EDISP (NWL3) (English) Digital Signal Processing DFT Windowing, FFT

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DFT resolution 1

- ▶ N-point DFT \longrightarrow frequency sampled at $\theta_k = \frac{2\pi k}{N}$, so the resolution is f_s/N
- If we want more, we use $N_1 > N$ filling with zeros (zero-padding)
- but IDFT will give N₁-periodic signal
- and the spectrum will have sidelobes



DFT resolution 2a: 16pt DFT



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DFT resolution 2b: 64pt DFT (zero-padded 16pt signal)



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Limited observation time

For DFT we used to cut a fragment of the signal

$$x_0[n] = x[n]g[n]$$
, where $g[n] = \begin{cases} 1 & \text{for } n = 0, 1, ..., N-1 \\ 0 & \text{for } & \text{other } n \end{cases}$

g[n] is a window function. Here - a *boxcar window* Window effect:

- selection of a signal fragment
- ► $x[n] \cdot g[n]$ in time $\longrightarrow X(\theta) * G(\theta)$ in spectral domain \longrightarrow sidelobes or spectral leakage



Windowing a pure cosine

Example to be done on slide, temporarily on blackboard (:-).

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Leakage example



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Window (apodization) functions



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Raised cosine window family

► Hann window: Julius von Hann, 1839 – 1921, Austrian meteorologist; *hanning* is a verb form (*to hann*) $w(n) = 0.5 \left(1 - \cos\left(\frac{2\pi n}{N-1}\right)\right)$

- ► Hamming window: Richard Hamming, 1915 1998, American mathematician; $w(n) = 0.53836 0.46164 \cos\left(\frac{2\pi n}{N-1}\right)$
- Blackman window $w(n) = 0.42 0.5 \cos\left(\frac{2\pi n}{N-1}\right) + 0.08 \cos\left(\frac{4\pi n}{N-1}\right)$

Kaiser window

(D. Slepian, H.O. Pollak, H.J. Landau, around 1961, Prolate spheroidal wave functions ...)

- time limited sequence with energy concentrated in finite frequency interval
- a family of windows with many degrees of freedom
- Kaiser (1974) an approximation to optimal window: standard method to compute the optimal window was numerically ill-conditioned.

$$w_n = egin{cases} rac{l_0 \left(lpha \sqrt{1 - \left(rac{2n}{N} - 1
ight)^2}
ight)}{l_0 (lpha)} & ext{if } 0 \leq n \leq N \ 0 & ext{otherwise} \end{cases}$$

 I_0 – zeroth order modified Bessel function of the first kind,

- α (real number) determines the shape of the window:
 - α = 0 gives Boxcar,
 - $\alpha = 4$ gives -30 dB first sidelobe, -50 asymptotic,
 - $\alpha = 8$ gives -60 dB first sidelobe, -90 asymptotic,

Kaiser window



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Fast DFT algorithms \longrightarrow FFT

► Direct computation with pre-computed *twiddle factors* $W_N^{kn} = (W_N)^{kn} = (e^{-j2\pi/N})^{kn}$

$$X\left(e^{j\theta_{k}}\right) = \sum_{n=0}^{N-1} x(n) (W_{N})^{kn}$$

 \longrightarrow complexity: N^2 complex multiplications & additions

• Goertzel algorithm: $X(k) = y_k(N)$, where

$$y_k(n) = \sum_{r=0}^{N-1} x(r) W_N^{-k(n-r)}$$

 \rightarrow filtering: $y_k(n) = x(n) + y_k(n-1) \cdot W_N^{-k}$ Also N^2 , but after decomposition majority is real×real (see next slide). Useful when not all *N* frequencies are needed.

▶ Divide-by-two (or decimation) in time \longrightarrow FFT algorithm, complexity $N \log_2(N)$

Goertzel algorithm (1958)

Calculate a single sample of DFT (at $\omega=\omega_{\textit{k}})$ by filtering

Gerald Goertzel (1919 – 2002), theoretical physicist, worked with Manhattan Project and later Sage Instruments and IBM

- A convolution with sinusoid: $s(n) = x(n) + 2\cos(\theta_k)s(n-1) s(n-2)$
- After N samples X(k) is computed as $X(k) = y(N) = s(n) e^{-j\theta_k}s(n-1)$



Fast DFT algorithms \longrightarrow FFT

Decimation in time **FFT** (first stage):

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

= $\sum_{n \in V(n)} x(n) W_N^{nk} + \sum_{n \in V(n)} x(n) W_N^{nk} =$
= $\sum_{r=0}^{N/2-1} x(2r) (W_{N/2})^{rk} + W_N^k \sum_{r=0}^{N/2-1} x(2r+1) (W_{N/2})^{rk}$

radix-2 FFT

$$X(k) = \sum_{n \in \text{ven}} x(n) W_N^{nk} + \sum_{n \text{odd}} x(n) W_N^{nk} =$$
$$\sum_{r=0}^{l/2-1} x(2r) (W_{N/2})^{rk} + W_N^k \sum_{r=0}^{N/2-1} x(2r+1) (W_{N/2})^{rk}$$

- ► If $N = 2^L$... We can continue with this trick decimating each half into sub-halves, each sub-half into sub-sub... *L* times
- For k > N/2, W^k_N = −W^{k−N/2}_N and FT_{N/2} is periodic with period N/2
- DFT with size 1 is rather trivial

Ν

Effect: We have *L* layers of N/2 butterflies. Each butterfly is one multiplication, one addition, one subtraction. In the result, we have $O(N\log_2 N)$ operations

FFT inventors

James W. Cooley and John W. Tukey, "An algorithm for the machine calculation of complex Fourier series," Math. Comput. 19, pp. 297-301 (1965).







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Indexing for FFT





 \rightarrow bit-reversal does the job!

Processors designed for FFT do have the bit-reversal mode of indexing. (And they do a butterfly in one or two cycles)

Decimation in frequency FFT

• We split the definition formula for k even (=2r) or odd (=2r+1)

• We note that
$$W_N^{2nr} = W_{N/2}^{nr}$$
 or $W_N^{n(2r+1)} = W_N^n \cdot W_{N/2}^{nr}$

• Further, for
$$n > N/2$$
 $W_N^n = -W_N^{n-N/2}$

and so on - please sketch the DIF FFT diagram by yourselves

 \longrightarrow here, we need to re-index the frequencies...

Specials

- Non-radix2 FFT slower than radix2, but still faster than direct
- Chirp-z transform one use of it is to calculate FT for θ 's not equal to $2\pi/N$
- Non-uniform FFT ...
- FFTW the Fastest FFT in the West a free library, used by many free and commercial products (Frigo & Johnson from MIT)

FFTS - in the South - (New Zealand)

Summary

Fourier transforms:

DTFT - spectrum of a discrete-time signal (defined for a limited-energy signal or a limited mean power signal in a different manner) *periodic, continuous or discrete function of* θ

- ▶ DFT samples of DTFT of a limited duration signal (or a segment....) periodic, discrete X(k)
- FFT a trick (method[s]) to compute DFT efficiently

To window or not to window?

- ▶ If we need to *analyse* the signal YES,
- If we need to manipulate spectrum and then reconstruct the signal back NO.