Administrative Stuff

- Matlab4 is posted due Thursday Nov 2
- Exam1 handed back after class



EE-125: Digital Signal Processing

FFT and Related Algorithms (P&M Chapter 8, in 1 lecture!)

Professor Tracey



Outline

- Writing the DFT as a matrix
- Big picture / motivation for FFT
- Reviewing the FFT video: decimation in time
 - Van Veen covers the 'decimation in time' algorithm splitting data x(n) into even and odd samples n
- If time permits: 'decimation in frequency' algorithm
 - split X(k) into even and odd frequencies n
- Some other transforms
 - You'll use DCT in the next Matlab project



Big picture with FFT-1

We saw earlier that the DFT can be written in matrix form:

$$X = W X$$

where X and x are N-point vectors (in freq and time, respectively, and **W** is a NxN matrix with exponential terms

- Computing the DFT using the matrix approach takes
 O(N^2) calculations
- Same is true for inverse DFT (IDFT)

$$X = W^{-1} X$$



Big picture with FFT - 2

- •FFT exploits symmetries in **W** to speed things up
 - -Break original problem into even and odd: 2 problems, each half the length
 - -Keep breaking down until we get to many 2x2 systems
 - -Most **W** factors for that system are very simplified; many terms come down to exp(0) (i.e., 1) or exp(pi) (i.e. -1) so we can add and subtract instead of doing multiplications
- •Result: calculation is done in O(N log2 N) calculations but assumes we start with data vector with length power-of-2 (or, zero-pad to a power of 2)
- The above splitting by 2 is called a 'radix 2' FFT; other versions are available (radix 4, etc) but less common



The Fast Fourier Transform Algorithm

Pefine $W_N = e^{-j2\pi l_N}$ Define $W_N = e^{-j2\pi l_N}$ Note that $W_N = W_N^{-k_N} = (W_N^{-k_N})^*$ Define $W_N = W_N^{-k_N} = (W_N^{-k_N})^*$ Note that $W_N = W_N^{-k_N} = (W_N^{-k_N})^*$

2) Periodicity in n, k
$$W_N^{kn} = W_N^{k(N+n)} = W_N^{(k+n)n}$$

Decimation in Time FFT (one of many)
- build a big DFT from smaller ones
- Assume N = 2^m

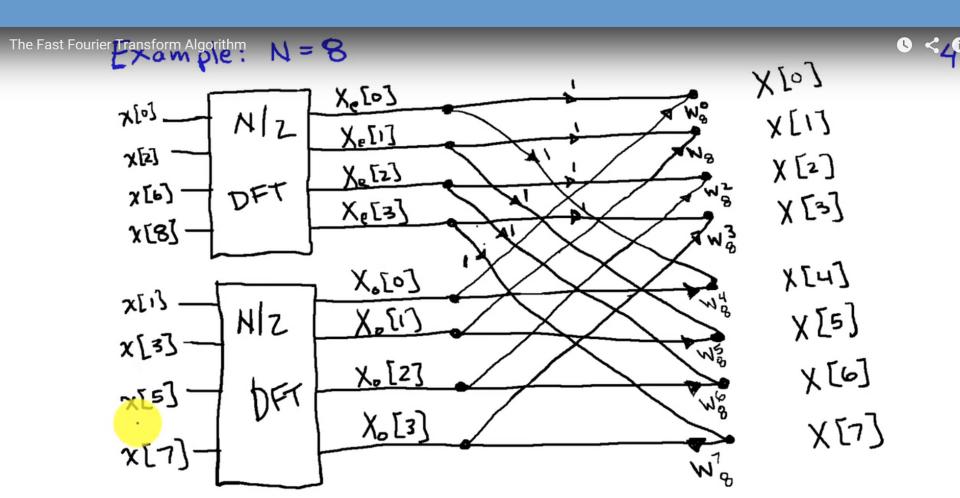
separate $\chi[n]$ into even and odd-indexed subsequences $\chi[k] = \sum_{n=0}^{N-1} \chi[n] W_{n}^{kn} = \sum_{n=0}^{N-1} \chi[n] W_{n}^{kn} + \sum_{n=0}^{N-1} \chi[n] W_{n}^{kn}$

The Fast Fourier Transform Algorithm

odd indices

$$\begin{aligned}
N &= 2r \\
N &= 2r \\$$

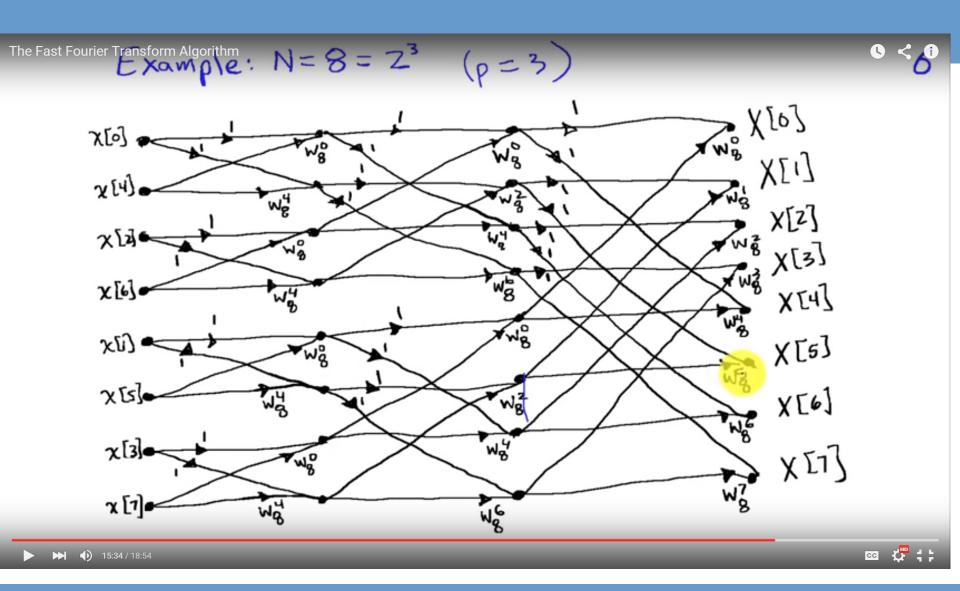
Tufts







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Tufts

Related transforms - Discrete Cosine Transform

- Uses Fourier transform property: if x(n) is real and even, X(k) is real and even.
 - -For real and even x(n), exponential in DFT can be replaced by a cosine, since sine only contributes to imag(X(k)).
- To use the DCT, we copy (extend) non-symmetric signals to make them symmetric: see below.

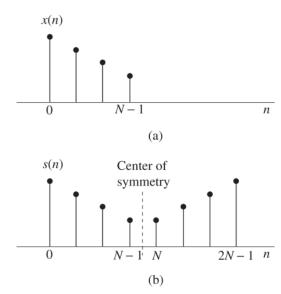
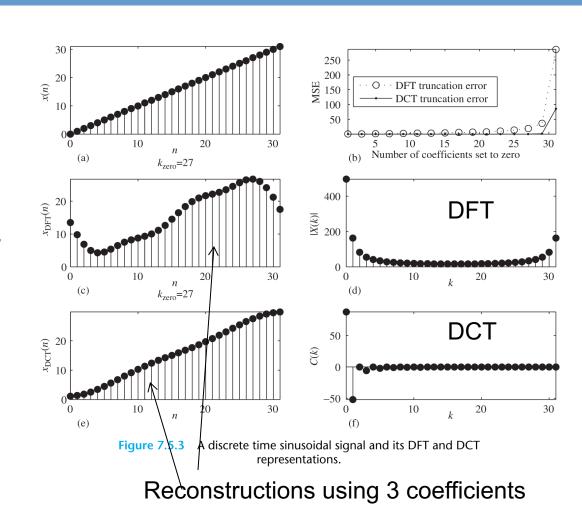


Figure 7.5.1 Original sequence x(n), $0 \le n \le N-1$ and its 2N-point even extension s(n), 0 < n < 2N-1.



Why use the DCT?

- Compared to the DFT, the DCT gives a less compact representation of sinusoids (Figure 7.5.2, book), but more compact representation of other signals. Here, 'compact' means 'fewer coefficients with significant amplitude'
- DCT turns out to be good for many images
- DCT is the basis of the JPEG image compression algorithm





Other algorithms to know about

- Goertzel (P&M 8.3)
 - Rewrite the DFT so we have a parallel bank of filters, each one of which gives the output for a single frequency in the DFT
 - Advantage is that we don't need to implement every frequency; so can be faster than FFT if we just need answers at a few frequencies
 - Classic use: processing of dial tones
- Chirp z-transform (P&M 8.3)
 - Lets us evaluate the transform at points other than the unit circle
 - Used in speech analysis (on-line, see "The Chirp z-Transform Algorithm—A Lesson in Serendipity")

