
- § Align leftmost digits in dividend and divisor
- § **Repeat**
- § **If** that portion of the dividend above the divisor is greater than or equal to the divisor
- § **Then** subtract divisor from that portion of the dividend and
- § Concatenate 1 to the right hand end of the quotient
- § **Else** concatenate 0 to the right hand end of the quotient
- § Shift the divisor one place right
- § **Until** dividend is less than the divisor
- § quotient is correct, dividend is remainder
- § **STOP**

Binary Multiply - Repeated Shift and Add

Repeated shift and add - starting with a result of 0, shift the second multiplicand to correspond with each 1 in the first multiplicand and add to the result. Shifting each position left is equivalent to multiplying by 2, just as in decimal representation a shift left is equivalent to multiplying by 10

- § Set result to 0
- § Repeat
- § Shift Second Multiplicand left until rightmost digit is lined up with leftmost 1 in first multiplicand
- § Add 2nd multiplicand in that position to result
- § Remove that 1 from 1st multiplicand
- § Until 1st multiplicand is zero
- § Result is correct
- § STOP

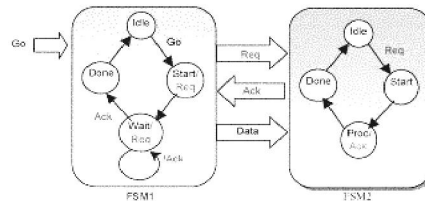
Digital Electronics Solved Questions

54) What are the different ways synchronize between two clock domains?

The following section explains clock domain interfacing

one of the biggest challenges of system-on-chip (SOC) designs is that different blocks operate on independent clocks.

Integrating these blocks via the processor bus, memory ports, peripheral busses, and other interfaces can be troublesome because unpredictable behavior can result when the asynchronous interfaces are not properly synchronized. A very common and robust method for synchronizing multiple data signals is a handshake technique as shown in diagram below. This is popular because the handshake technique can easily manage changes in clock frequencies, while minimizing latency at the crossing. However, handshake logic is significantly more complex than standard synchronization structures.



FSM1 (Transmitter) asserts

the req (request) signal, asking the receiver to accept the data on the data bus. FSM2 (Receiver) generally a slow module asserts the ack (acknowledge) signal, signifying that it has accepted the data. it has loop holes: when system Receiver samples the systems Transmitter req line and Transmitter samples system Receiver ack line, they have done it with respect to their internal clock, so there will be setup and hold time violation. To avoid this we go for double or triple stage synchronizers, which increase the MTBF and thus are immune to metastability to a good extent. The figure below shows how this is done.

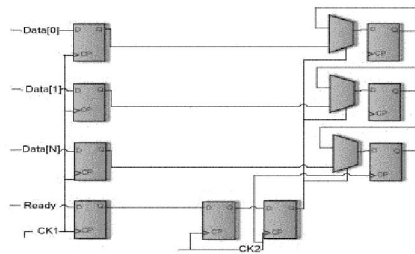


Figure 1b — Multi-bit sync

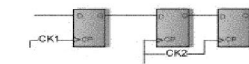


Figure 1a — Single-bit metastability sync

55) What is FIFO? How to Calculate the Depth of FIFO?

One of the most common questions in interviews is how to calculate the depth of a FIFO. FIFO is used as buffering element or queuing element in the system, which is by common sense, is required only when you slow at reading than the write operation. So size of the FIFO basically implies the amount of data required to buffer, which depends upon data rate at which data is written and the data rate at which data is read. Statistically, Data rate varies in the system majorly depending upon the load in the system. So to obtain safer FIFO size we need to consider the worst case scenario for the data transfer across the FIFO under consideration. For worst case scenario, Difference between the data rate between write and read should be maximum. Hence, for write operation maximum data rate should be considered and for read operation minimum data rate should be considered. So in the question itself, data rate of read operation is specified by the number of idle cycles and for write operation, maximum data rate should be considered with no idle cycle.

So for write operation, we need to know Data rate = Number of data * rate of clock.

Writing side is the source and reading side becomes sinking, data rate of reading side depends upon the writing side data rate and its own reading rate which is Frd/Idle_cycle_rd. In order to know the data rate of write operation, we need to know Number of data in a Burst which we have assumed to be B. So following up with the equation as explained below: Fifo size = Size to be buffered = $B - B * Frd / (Fwr * Idle_cycle_rd_rd)$.

Here we have not considered the synchronizing latency if Write and Read clocks are Asynchronous. Greater the Synchronizing latency, higher the FIFO size requirement to buffer more additional data written.

FIFO Depth Calculation:

Assume that we have to design a FIFO with following requirements and we want to calculate minimum FIFO depth

- A synchronized FIFO
- Writing clock 30MHz - F1
- Reading clock 40MHz - F2
- Writing Burst Size - B
- Case 1 : There is 1 idle clock cycle for reading side - I
- Case 2 : There is 10 idle clock cycle for reading side - I

FIFO depth calculation = $B - B * F2 / (F1 * I)$

If we have alternate read cycles i.e. between two read cycles there is IDLE cycle.

FIFO depth calculation = $B - B * F2 / (F1 * 2)$

In our present problem FIFO depth = $B - B * 40 / (30 * 2) = B (1 - 2/3) = B/3$

That means if our Burst amount of data is 10

FIFO DEPTH = $10/3 = 3.333 = 4$ (approximately)

If B = 20 FIFO depth = $20/3 = 6.6 = 7$ or 8 (Clocks are asynchronous)

If B = 30 FIFO depth = $30/3 = 10 \Rightarrow 10 + 1 = 11$ (clocks are asynchronous)

If 10 IDLE cycles between two read cycles.

FIFO DEPTH = $B - B * F2 / (F1 * 10)$

= $B (1 - 4/30)$

= $B * 26/30$