

Lecture 9

based on slides by J.M. Kahn

M. Lustig, EECS UC Berkeley

Announcements

- Lab 01 part I and II posted will post III today or tomorrow
- Lab-bash Tuesday 2-3pm 521 Cory
- Three shorter Midterms:
 - 02/26 in class
 - -04/02 in class
 - 04/30 (or 28 TBD) in class
 - 05/05 or 05/06 (TBD) project presentations.
 - Posters and demos

M. Lustig, EECS UC Berkele

Announcements

- Last time:
 - -Frequency analysis with DFT
 - -Windowing
- Today:
 - Continue
 - Effect of zero-padding
 - Start Short-time Fourier Transform

Windows Properties

These are characteristic of the window type

Window	Main-lobe	Sidelobe δ_s	Sidelobe $-20\log_{10}\delta_s$
Rect	$\frac{4\pi}{M+1}$	0.09	21
Bartlett	$\frac{8\pi}{M+1}$	0.05	26
Hann	$\frac{8\pi}{M+1}$	0.0063	44
Hamming	$\frac{8\pi}{M+1}$	0.0022	53
Blackman	$\frac{12\pi}{M+1}$	0.0002	74

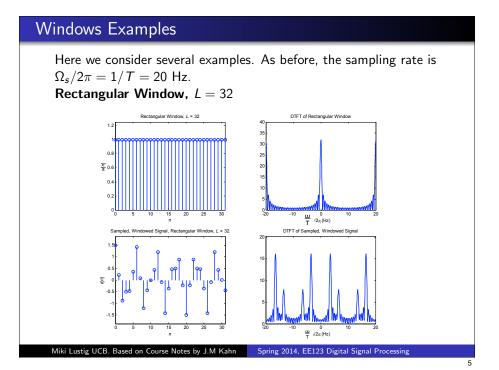
Most of these (Bartlett, Hann, Hamming) have a transition width that is twice that of the rect window.

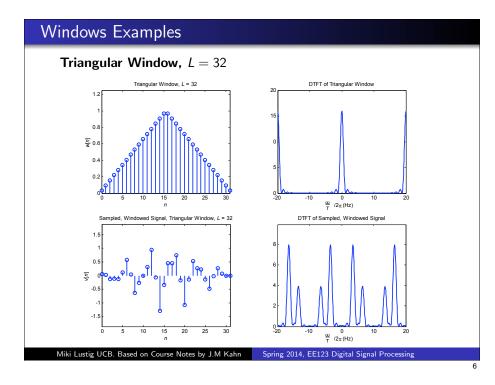
Warning: Always check what's the definition of M

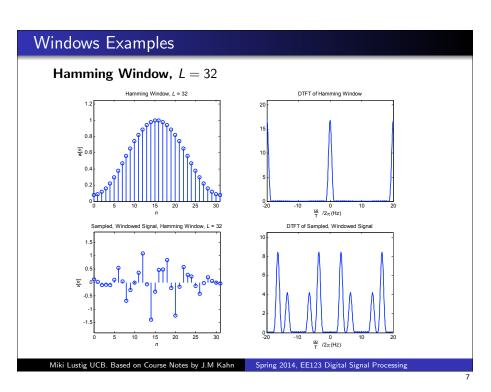
Adapted from A Course In Digital Signal Processing by Boaz Porat, Wiley, 1997

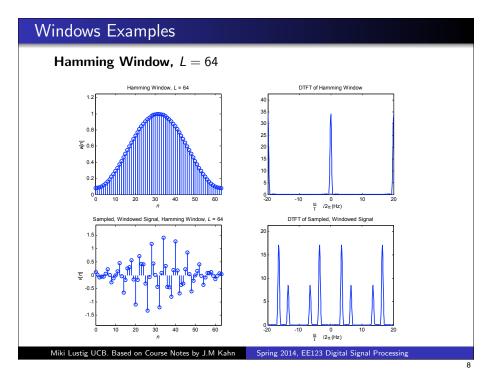
M. Lustig, EECS UC Berkeley

Miki Lustig UCB. Based on Course Notes by J.M Kahn Spring 2014, EE123 Digital Signal Processing









Optimal Window: Kaiser

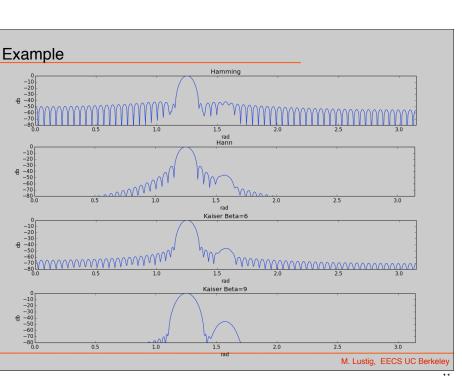
 Minimum main-lobe width for a given sidelobe energy %

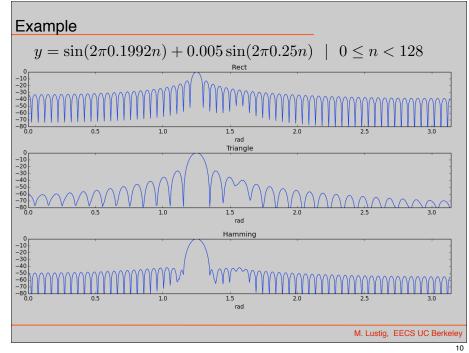
$$\frac{\int_{\text{sidelobes}} |H(e^{j\omega})|^2 d\omega}{\int_{-\pi}^{\pi} |H(e^{j\omega})|^2 d\omega}$$

- Window is parametrized with L and β OS Eq 10.12
 - β determines side-lobe level
 - L determines main-lobe width

M. Lustig, EECS UC Berkeley







Zero-Padding

• In preparation for taking an N-point DFT, we may zero-pad the windowed block of signal samples to a block length $N \ge L$:

$$\begin{cases} v[n] & 0 \le n \le L - 1 \\ 0 & L \le n \le N - 1 \end{cases}$$

• This zero-padding has no effect on the DTFT of v[n], since the DTFT is computed by summing over $-\infty < n < \infty$.

Effect of Zero Padding

• We take the *N*-point DFT of the zero-padded v[n], to obtain the block of *N* spectral samples:

$$V[k], \quad 0 \le k \le N-1$$

Miki Lustig UCB. Based on Course Notes by J.M Kahr

pring 2014, EE123 Digital Signal Processin

Zero-Padding

• Consider the DTFT of the zero-padded v[n]. Since the zero-padded v[n] is of length N, its DTFT can be written:

$$V(e^{j\omega}) = \sum_{n=0}^{N-1} v[n]e^{-jn\omega}, \quad -\infty < \omega < \infty$$

The *N*-point DFT of v[n] is given by:

$$V[k] = \sum_{n=0}^{N-1} v[n] W_N^{kn} = \sum_{n=0}^{N-1} v[n] e^{-j(2\pi/N)nk}, \quad 0 \le k \le N-1$$

We see that V[k] corresponds to the samples of $V(e^{j\omega})$:

$$V[k] = V(e^{j\omega})\big|_{\omega=k\frac{2\pi}{N}}, \quad 0 \le k \le N-1$$

To obtain samples at more closely spaced frequencies, we zero-pad v[n] to longer block length N. The spectrum is the same, we just have more samples.

Miki Lustig UCB. Based on Course Notes by J.M Kahn Spring 2014, EE123 Digital Signal Processing

Frequency Analysis with DFT

Note that the ordering of the DFT samples is unusual.

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

The DC sample of the DFT is k = 0

$$V[0] = \sum_{n=0}^{N-1} v[n] W_N^{0n} = \sum_{n=0}^{N-1} v[n]$$

- The positive frequencies are the first N/2 samples
- The first N/2 negative frequencies are circularly shifted

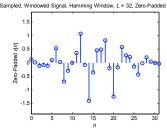
$$((-k))_N = N - k$$

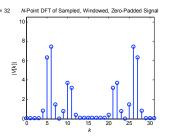
so they are the last N/2 samples. (Use fftshift to reorder)

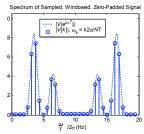
Miki Lustig UCB. Based on Course Notes by J.M Kahn Spring 2014, EE123 Digital Signal Processing

Frequency Analysis with DFT Examples:

Hamming Window, L = 32, N = 32



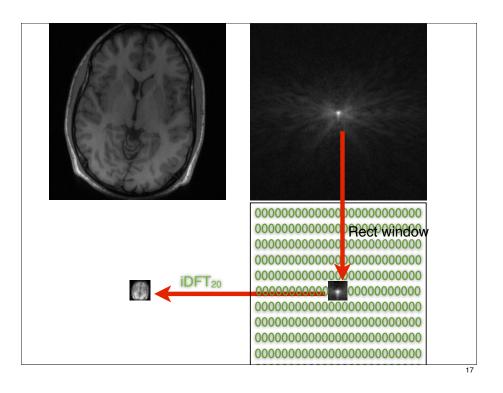


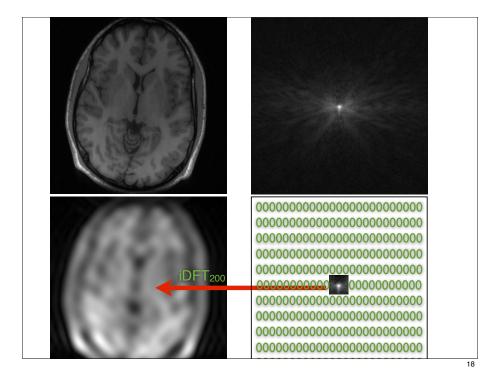


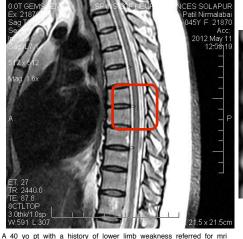
Spring 2014, EE123 Digital Signal Processing

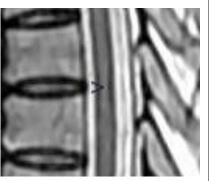


Frequency Analysis with DFT Examples: Hamming Window, L=32, Zero-Padded to N=64Sampled, Windowed Signal, Hamming Window, L = 32, Zero-Padded to N = 64 N-Point DFT of Sampled, Windowed, Zero-Padded Signal









A 40 yo pt with a history of lower limb weakness referred for mri screening of brain and whole spine for cord. MRI sagittal T2 screening of dorsal region shows a faint uniform linear high signal at the center of the cord. The signal abnormality likely to represent:

- (1) Cord demyelination.
- (2) Syrinx (spinal cord disease).

Answer: Its an artifact, known as truncation or Gibbs artifact

http://www.neuroradiologycases.com

Frequency Analysis with DFT

- Length of window determines spectral resolution
- Type of window determines side-lobe amplitude. (Some windows have better tradeoff between resolution-sidelobe)
- Zero-padding approximates the DTFT better. Does not introduce new information!

Miki Lustig UCB. Based on Course Notes by J.M Kahn Spring 2014, EE123 Digital Signal Processing

Potential Problems and Solutions

Potential Problems and Solutions

Problem	Possible Solutions		
1. Spectral error from aliasing Ch.4	a. Filter signal to reduce frequency content above $\Omega_s/2=\pi/T$. b. Increase sampling frequency $\Omega_s=2\pi/T$.		
2. Insufficient frequency resolution.	a. Increase <i>L</i> b. Use window having narrow main lobe.		
3. Spectral error from leakage	a. Use window having low side lobes. b. Increase <i>L</i>		
4. Missing features due to spectral sampling.	a. Increase L , b. Increase N by zero-padding $v[n]$ to length $N>L$.		

Miki Lustig UCB. Based on Course Notes by J.M Kahn

Spring 2014, EE123 Digital Signal Processing

iSpectrum Example

Discrete Transforms (Finite)

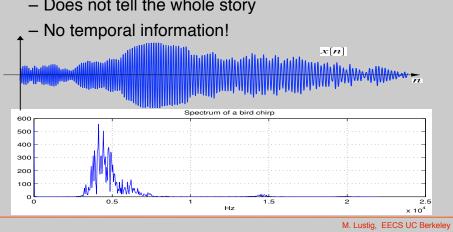
- · DFT is only one out of a LARGE class of transforms
- Used for:
 - -Analysis
 - -Compression
 - -Denoising
 - -Detection
 - -Recognition
 - -Approximation (Sparse)

Sparse representation has been one of the hottest research topics in the last 15 years in sp

M. Lustig, EECS UC Berkeley

Example of spectral analysis

- Spectrum of a bird chirping
 - Interesting,.... but...
 - Does not tell the whole story



Time Dependent Fourier Transform

 To get temporal information, use part of the signal around every time point

$$X[n,\omega) = \sum_{m=-\infty}^{\infty} x[n+m]w[m]e^{-j\omega m}$$

*Also called Short-time Fourier Transform (STFT)

- Mapping from 1D \Rightarrow 2D, n discrete, w cont.
- Simply slide a window and compute DTFT

M. Lustig, EECS UC Berkeley

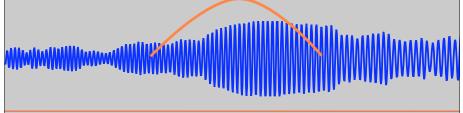
25



 To get temporal information, use part of the signal around every time point

$$X[n,\omega) = \sum_{m=-\infty}^{\infty} x[n+m]w[m]e^{-j\omega m}$$

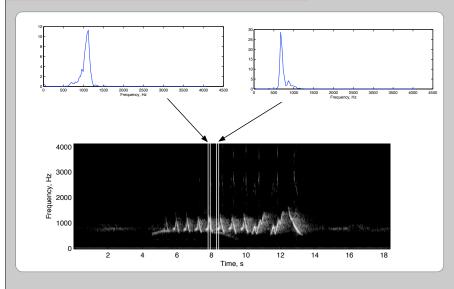
*Also called Short-time Fourier Transform (STFT)



M. Lustig, EECS UC Berkeley

26

Spectrogram



Discrete Time Dependent FT

$$X_r[k] = \sum_{m=0}^{L-1} x[rR + m]w[m]e^{-j2\pi km/N}$$

- L Window length
- R Jump of samples
- N DFT length
- Tradeoff between time and frequency resolution

M. Lustig, EECS UC Berkeley

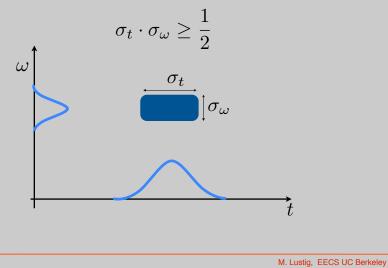
27

M. Lustig, EECS UC Berkeley

Heisenberg Boxes



• Time-Frequency uncertainty principle



 $X[k] = \sum_{n=0}^{N-1} x[n]e^{-j2\pi kn/N}$ $\Delta \omega = \frac{2\pi}{N}$ $\Delta t = N$ $\Delta \omega \cdot \Delta t = 2\pi$ One DFT coefficient

30