ESE 531: Digital Signal Processing

Lec 22: April 16, 2020 Spectral Analysis





- Spectral Analysis with DFT
- Windowing
- Effect of zero-padding
- Time-dependent Fourier transform
 - Aka short-time Fourier transform

Spectral Analysis Using the DFT

DFT is a tool for spectrum analysis

- Find out what frequencies are in your signal
- Should be simple:
 - Take a block, compute spectrum with DFT

Spectral Analysis Using the DFT

- DFT is a tool for spectrum analysis
 - Find out what frequencies are in your signal
- Should be simple:
 - Take a block, compute spectrum with DFT
- □ But, there are issues and tradeoffs:
 - Signal duration vs spectral resolution
 - Sampling rate vs spectral range
 - Spectral sampling rate
 - Spectral artifacts



□ Steps for processing continuous time (CT) signals



Spectral Analysis Using the DFT

• Two important tools:

- Applying a window \rightarrow reduced artifacts
- Zero-padding \rightarrow increases spectral sampling

Parameter	Symbol	Units
Sampling interval	T	S
Sampling frequency	$\Omega_s = rac{2\pi}{T}$	rad/s
Window length	L	unitless
Window duration	$L \cdot T$	S
DFT length	$N \ge L$	unitless
DFT duration	$N \cdot T$	S
Spectral resolution	$\frac{\Omega_s}{I} = \frac{2\pi}{I \cdot T}$	rad/s
Spectral sampling interval	$\frac{\overline{\Omega_s}}{N} = \frac{\overline{2\pi}}{N \cdot T}$	rad/s



$$\begin{aligned} x_c(t) &= A_1 \cos(\omega_1 t) + A_2 \cos(\omega_2 t) \\ X_c(j\Omega) &= A_1 \pi \Big[\delta(\Omega - \omega_1) + \delta(\Omega + \omega_1) \Big] + A_2 \pi \Big[\delta(\Omega - \omega_2) + \delta(\Omega + \omega_2) \Big] \end{aligned}$$



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FT of Original CT Signal (heights represent areas of $\delta(\Omega)$ impulses)



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Sampled CT Signal Example

□ If we sample the signal over an infinite time duration, we would have:

$$x[n] = x_c(t)\Big|_{t=nT}, \quad -\infty < n < \infty$$

□ With the discrete time Fourier transform (DTFT):

$$X(e^{j\Omega T}) = \frac{1}{T} \sum_{r=-\infty}^{\infty} X_c \left(j \left(\Omega - r \frac{2\pi}{T} \right) \right), \quad -\infty < \Omega < \infty$$



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FT of Original CT Signal (heights represent areas of $\delta(\Omega)$ impulses)



Sampled CT Signal Example

• Sampling with $\Omega s/2\pi = 1/T = 20Hz$





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Windowed Sampled CT Signal

In any real system, we sample only over a finite block of L samples:

$$x[n] = x_c(t) \Big|_{t=nT}, \quad 0 < n < L-1$$

- This simply corresponds to a rectangular window of duration L
- Recall there are many other window types
 - Hann, Hamming, Blackman, Kaiser, etc.







Name(s)	Definition	MATLAB Command	Graph (<i>M</i> = 8)
Hann	$w[n] = \begin{cases} \frac{1}{2} \left[1 + \cos\left(\frac{\pi n}{M/2}\right) \right] & n \le M/2 \\ 0 & n > M/2 \end{cases}$	hann (M+1)	hann(M+1), M = 8
Hanning	$w[n] = \begin{cases} \frac{1}{2} \left[1 + \cos\left(\frac{\pi n}{M/2 + 1}\right) \right] & n \le M/2 \\ 0 & n > M/2 \end{cases}$	hanning (M+1)	hanning(M+1), $M = 8$
Hamming	$w[n] = \begin{cases} 0.54 + 0.46 \cos\left(\frac{\pi n}{M/2}\right) & n \le M/2 \\ 0 & n > M/2 \end{cases}$	hamming (M+1)	hamming(M+1), $M = 8$ $\begin{array}{c} 1 \\ 0.8 \\ \overline{5} & 0.6 \\ 0.4 \\ 0.2 \\ 0 \\ 5 & 0 \\ 5 & 0 \\ 5 & 0 \\ 7 & 5 \\ n \\ 7 \\ $





Windowed Sampled CT Signal

 We take the block of signal samples and multiply by a window of duration L, obtaining:

$$v[n] = x[n] \cdot w[n], \quad 0 < n < L-1$$

If the window w[n] has DTFT, W(e^{jω}), then the windowed block of signal samples has a DTFT given by the periodic convolution between X(e^{jω}) and W(e^{jω}):

$$V(e^{j\omega}) = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\theta}) W(e^{j(\omega-\theta)}) d\theta$$

Windowed Sampled CT Signal

- Convolution with W(e^{jω}) has two effects in the spectrum:
 - It limits the spectral resolution (spectral spreading)
 - Main lobes of the DTFT of the window
 - The window can produce spectral leakage
 - Side lobes of the DTFT of the window
- □ These two are always a tradeoff
 - time-frequency uncertainty principle
 - More later...

Sampled CT Signal Example

• Sampling with $\Omega s/2\pi = 1/T = 20Hz$





DTFT of Sampled Signal (heights represent areas of $\delta(\alpha)$ impulses)

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- As before, the sampling rate is $\Omega s/2\pi = 1/T = 20Hz$
- **\Box** Rectangular Window, L = 32



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□ Hamming Window, L = 32



• As before, the sampling rate is $\Omega s/2\pi = 1/T = 20Hz$

□ Hamming Window, L = 64



Optimal Window: Kaiser

 Minimum main-lobe width for a given sidelobe energy percentage

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

- Window is parameterized with L and β
 - β determines side-lobe level
 - L determines main-lobe width



$y[n] = \sin(2\pi 0.1992n) + 0.005\sin(2\pi 0.25n) | 0 \le n \le 128$



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In preparation for taking an N-point DFT, we may zero-pad the windowed block of signal samples

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

- This zero-padding has no effect on the DTFT of v[n], since the DTFT is computed by summing over infinity
- Effect of Zero Padding
 - We take the N-point DFT of the zero-padded v[n], to obtain the block of N spectral samples:



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

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

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

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

• We know that the DFT is a sample $V(e^{j\omega})$:

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

Frequency Analysis with DFT

• Hamming window, L = N = 32



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\Box Hamming window, L = 32, Zero-padded to N = 64

|V[k]|

k

Sampled, Windowed Signal, Hamming Window, L = 32, Zero-Padded to N = 64

N-Point DFT of Sampled, Windowed, Zero-Padded Signal



Spectrum of Sampled, Windowed, Zero-Padded Signal



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Frequency Analysis with DFT

- □ Length of window determines spectral resolution
- Type of window determines side-lobe amplitude/ main-lobe width (spectral leakage/spreading)
 - Some windows have better tradeoff between resolution and side-lobe height
- Zero-padding approximates the DTFT better (spectral sampling). Does not introduce new information!

Potential Problems and Solutions

- □ 1. Spectral error
 - 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 L
 - b. Use window having narrow main lobe.
- □ 3. Spectral error from leakage
 - a. Use window having low side lobes.
 - b. Increase L
- 4. Missing features due to spectral sampling
 - a. Increase L
 - b. Increase N by zero-padding v[n] to length N > L

Time Dependent DFT



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

Example of Spectral Analysis

• Spectrum of a bird chirping

Interesting,.... but...

- Does not tell the whole story
- No temporal information!





https://dogparksoftware.com/iSpectrum.html



- Also called short-time Fourier transform
- To get temporal information, use part of the signal around every time point

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

Mapping from 1D → 2D, n discrete, λ cont.
 Simply slide a window and compute DTFT











D Plotting $Y[n, \lambda)$

$$y[n] = \begin{cases} 0 & n < 0\\ \cos(\alpha_0 n^2) & 0 \le n \le 20,000\\ \cos(0.2\pi n) & 20,000 < n \le 25,000\\ \cos(0.2\pi n) + \cos(0.23\pi n) & 25,000 < n. \end{cases}$$



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Discrete Time-Dependent Fourier Transform

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

$$X[rR,k] = X[rR,2\pi k / N] = \sum_{m=0}^{L-1} x[rR+m]w[m]e^{-j(2\pi/N)km}$$

- **L** Window length
- **R** Jump of samples
- □ N DFT length

Discrete Time-Dependent Fourier Transform

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$$X[rR,k] = X[rR,2\pi k / N) = \sum_{m=0}^{L-1} x[rR+m]w[m]e^{-j(2\pi/N)km}$$

$$X_{r}[k] = \sum_{m=0}^{L-1} x[rR + m]w[m]e^{-j(2\pi/N)km}$$

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□ https://www.youtube.com/watch?v=vvr9AMWEU-c













What is the difference between the spectrograms?
a) Window size B<A
b) Window size B>A









What is the difference between the spectrograms?
a) Window size B<A
b) Window size B>A







□ FSK Communications

• Spectrogram transmitting 'H' (ASCII H = 01001000)





□ If $R \le L \le N$, then we can recover x[n] block-by-block from $X_r[k]$

□ For non-overlapping windows, R=L

$$x_{r}[m] = \frac{1}{N} \sum_{k=0}^{N-1} X_{r}[k] e^{j2\pi km/N}$$



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$$x_{r}[m] = \frac{1}{N} \sum_{k=0}^{N-1} X_{r}[k] e^{j2\pi km/N}$$

$$x[n] = \frac{x_r[n - rR]}{w[n - rR]} \qquad \forall rR \le n \le (r+1)R - 1$$

SFTF Reconstruction with overlap

- □ Practically make R<L<N
- If we choose R, L, and N appropriately with window, the overlap-add will negate the window effects





□ Frequency analysis with DFT

- Nontrivial to choose sampling frequency, signal length, window type, DFT length (zero-padding)
- Get accurate representation of DFT
- Time-dependent Fourier transform
 - Aka short-time Fourier transform
 - Includes temporal information about signal
 - Useful for many applications
 - Analysis, Compression, Denoising, Detection, Recognition, Approximation (Sparse)
 - Overlap for reconstruction



Project

- Due 4/28
- Rubric in Canvas