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## Lecture 3

# FIR Design and Decision Feedback Equalization

Mark Horowitz  
Computer Systems Laboratory  
Stanford University  
[horowitz@stanford.edu](mailto:horowitz@stanford.edu)

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Stefanos Sidiropoulos, and Bora Nikolic

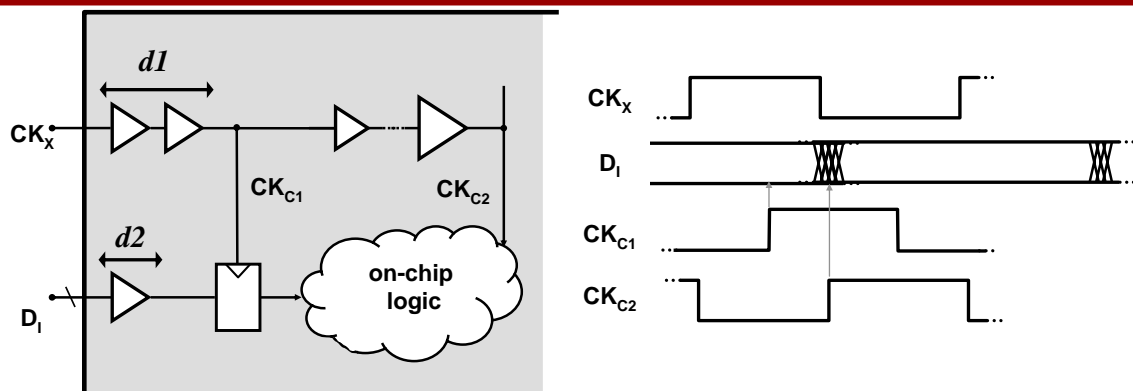
## Readings

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- Readings (for next lecture on adders)
    - Chandrakasan Chapter 10.1-10.2.10
    - Harris Taxonomy of adders (either paper on web or WH 10.2 to 10.2.2.9)
  - Overview:
    - Finish up some timing issues from high-speed links
    - Your project will be the design of a decision feedback equalizer, but most of the hardware will be the same as a normal FIR filter. So the lecture will start talking about FIR filter design, and then will go into the added issues with building a DFE. WARNING: I am not an expert in this area, so there might be better ideas out there (and some bugs in these notes)
    - The FIR notes are from Bora Nikolic at UCB.

## I/O Clocking Issues

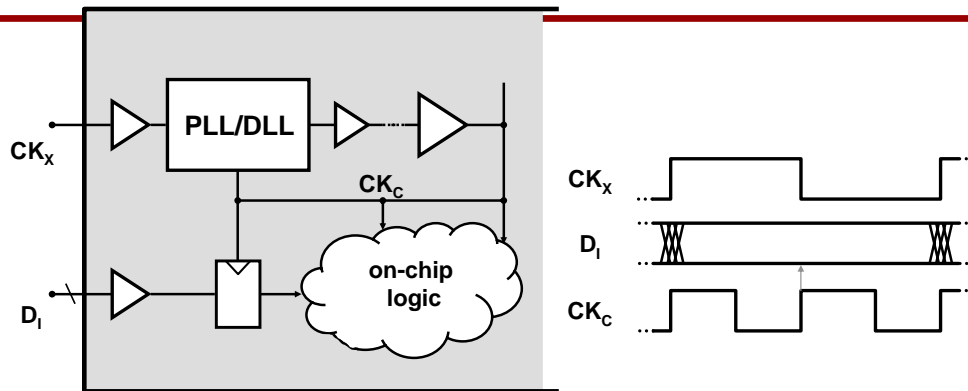
- Remember the clocking issues:
  - Long path constraint (setup time)
  - Short path constraint (hold time)
  - Need to worry about them for I/O as well
- For I/O need to worry about a number of delays
  - Clock skew between chips
  - Data delay between chips
    - Can be larger than a clock cycle (speed of light)
  - Clock skew between external clock and internal clock
    - This can be very large if not compensated
    - It is essentially the insertion delay of the clock tree

## System Clocking: Simple Synchronous Systems



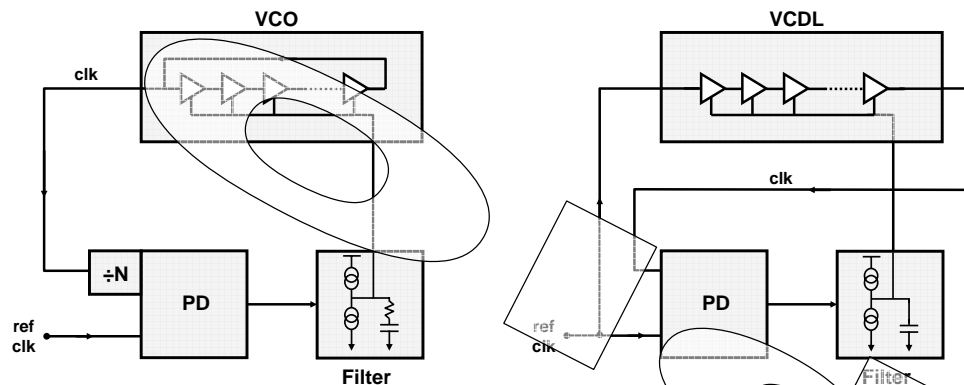
- Long bit times compared to on chip delays:
  - Rely on buffer delays to achieve adequate timing margin

## PLLs: Creating Zero Delay Buffers



- On-chip clock might be a multiple of system clock:
  - Synthesize on-chip clock frequency
- On-chip buffer delays do not match
  - Cancel clock buffer delay

## Used to Argue About PLLs vs DLLs



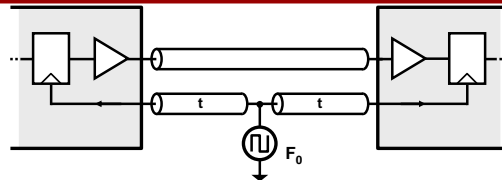
- Second/third order loop:
  - ➔ Stability is an issue
  - ➔ Frequency synthesis easy
  - ➔ Ref. Clk jitter gets filtered
  - ➔ Phase error accumulates
- First order loop:
  - ➔ Stability guaranteed
  - ➔ Frequency synthesis problematic
  - ➔ Ref. Clk jitter propagates
  - ➔ Phase error does not accumulate

## After Many Years of Research

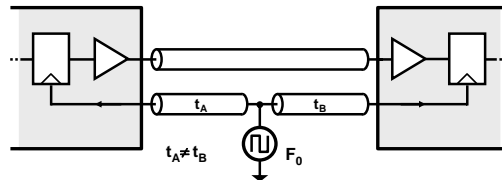
- And many papers and products
- One can mess up either a DLL or PLL
  - Each has its own strengths and weaknesses
- If designed correctly, either will work well
  - Jitter will be dominated by other sources
- Many good designs have been published
  - It is now a building block that is often reused
  - We all have our favorites, mine is the dual-loop design
- And yes, people use ring oscillators
  - Still an open question about how much LC helps (in system)

## Clocking Structures

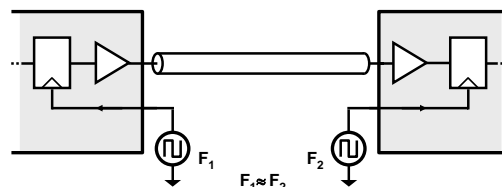
- Synchronous:  
Same frequency and phase
  - Conventional buses



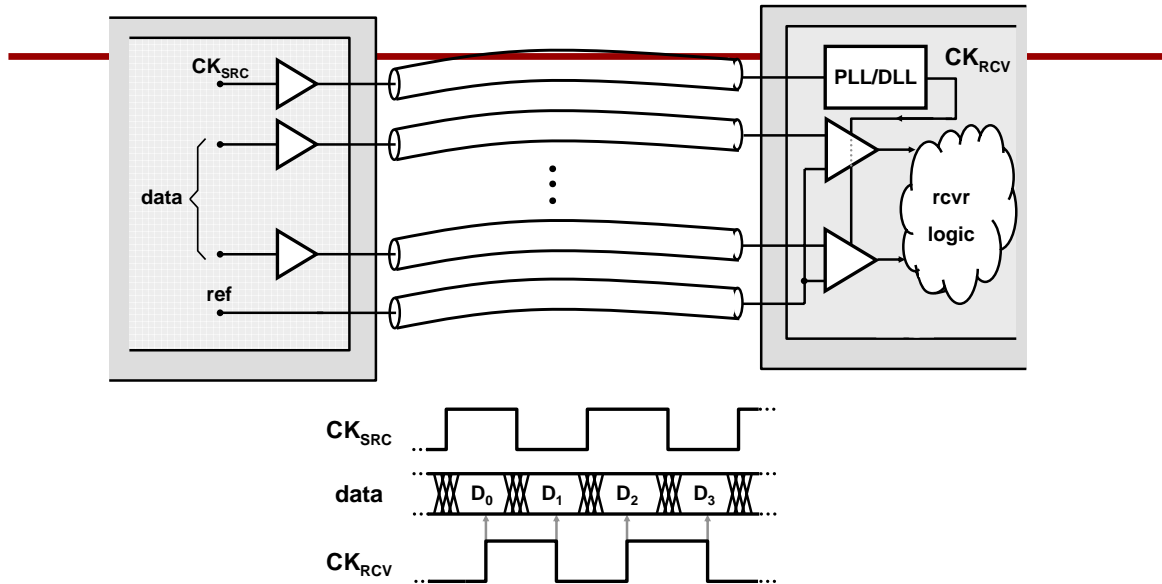
- Mesochronous  
Same frequency, unknown phase
  - Fast memories
  - Internal system interfaces
  - MAC/Package interfaces



- Plesiochronous:  
Almost the same frequency
  - Mostly everything else today

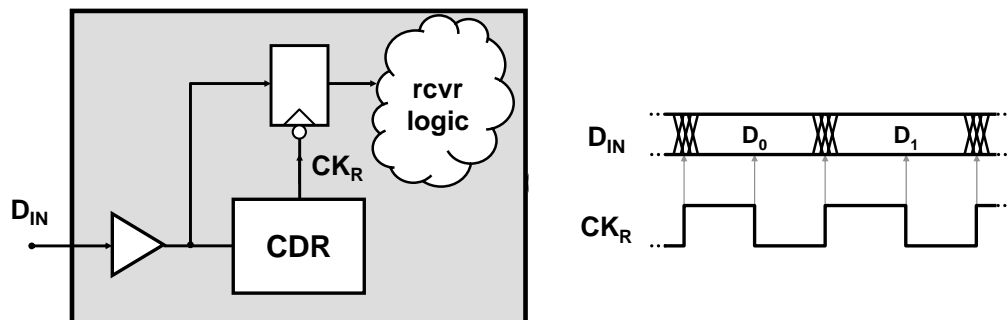


## Source Synchronous Systems



- Position on-chip sampling clock at the optimal point  
i.e. maximize “timing” margin

## Serial Link Circuit



- Recover incoming data fundamental frequency
- Position sampling clock at the “optimal” point

## Finite Impulse Response Filters

- In DSP filters are done in the discrete time domain
  - Instead of  $x(t)$ ,  $x_n$

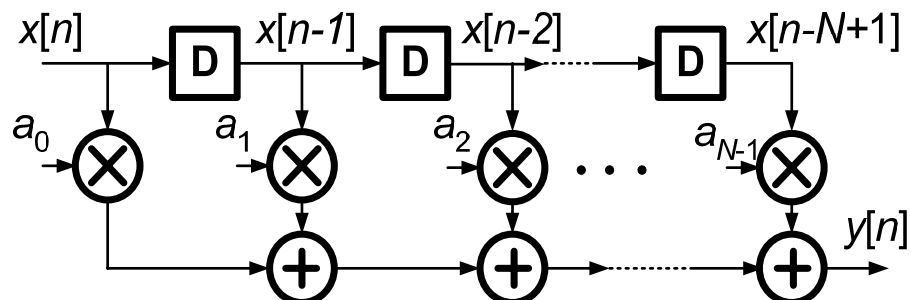
- Filter is formed by convolution of input with filter  $h(t)$ 
  - Output at every point is the sum:

$$y[n] = a_0 x[n] + a_1 x[n-1] + a_2 x[n-2] + \dots + a_N x[n-N+1]$$

- This is generally called an FIR filter
  - Finite impulse response filter (output depends only on input)
  - IIR filters have output depend on prior output
    - Infinite impulse response (like RC circuits)

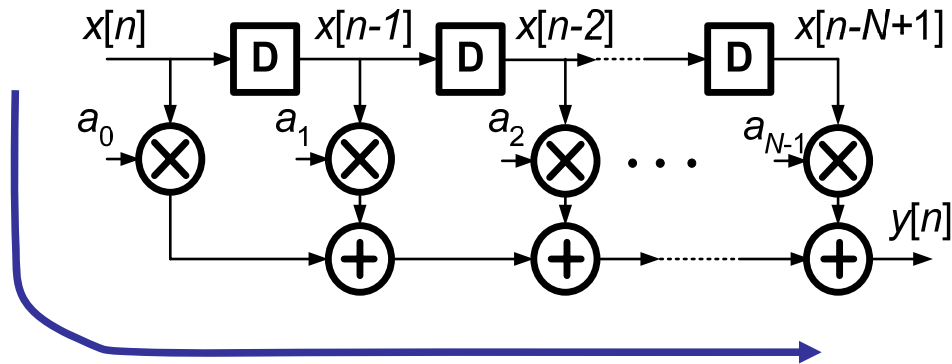
## Transversal Filter

$$y[n] = a_0 x[n] + a_1 x[n-1] + a_2 x[n-2] + \dots + a_N x[n-N+1]$$



## Critical Path

- Digital FIR



$$T = T_{mult} + (N-1)T_{add}$$

## One Point To Keep In Mind

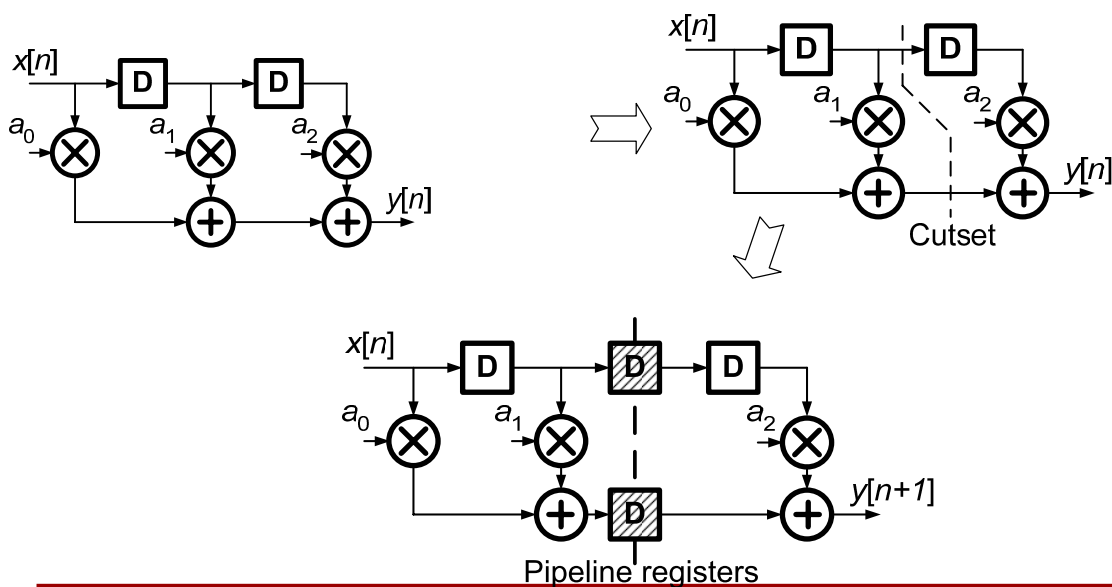
- We are working with small signal values
  - For binary (2 PAM)  $x$  is in  $\{0,1\}$
  - For 4PAM  $x$  is in  $\{0,1,2,3\}$
- So multiplication is generally not an issue
  - For 2 PAM it is trivial
  - For 4 PAM one shift and add
- The problem is the adds
  - While  $x$  is one or two bits, the “a” are larger
    - Generally larger than input precision
    - Since you need to add many of them up and have small quantization errors.

# Pipelining

- Pipelining can be used to increase throughput
  - True for digital and mixed signal implementations
- Pipelining: Adding same number of delay elements
  - In each forward cutset (in the data-flow graph)
    - From the input to the output
- Cutset: set of edges that if removed, graph becomes disjoint
  - Forward cutset: cutset from input to output over all edges
- Plus - Increases frequency
- Minus - Increases latency and register overhead (power, area)

# Pipelining

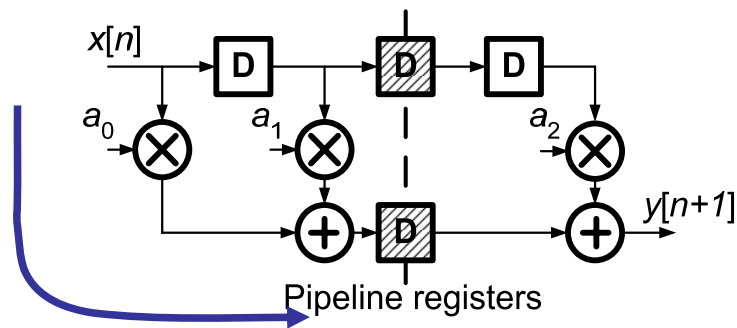
- 3-tap FIR





## Pipelined Direct FIR

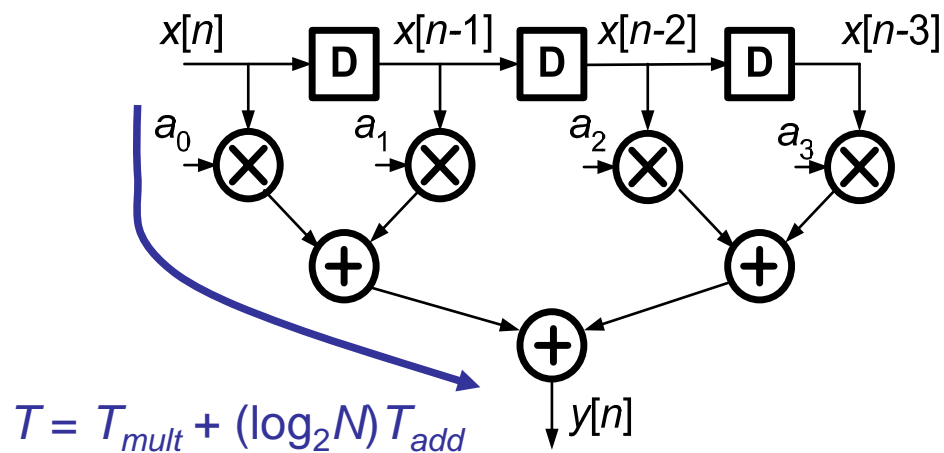
- Critical path



$$T = T_{mult} + T_{add}$$

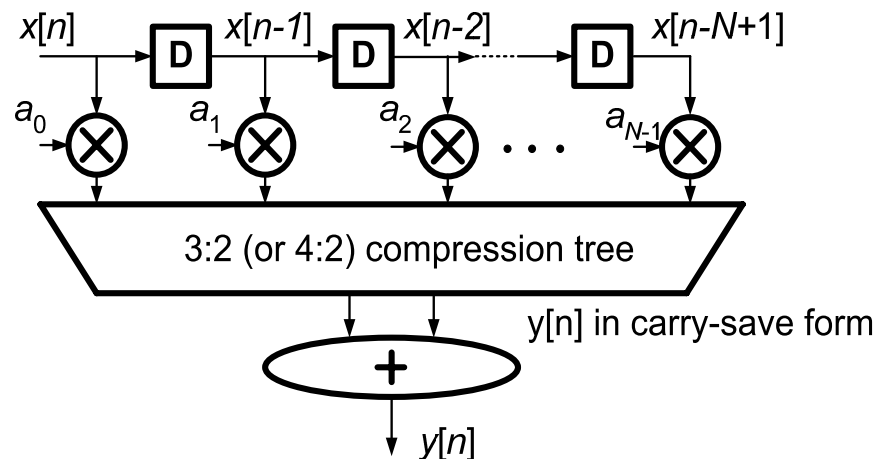
## Multi-Operand Addition

- Adders form a tree



## Multi-Operand Addition

- Using 3:2 or 4:2 compression
  - This is the same as a multiplier tree (in two lectures)



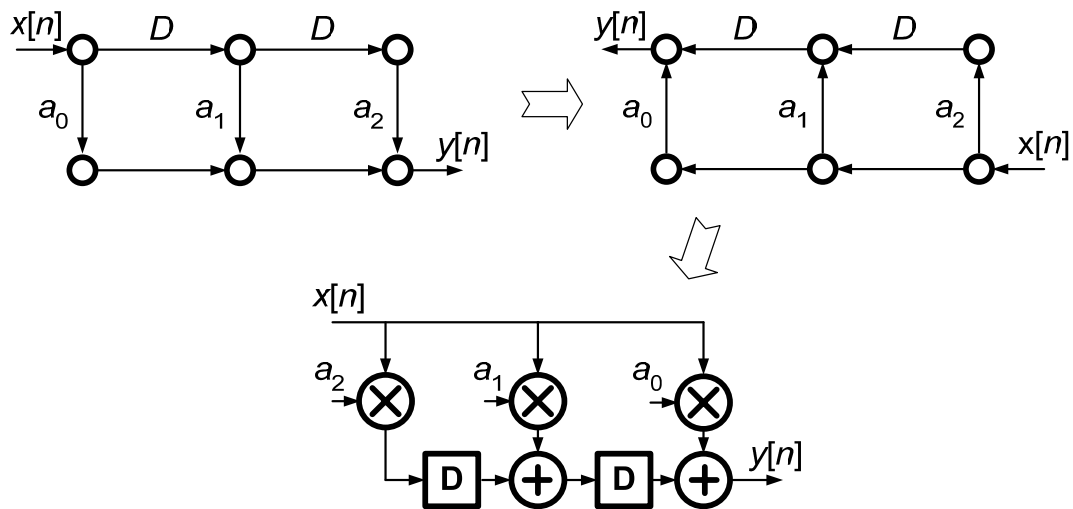
- Optional pipelining, 1-2 stages

## Transposing FIR

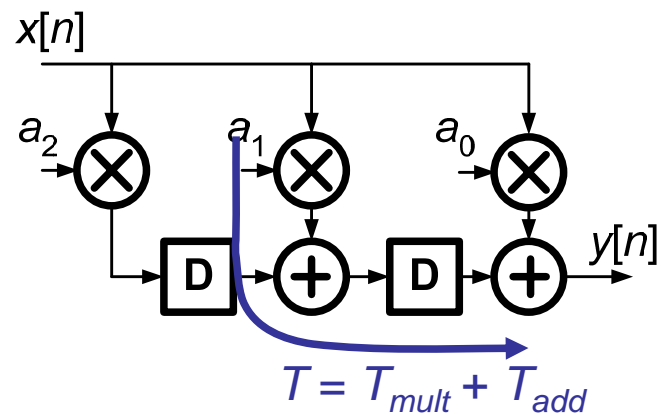
- Transposition:
  - Reversing the direction of all the edges
    - In a signal-flow graph,
  - Interchanging the input and output ports
  - Functionality unchanged

## Transposed FIR

- Represent as a signal-flow graph



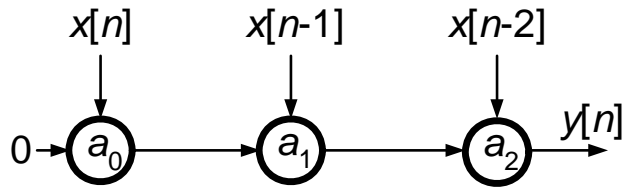
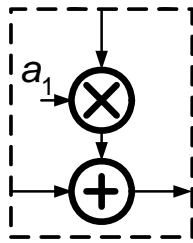
## Transposed FIR



- Critical path shortened
- Input loading increased

## Parallel FIR

- Feed-forward algorithms are easy to parallelize
- Processing element representation of a transversal filter

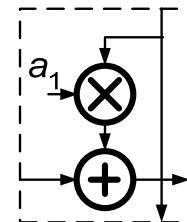
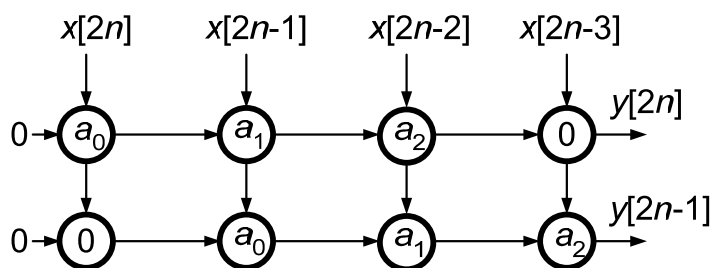


**Processing element**

**Transversal filter**

## Parallel FIR

- Two parallel paths
- Two cycles to complete operation
- Can be extended to more



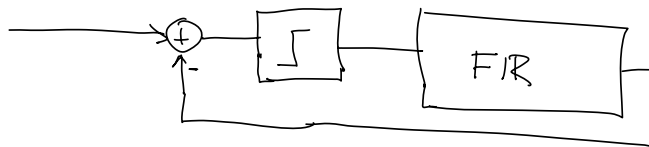
**Two parallel path FIR**

**Processing element**

## Table Lookup

- If the input data is only one or two bits
  - There are not that many input combinations
- Rather than adding the numbers together
  - Add them before hand, and just store the results in a SRAM
  - Address of SRAM is just sequence of inputs to filter
    - $x_n x_{n-1} x_{n-2} x_{n-3} x_{n-4}$
  - Values in memory
    - 00000            0                    00001             $a_4$
    - 00010             $a_3$                     00011             $a_3+a_4$
- Replaces adds and multipliers by memory
  - But it grows exponentially with number of bits needed

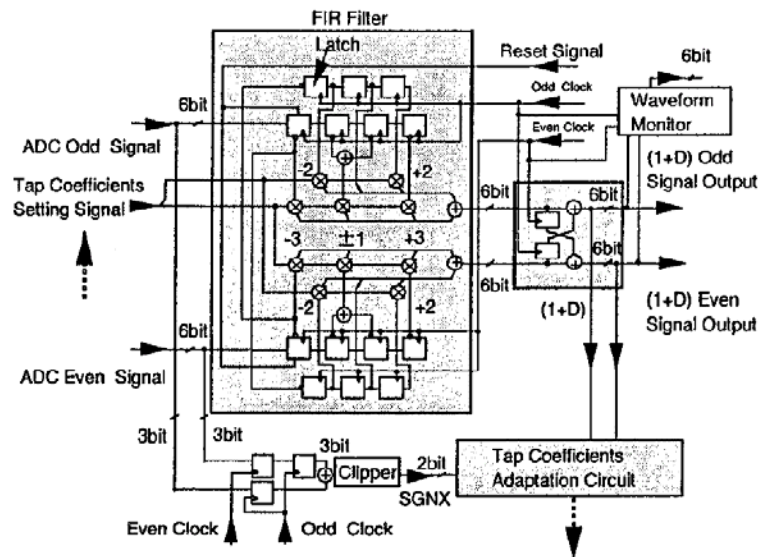
## Decision Feedback Equalization



- The main problem with DFE
  - You need the output of the FIR filter NOW
    - Need it to generate the next bit
- Latency in the FIR filter is a problem

## Practical Digital Equalizers

- Mita, ISSCC'96, two parallel paths
- 150Mb/s 0.7 $\mu$ m BiCMOS

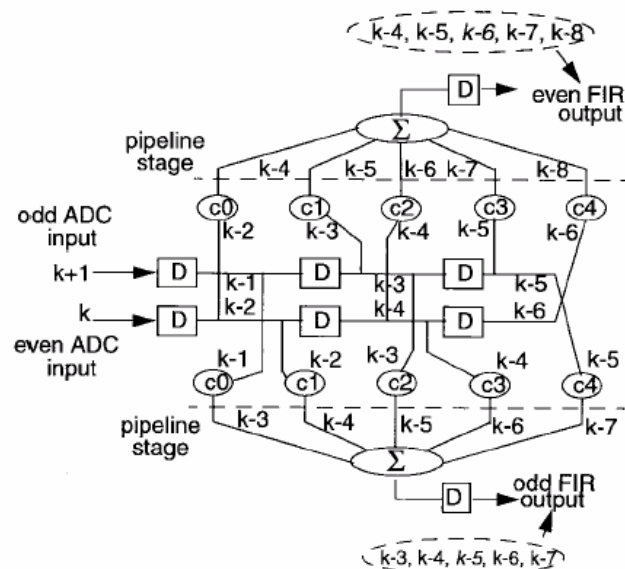


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## Practical Digital Equalizers

- Moloney, JSSC 7/98, 2 parallel paths, 3:2 Wallace
- 150Mb/s 0.7 $\mu$ m BiCMOS

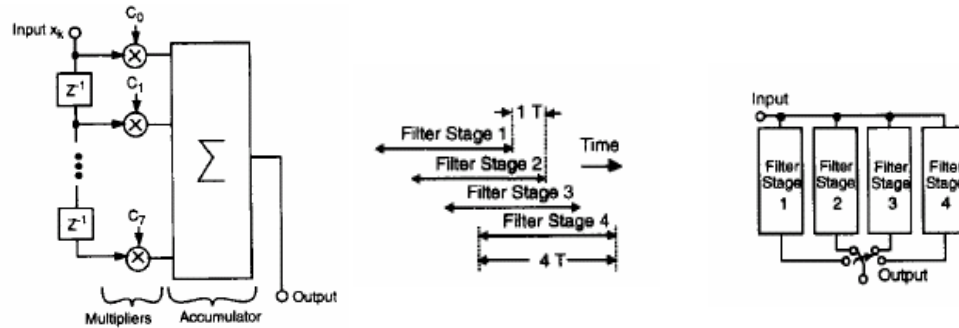


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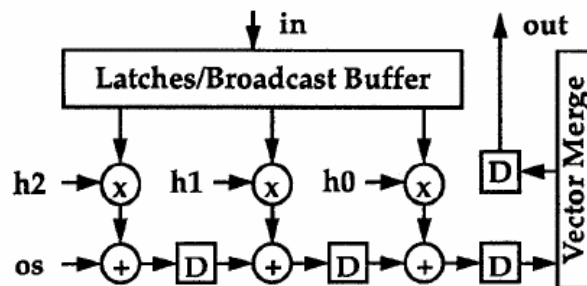
## Practical Digital Equalizers

- Wong, Rudell, Uehara, Gray JSSC 3/95, 4 parallel paths
- 50Mb/s, 1.2 $\mu$ m CMOS



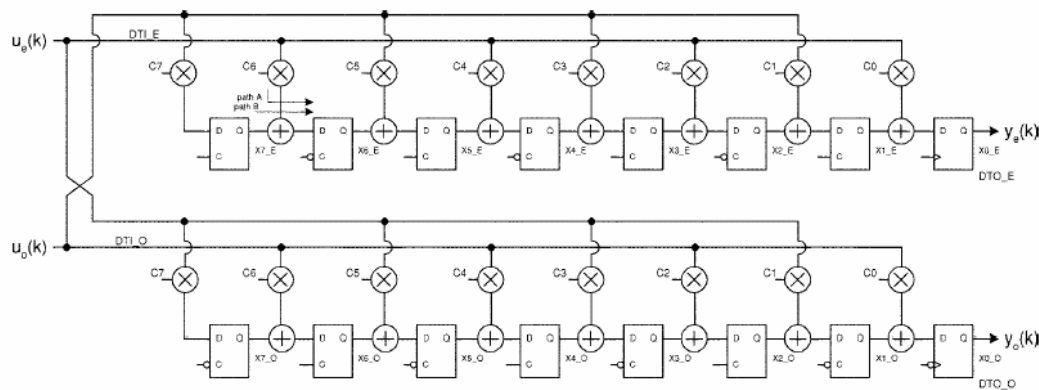
## Practical Digital Equalizers

- Thon, ISSCC'95
- Transposed filter, 240Mb/s 0.8 $\mu$ m 3.7V CMOS, 150mW
- Semi-static coefficients, Booth-encoded



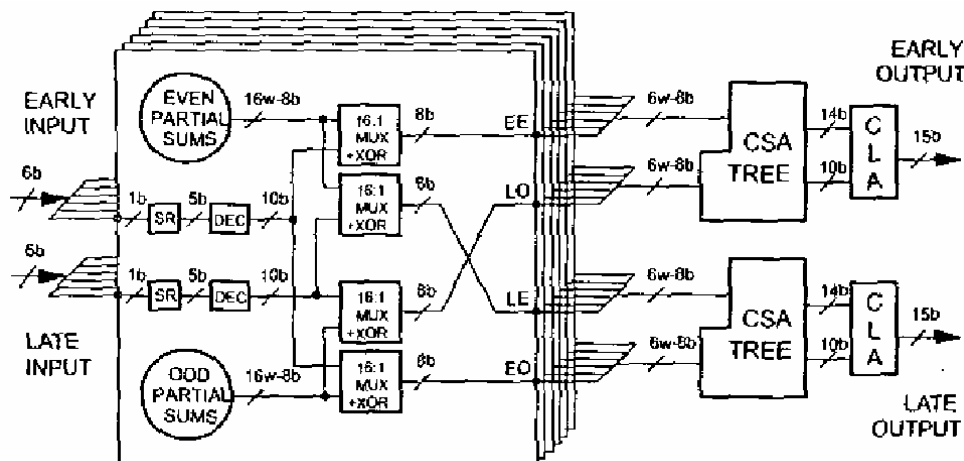
## Practical Digital Equalizers

- Staszewski, JSSC 8/00
- 2 parallel transposed paths, Booth encoded data
- 550Mb/s 0.21 $\mu$ m CMOS, 36mW



## Practical Digital Equalizers

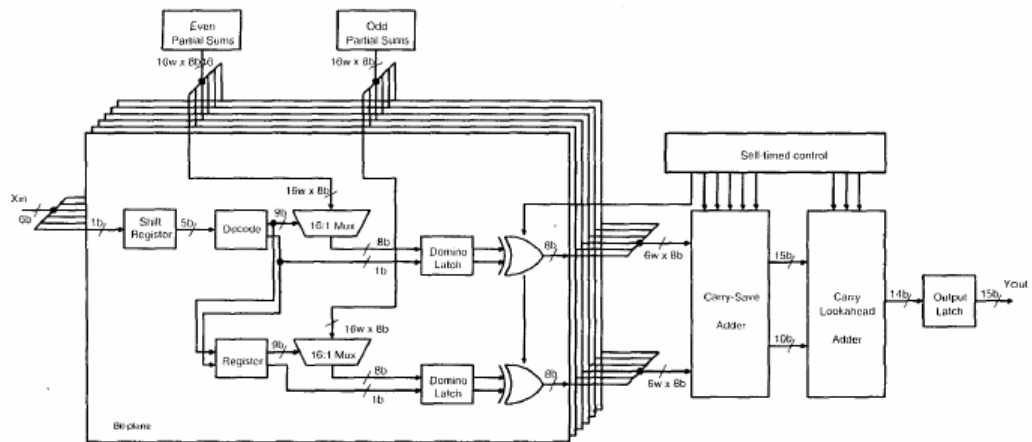
- Rylov, ISSCC'01
- 2.3Gb/s, 1.2W, 0.18 $\mu$ m domino CMOS
- Distributed arithmetic





## Practical Digital Equalizers

- Tierno, ISSCC'02
- 1.3Gb/s, 450mW, 0.18 $\mu$ m 2.1V domino CMOS



## TI DFE Design ISSCC 07

- Uses Memory lookup
  - Runs at 12Gs/s
  - Binary
- Check it out ...

# References from Bora Nikolic

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