# EBU6018 Advanced Transform Methods

## Discrete Fourier Transform\_2 Fast Fourier Transform

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### **Fast Fourier Transform (FFT)**

- What is the FFT?
  - A collection of "tricks" that exploit the symmetry of the DFT calculation to make its execution much faster
  - Speedup increases with DFT size
- This lecture: outline the basic workings of the simplest formulation, the radix-2 decimation-intime algorithm.
- Radix-2 means that the number of elements is a power of 2 (4, 8, 16, 32, etc)
- Decimation-in-time means that the input function is sampled in the time domain.



### Introduction, continued

- Some dates:
  - ~1880 algorithm first described by Gauss
  - 1965 algorithm rediscovered (not for the first time)
     by Cooley and Tukey
- FFT Revolutionized digital signal processing from 1960s
- E.g. in 1967 8192-point DFT on mainframe IBM 7094:
  - ~30 minutes using conventional techniques
  - ~5 seconds using FFTs







# Measures of computational efficiency

- Could consider
  - Number of additions
  - Number of multiplications
  - Amount of memory required
  - Scalability and regularity
- Focus most on number of multiplications
  - More costly than additions for fixed-point processors
  - Same cost as additions for floating-point processors, but number of operations is comparable



# Computational Cost of Discrete-Time Operation

- In general, the output of a system (for example a transform) is the convolution of the input with the impulse response of the system
- So, the convolution of an N-point input with an M-point unit sample response ....
- Direct convolution:  $y[n] = \sum_{k=-\infty} x[k]h[n-k]$

Number of multiplies ≈ MN

For N >> M the computation is  $O(N^2)$ 





### Computational Cost of Discrete-Time

- Convolution of an *N*-point input with an *M*-point unit sample response ....
  - Replace the convolution by multiplying the transforms of the two functions then getting the inverse transform of the product.
- Using transforms directly:  $X[k] = \sum_{n=0}^{N-1} x[n]e^{-j2\pi kn/N}$ 
  - Computation of N-point DFTs requires  $N^2$  multiplies
  - Each convolution (two direct transforms plus an inverse transform) requires three DFTs

For N >> M the computation is approx  $3N^2$   $O(N^2)$ So, no time is saved replacing convolution by DFT, but can now use FFT to be faster





#### **Cooley-Tukey decimation-in-time** algorithm

Consider DFT algorithm for an integer power of 2,

$$X[k] = \sum_{n=0}^{N-1} x[n]W_N^{nk} = \sum_{n=0}^{N-1} x[n]e^{-j2\pi nk/N} \qquad W_N = e^{-j2\pi/N}$$

Take alternate values of the input sequence.

 Create separate sums for even and odd values of n:

$$X[k] = \sum_{n \text{ even}} x[n]W_N^{nk} + \sum_{n \text{ odd}} x[n]W_N^{nk}$$
 factor in this lecture – common in FFT to

Note different sign in twiddle common in FFT texts

• Letting n = 2r for n even and n = 2r + 1 for n odd,

we get 
$$X[k] = \sum_{r=0}^{(N/2)-1} x[2r]W_N^{2rk} + \sum_{r=0}^{(N/2)-1} x[2r+1]W_N^{(2r+1)k}$$





#### Cooley-Tukey decimation in time algorithm

Splitting indices in time, we have obtained

$$X[k] = \sum_{r=0}^{(N/2)-1} x[2r]W_N^{2rk} + \sum_{r=0}^{(N/2)-1} x[2r+1]W_N^{(2r+1)k}$$

• But 
$$W_N^2 = e^{-j2\pi 2/N} = e^{-j2\pi/(N/2)} = W_{N/2}$$

• and 
$$W_N^{2rk}W_N^k = W_N^kW_{N/2}^{rk}$$

So: 
$$X[k] = \sum_{r=0}^{(N/2)-1} x[2r]W_{N/2}^{rk} + W_N^k \sum_{r=0}^{(N/2)-1} x[2r+1]W_{N/2}^{rk}$$

$$N/2\text{-point DFT of } x[2r] \qquad N/2\text{-point DFT of } x[2r+1]$$



### Savings so far ...

We have split the DFT computation into two halves:

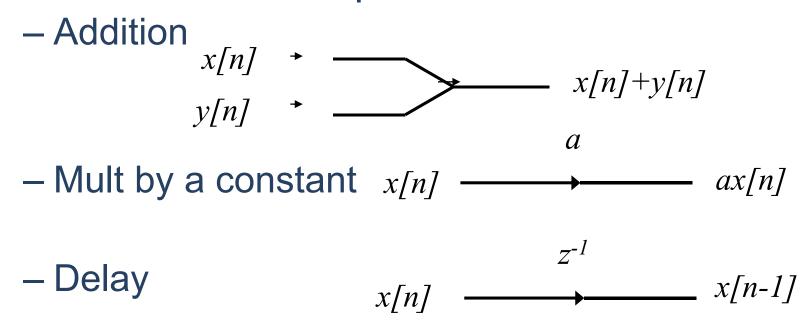
$$X[k] = \sum_{k=0}^{N-1} x[n]W_N^{nk}$$
 If x[n] is 8-point, then we now have two 4-point 
$$= \sum_{r=0}^{(N/2)-1} x[2r]W_{N/2}^{rk} + W_N^k \sum_{r=0}^{(N/2)-1} x[2r+1]W_{N/2}^{rk}$$

- Have we gained anything? Consider the nominal number of multiplications for N=8
  - Original form produces  $8^2 = 64$  multiplications
  - New form produces  $2(4^2)+8=40$  multiplications
  - So we're already ahead …… Let's keep going!!



### Signal flowgraph notation

- In generalizing this formulation, it is most convenient to adopt a graphic approach ...
- Signal flowgraph notation describes the three basic DSP operations:



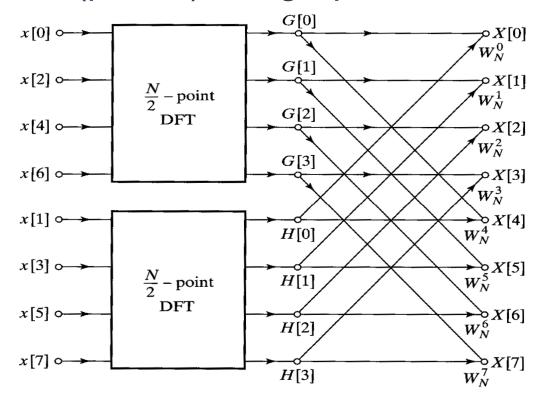




### Signal flowgraph representation of 8-point DFT

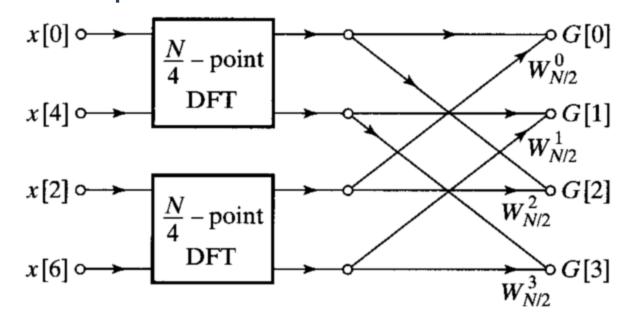
Recall that the DFT is now of the form

• The DFT in (partial) flowgraph  $X[k] = G[k] + W_N^k H[k]$ 



#### Continuing with the decomposition

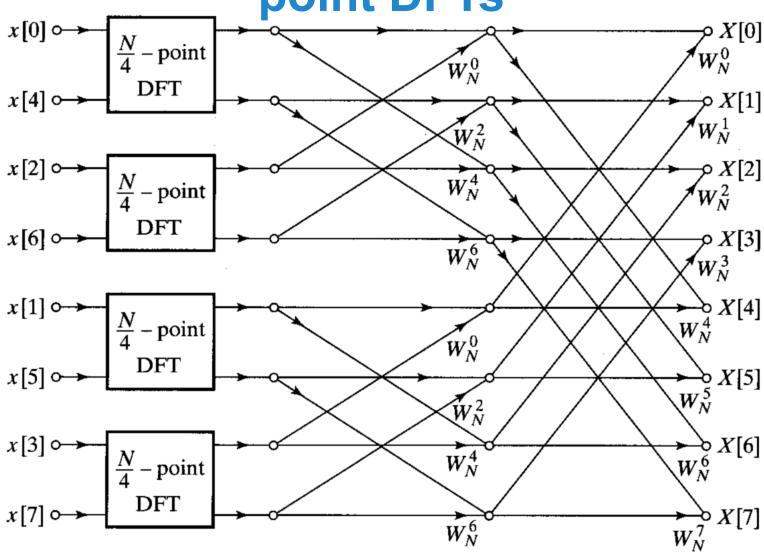
- So why not break up into additional DFTs?
- Let's take the upper 4-point DFT and break it up into two 2-point DFTs:





The complete decomposition into 2-









# Now let's take a closer look at the 2-point DFT

The expression for the 2-point DFT is:

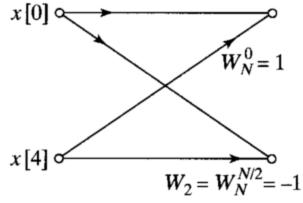
$$X[k] = \sum_{n=0}^{1} x[n]W_2^{nk} = \sum_{n=0}^{1} x[n]e^{-j2\pi nk/2}$$

• Evaluating for k = 0,1 we obtain

$$X[0] = x[0] + x[1]$$

$$X[1] = x[0] + e^{-j2\pi 1/2}x[1] = x[0] - x[1]$$

which in signal flowgraph notation looks like ...

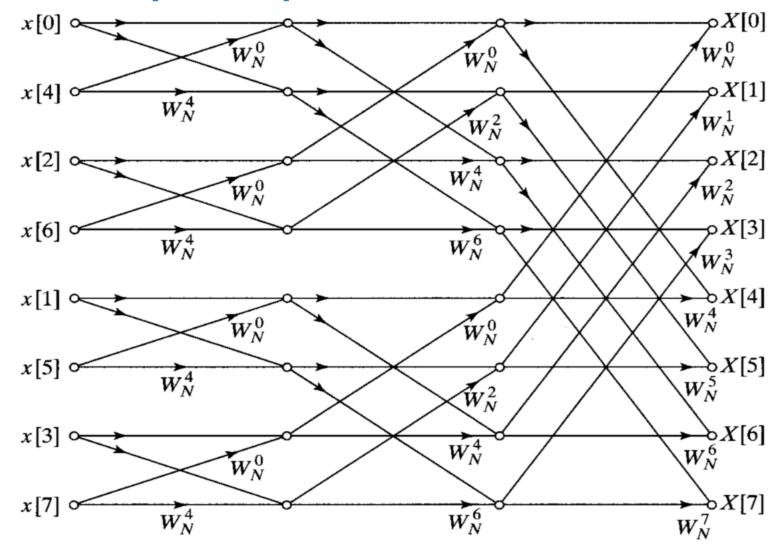


This topology is called the basic "butterfly"





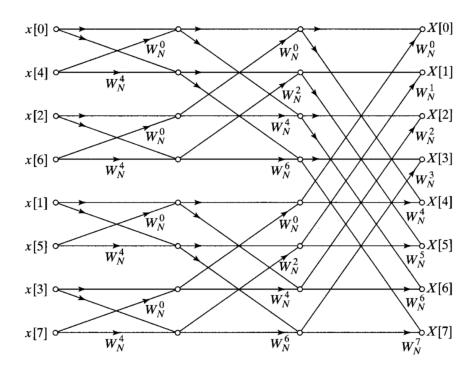
#### The complete 8-point decimation-in-time FFT







#### Number of multiplies for N-point FFT

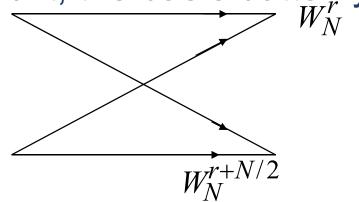


- Let  $N = 2^{\nu}$  where  $\nu = \log_2(N)$
- $(\log_2(N) \text{ columns})(N/2 \text{ butterflys/column})(2 \text{ mults/butterfly})$  or approx  $N\log_2(N)$  multiplications

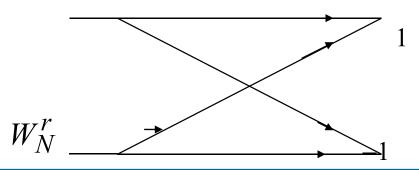


# Additional timesavers: reducing multiplications in the basic butterfly

As we derived it, the basic butterfly is of the form



Since  $W_N^{N/2} = -1$  we can reduce computation by 2 by premultiplying by  $W_N^r$ 

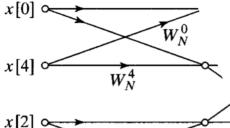




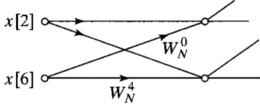


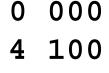
### Bit reversal of the input

Recall the first stages of the 8-point FFT:



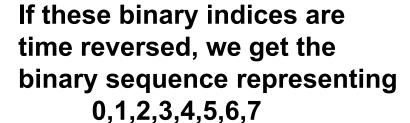
Consider the binary representation of the indices of the input:

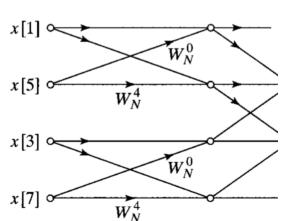




010

110





1 001

5 101

3 011

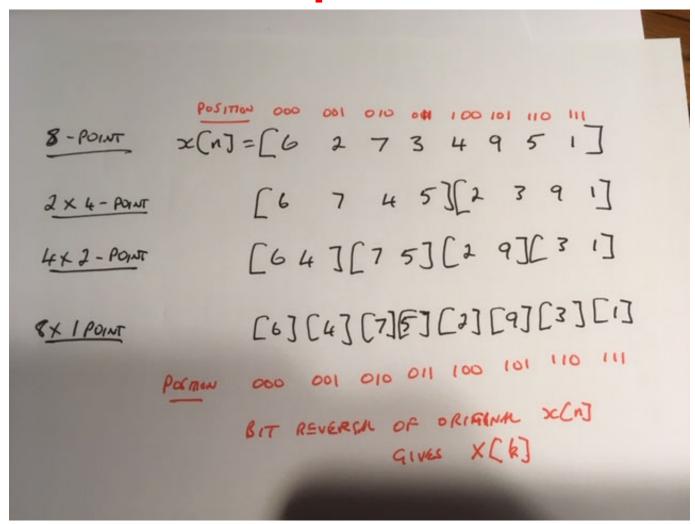
7 111

Hence the indices of the FFT inputs are said to be in

bit-reversed order



## Example with an 8-point input sequence









#### Some comments on bit reversal

- This implementation of FFT: input is bit reversed, output is in natural order
- Sometimes convenient to implement filtering applications by
  - Use FFTs with input in natural order, output in bitreversed order
  - Multiply frequency coefficients together (in bitreversed order)
  - Use inverse FFTs with input in bit-reversed order, output in natural order
- Computing in this fashion means we never have to compute bit reversal explicitly



