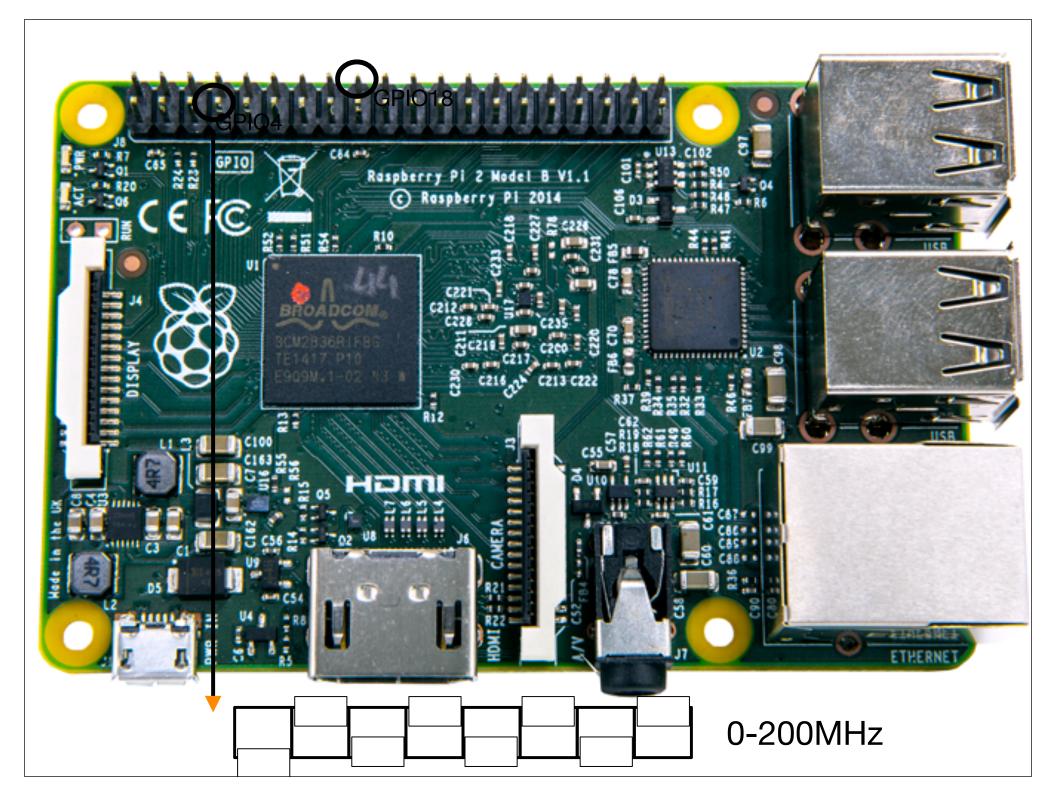


# **Digital Signal Processing**

# Lecture 7 The FFT

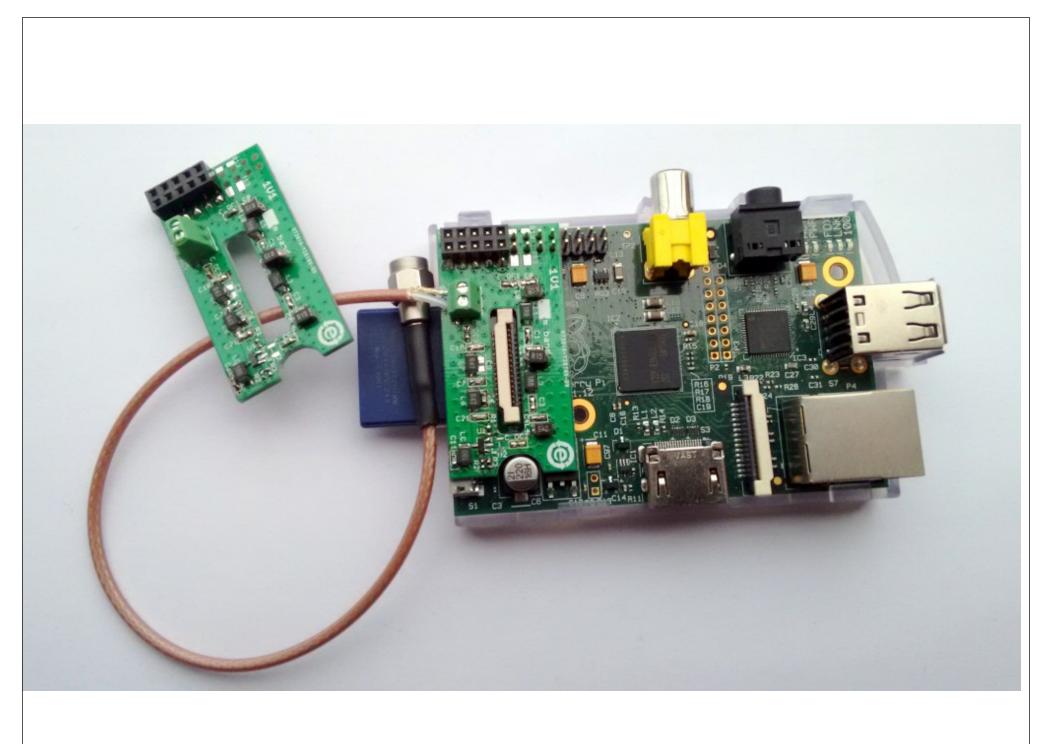
based on slides by J.M. Kahn

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## PiFM/QRPi

- PI FM: <u>http://www.icrobotics.co.uk/wiki/index.php/</u> <u>Turning the Raspberry Pi Into an FM Transmit</u> <u>ter#Steps\_to\_play\_sound</u>:
- RPiTX: <u>https://github.com/F5OEO/rpitx</u>
- WsprryPi: <u>https://github.com/JamesP6000/WsprryPi</u>
- QRPi: <u>http://rfsparkling.com/qrpi/</u>
- qtcsdr: <u>https://github.com/ha7ilm/qtcsdr</u>



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#### Last Time

- Discrete Fourier Transform
  - Linear convolution through circular
  - Linear convolutions through DFT
    - Overlap and add
    - Overlap and save
- Today
  - The Fast Fourier Transform

#### Circular Convolution as Matrix Operation

Circular convolution:

$$h[n] \otimes x[n] = \begin{bmatrix} h[0] & h[N-1] & \cdots & h[1] \\ h[1] & h[0] & & h[2] \\ & & \vdots & \\ h[N-1] & h[N-2] & & h[0] \end{bmatrix} \begin{bmatrix} x[0] \\ x[1] \\ \vdots \\ x[N-I] \end{bmatrix}$$
$$= H_c x$$

- $H_c$  is a circulant matrix
- The columns of the DFT matrix are Eigen vectors of circulant matrices.
- Eigen vectors are DFT coefficients. How can you show?
   Proof in HW

#### Circular Convolution as Matrix Operation

• Diagonalize:

$$W_{N}H_{c}W_{n}^{-1} = \begin{bmatrix} H[0] & 0 \cdots & 0 \\ 0 & H[1] \cdots & 0 \\ \vdots & 0 & H[N-1] \end{bmatrix}$$

• Right-multiply by  $W_N$ 

$$W_N H_c = \begin{bmatrix} H[0] & 0 \cdots & 0 \\ 0 & H[1] \cdots & 0 \\ \vdots & 0 & H[N-1] \end{bmatrix} W_N$$

• Multiply both sides by x

$$W_N H_c x = \begin{bmatrix} H[0] & 0 \cdots & 0 \\ 0 & H[1] \cdots & 0 \\ \vdots & 0 & H[N-1] \end{bmatrix} W_N x$$

#### Fast Fourier Transform Algorithms

• We are interested in efficient computing methods for the DFT and inverse DFT:

$$X[k] = \sum_{n=0}^{N-1} x[n] W_N^{kn}, \quad k = 0, \dots, N-1$$
$$x[n] = \sum_{k=0}^{N-1} X[k] W_N^{-kn}, \quad n = 0, \dots, N-1$$

where

$$W_N = e^{-j\left(\frac{2\pi}{N}\right)}.$$

• Recall that we can use the DFT to compute the inverse DFT:

$$\mathcal{DFT}^{-1}{X[k]} = \frac{1}{N} \left(\mathcal{DFT}{X^*[k]}\right)^*$$

Hence, we can just focus on efficient computation of the DFT.

 Straightforward computation of an N-point DFT (or inverse DFT) requires N<sup>2</sup> complex multiplications.  Fast Fourier transform algorithms enable computation of an N-point DFT (or inverse DFT) with the order of just N · log<sub>2</sub> N complex multiplications.

This can represent a huge reduction in computational load, especially for large N.

N	<i>N</i> <sup>2</sup>	$N \cdot \log_2 N$	$\frac{N^2}{N \cdot \log_2 N}$
16	256	64	4.0
128	16,384	896	18.3
1,024	1,048,576	10,240	102.4
8,192	67,108,864	106,496	630.2
$6 \times 10^{6}$	$36 imes10^{12}$	$135 imes10^{6}$	$2.67 imes10^5$

\* 6Mp image size

• Most FFT algorithms exploit the following properties of  $W_N^{kn}$ :

#### • Conjugate Symmetry

$$W_N^{k(N-n)} = W_N^{-kn} = (W_N^{kn})^*$$

• Periodicity in *n* and *k*:

$$W_N^{kn} = W_N^{k(n+N)} = W_N^{(k+N)n}$$

• Power:

$$W_N^2 = W_{N/2}$$

- Most FFT algorithms decompose the computation of a DFT into successively smaller DFT computations.
  - Decimation-in-time algorithms decompose x[n] into successively smaller subsequences.
  - *Decimation-in-frequency* algorithms decompose X[k] into successively smaller subsequences.
- We mostly discuss <u>decimation-in-time</u> algorithms here.

Assume length of x[n] is power of 2 ( $N = 2^{\nu}$ ). If smaller zero-pad to closest power.

• We start with the DFT

$$X[k] = \sum_{n=0}^{N-1} x[n] W_N^{kn}, \quad k = 0, \dots, N-1$$

• Separate the sum into even and odd terms:

$$X[k] = \sum_{n \text{ even}} x[n] W_N^{kn} + \sum_{n \text{ odd}} x[n] W_N^{kn}$$

These are two DFT's, each with half of the samples.

Let 
$$n = 2r$$
 ( $n$  even) and  $n = 2r + 1$  ( $n$  odd):

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

• Note that:

$$W_N^{2rk} = e^{-j\left(\frac{2\pi}{N}\right)(2rk)} = e^{-j\left(\frac{2\pi}{N/2}\right)rk} = W_{N/2}^{rk}$$

Remember this trick, it will turn up often.

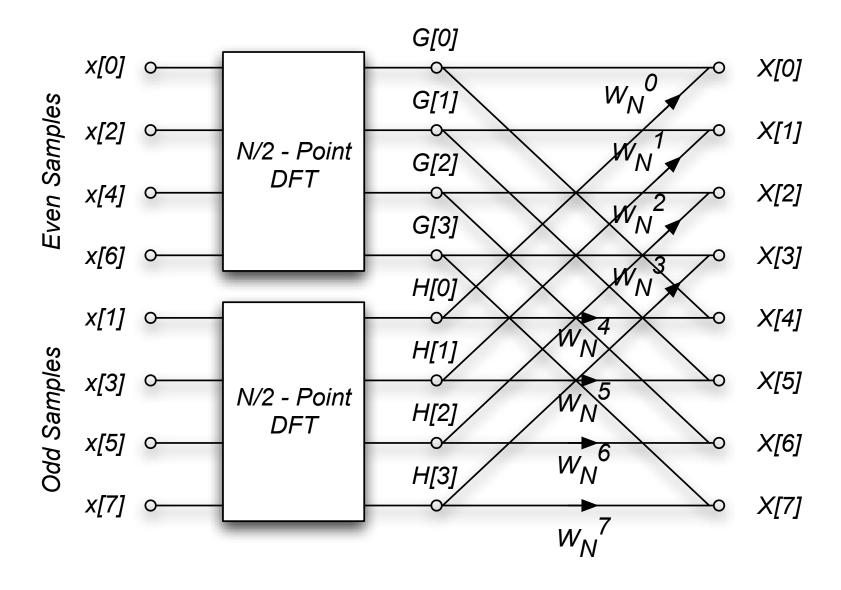
• Hence:

$$X[k] = \sum_{r=0}^{(N/2)-1} x[2r] W_{N/2}^{rk} + W_N^k \sum_{r=0}^{(N/2)-1} x[2r+1] W_{N/2}^{rk}$$
$$\triangleq G[k] + W_N^k H[k], \quad k = 0, \dots, N-1$$

where we have defined:

$$G[k] \stackrel{\Delta}{=} \sum_{r=0}^{(N/2)-1} x[2r] W_{N/2}^{rk} \implies \text{DFT of even idx}$$
$$H[k] \stackrel{\Delta}{=} \sum_{r=0}^{(N/2)-1} x[2r+1] W_{N/2}^{rk} \implies \text{DFT of odd idx}$$

An 8 sample DFT can then be diagrammed as



 Both G[k] and H[k] are periodic, with period N/2. For example

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

SO

$$G[k + (N/2)] = G[k]$$
  
 $H[k + (N/2)] = H[k]$ 

The periodicity of G[k] and H[k] allows us to further simplify.
For the first N/2 points we calculate G[k] and W<sup>k</sup><sub>N</sub>H[k], and then compute the sum

$$X[k] = G[k] + W_N^k H[k] \qquad \forall \{k : 0 \le k < \frac{N}{2}\}.$$

How does periodicity help for  $\frac{N}{2} \le k < N$ ?

A /

$$X[k] = G[k] + W_N^k H[k] \qquad \forall \{k : 0 \le k < \frac{N}{2} \}.$$
  
• for  $\frac{N}{2} \le k < N$ :  

$$W_N^{k+(N/2)} = ?$$
  

$$X[k + (N/2)] = ?$$

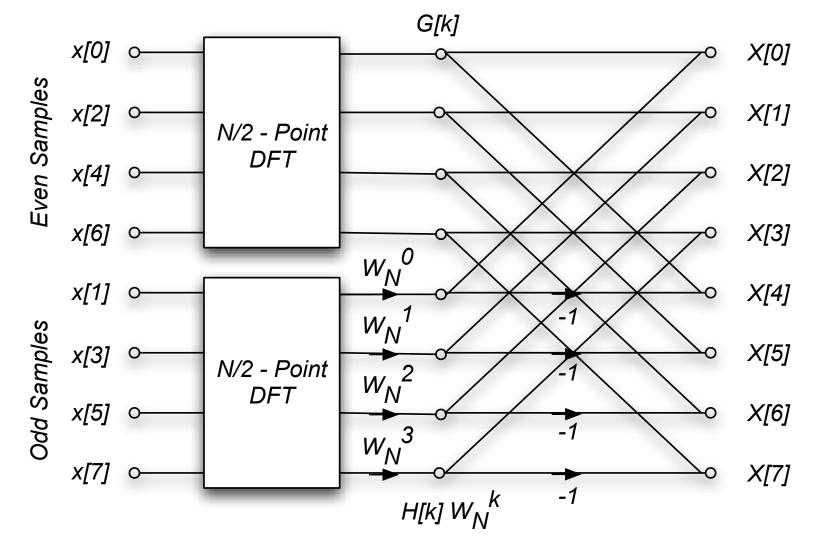
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$$X[k + (N/2)] = G[k] - W_N^k H[k]$$

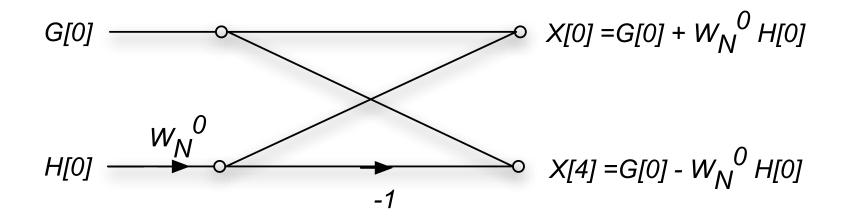
We previously calculated G[k] and  $W_N^k H[k]$ .

Now we only have to compute their difference to obtain the second half of the spectrum. No additional multiplies are required.

• The *N*-point DFT has been reduced two N/2-point DFTs, plus N/2 complex multiplications. The 8 sample DFT is then:

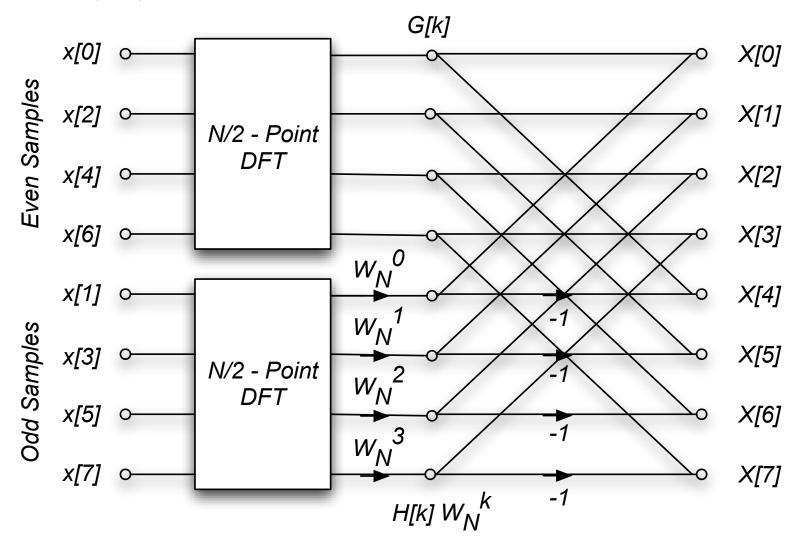


- Note that the inputs have been reordered so that the outputs come out in their proper sequence.
- We can define a *butterfly operation*, e.g., the computation of X[0] and X[4] from G[0] and H[0]:

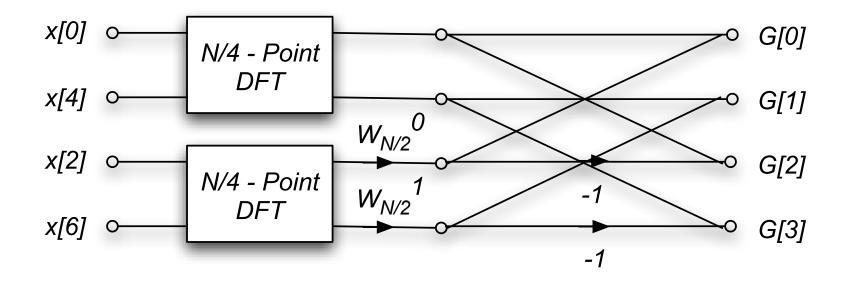


This is an important operation in DSP.

• Still  $O(N^2)$  operations..... What shall we do?



• We can use the same approach for each of the N/2 point DFT's. For the N = 8 case, the N/2 DFTs look like

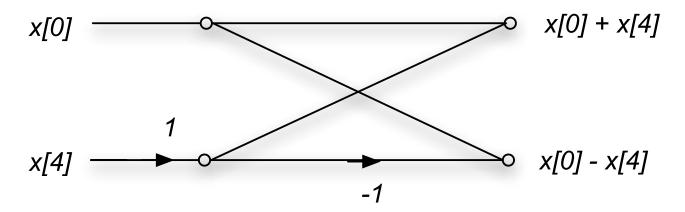


\*Note that the inputs have been reordered again.

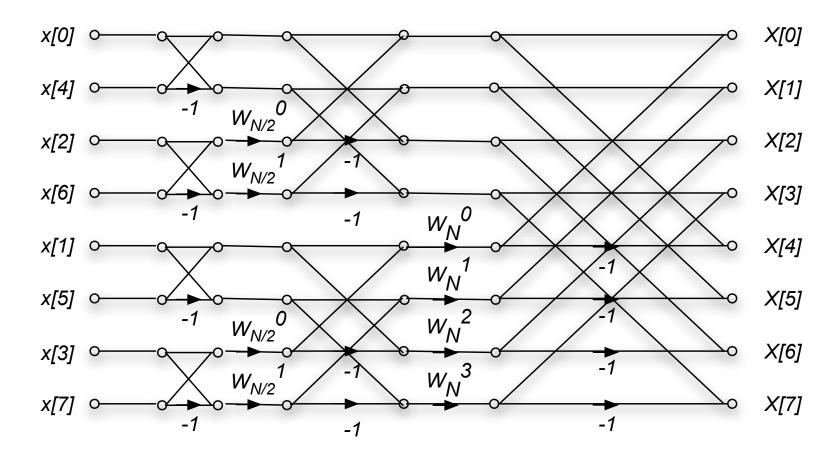
 At this point for the 8 sample DFT, we can replace the N/4 = 2 sample DFT's with a single butterfly. The coefficient is

$$W_{N/4} = W_{8/4} = W_2 = e^{-j\pi} = -1$$

The diagram of this stage is then



Combining all these stages, the diagram for the 8 sample DFT is:



This the decimation-in-time FFT algorithm.

- In general, there are  $\log_2 N$  stages of decimation-in-time.
- Each stage requires N/2 complex multiplications, some of which are trivial.
- The total number of complex multiplications is  $(N/2) \log_2 N$ .
- The order of the input to the decimation-in-time FFT algorithm must be permuted.
  - First stage: split into odd and even. Zero low-order bit first
  - Next stage repeats with next zero-lower bit first.
  - Net effect is reversing the bit order of indexes

This is illustrated in the following table for N = 8.

Decimal	Binary	Bit-Reversed Binary	Bit-Reversed Decimal
0	000	000	0
1	001	100	4
2	010	010	2
3	011	110	6
4	100	001	1
5	101	101	5
6	110	011	3
7	111	111	7

The DFT is

$$X[k] = \sum_{n=0}^{N-1} x[n] W_N^{nk}$$

If we only look at the even samples of X[k], we can write k = 2r,

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

We split this into two sums, one over the first N/2 samples, and the second of the last N/2 samples.

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

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But 
$$W_N^{2r(n+N/2)} = W_N^{2rn} W_N^N = W_N^{2rn} = W_{N/2}^{rn}$$
.  
We can then write

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

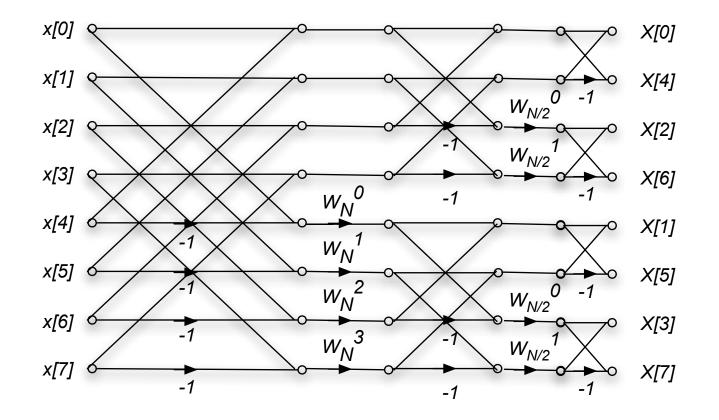
This is the N/2-length DFT of first and second half of x[n] summed.

$$X[2r] = \mathsf{DFT}_{\frac{N}{2}} \{ (x[n] + x[n + N/2]) \}$$
$$X[2r+1] = \mathsf{DFT}_{\frac{N}{2}} \{ (x[n] - x[n + N/2]) W_N^n \}$$

(By a similar argument that gives the odd samples)

Continue the same approach is applied for the N/2 DFTs, and the N/4 DFT's until we reach simple butterflies.

The diagram for and 8-point decimation-in-frequency DFT is as follows



This is just the decimation-in-time algorithm reversed! The inputs are in normal order, and the outputs are bit reversed.

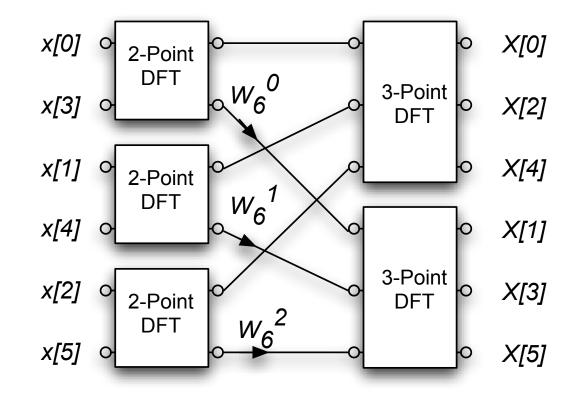
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#### Non-Power-of-2 FFT's

A similar argument applies for any length DFT, where the length N is a composite number.

For example, if N = 6, a decimation-in-time FFT could compute three 2-point DFT's followed by two 3-point DFT's



Good component DFT's are available for lengths up to 20 or so. Many of these exploit the structure for that specific length. For example, a factor of

$$W_N^{N/4} = e^{-j\frac{2\pi}{N}(N/4)} = e^{-j\frac{\pi}{2}} = -j$$
 Why?

just swaps the real and imaginary components of a complex number, and doesn't actually require any multiplies. Hence a DFT of length 4 doesn't require any complex multiplies.

Half of the multiplies of an 8-point DFT also don't require multiplication.

Composite length FFT's can be very efficient for any length that factors into terms of this order.

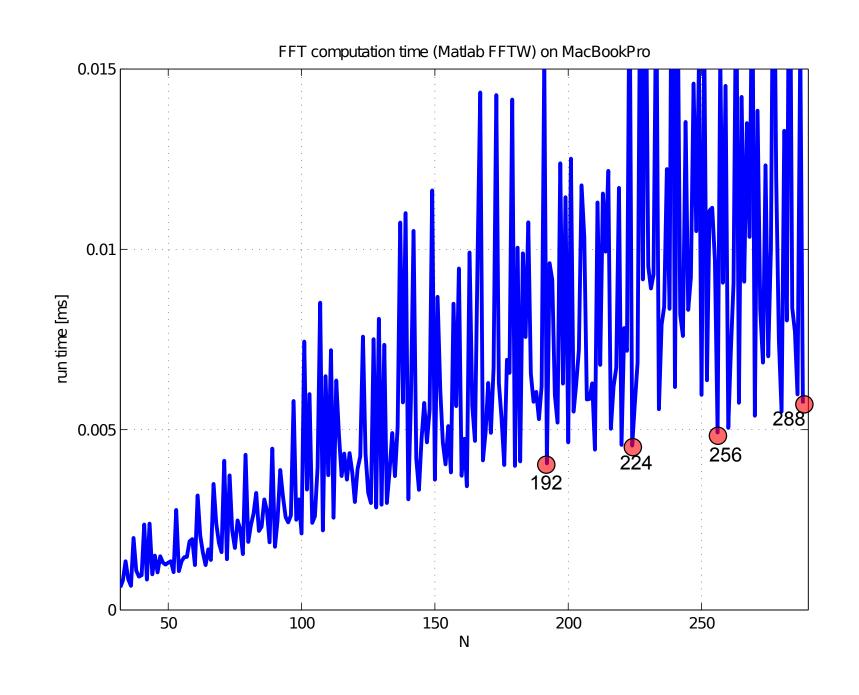
For example N = 693 factors into

$$N = (7)(9)(11)$$

each of which can be implemented efficiently. We would perform

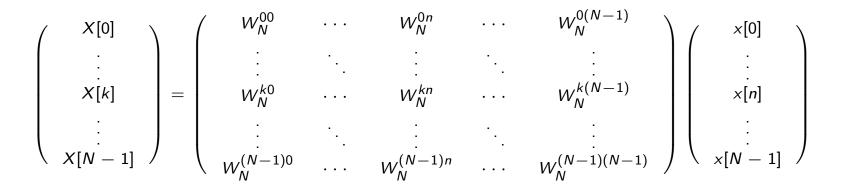
- $9 \times 11$  DFT's of length 7
- $7 \times 11$  DFT's of length 9, and
- $7 \times 9$  DFT's of length 11

- Historically, the power-of-two FFTs were much faster (better written and implemented).
- For non-power-of-two length, it was faster to zero pad to power of two.
- Recently this has changed. The free FFTW package implements very efficient algorithms for almost any filter length. Matlab has used FFTW since version 6



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### FFT as Matrix Operation



•  $W_N$  is fully populated  $\Rightarrow N^2$  entries.

## FFT as Matrix Operation

$$\begin{pmatrix} X[0] \\ \vdots \\ X[k] \\ \vdots \\ X[N-1] \end{pmatrix} = \begin{pmatrix} W_N^{00} & \cdots & W_N^{0n} & \cdots & W_N^{0(N-1)} \\ \vdots & \ddots & \vdots & \ddots & \vdots \\ W_N^{k0} & \cdots & W_N^{kn} & \cdots & W_N^{k(N-1)} \\ \vdots & \ddots & \vdots & \ddots & \vdots \\ W_N^{(N-1)0} & \cdots & W_N^{(N-1)n} & \cdots & W_N^{(N-1)(N-1)} \end{pmatrix} \begin{pmatrix} x[0] \\ \vdots \\ x[n] \\ \vdots \\ x[N-1] \end{pmatrix}$$

- $W_N$  is fully populated  $\Rightarrow N^2$  entries.
- FFT is a decomposition of  $W_N$  into a more sparse form:

$$F_{N} = \begin{bmatrix} I_{N/2} & D_{N/2} \\ I_{N/2} & -D_{N/2} \end{bmatrix} \begin{bmatrix} W_{N/2} & 0 \\ 0 & W_{N/2} \end{bmatrix} \begin{bmatrix} \text{Even-Odd Perm.} \\ \text{Matrix} \end{bmatrix}$$

• 
$$I_{N/2}$$
 is an identity matrix.  $D_{N/2}$  is a diagonal with entries   
1,  $W_N$ ,  $\cdots$ ,  $W_N^{N/2-1}$ 

#### Example: N = 4

$$F_4 = \begin{bmatrix} 1 & 0 & 1 & 0 \\ 0 & 1 & 0 & W_4 \\ 1 & 0 & -1 & 0 \\ 0 & 1 & 0 & -W_4 \end{bmatrix} \begin{bmatrix} 1 & 1 & 0 & 0 \\ 1 & -1 & 0 & 0 \\ 0 & 0 & 1 & 1 \\ 0 & 0 & 1 & -1 \end{bmatrix} \begin{bmatrix} 1 & 0 & 0 & 0 \\ 0 & 0 & 1 & 0 \\ 0 & 0 & 0 & 1 \end{bmatrix}$$

#### Beyond NlogN

- What if the signal x[n] has a k sparse frequency
  - A. Gilbert et. al, "Near-optimal sparse Fourier representations via sampling
  - H. Hassanieh et. al, "Nearly Optimal Sparse Fourier Transform"
  - Others.....
- O(K Log N) instead of O(N Log N)

