Spectral Analysis of Signals

- Spectral analysis is concerned with the determination of frequency contents of a continuous-time signal $g_a(t)$ using DSP methods
- It involves the determination of either the energy spectrum or the power spectrum of the signal
- If $g_a(t)$ is sufficiently bandlimited, spectral characteristics of its discrete-time equivalent g[n] should provide a good estimate of spectral characteristics of $g_a(t)$

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Spectral Analysis of Signals

- In most cases, $g_a(t)$ is defined for $-\infty < t < \infty$
- Thus, g[n] is of infinite extent, and defined for $-\infty < n < \infty$
- Hence, g_a(t) is first passed through an analog anti-aliasing filter whose output is then sampled to generate g[n]
- <u>Assumptions</u>: (1) Effect of aliasing can be ignored, (2) A/D conversion noise can be neglected

Spectral Analysis of Signals

- Three types of spectral analysis -
- 1) Spectral analysis of stationary sinusoidal signals
- 2) Spectral analysis of of nonstationary signals with time-varying parameters
- 3) Spectral analysis of random signals

- <u>Assumption</u> Parameters characterizing sinusoidal signals, such as amplitude, frequencies, and phase, do not change with time
- For such a signal *g*[*n*], the Fourier analysis can be carried out by computing the DTFT

$$G(e^{j\omega}) = \sum_{n=-\infty}^{\infty} g[n]e^{-j\omega n}$$

- In practice, the infinite-length sequence g[n] is first windowed by multiplying it with a length-N window w[n] to convert it into a length-N sequence γ[n]
- DTFT $\Gamma(e^{j\omega})$ of $\gamma[n]$ then is assumed to provide a reasonable estimate of $G(e^{j\omega})$
- Γ(e^{jω}) is evaluated at a set of R (R≥N) discrete angular frequencies equally spaced in the range 0≤ω≤2π by computing the *R*-point FFT Γ[k] of γ[n]

- We analyze the effect of windowing and the evaluation of the frequency samples of the DTFT via the DFT
- Now

$$\Gamma[k] = \Gamma(e^{j\omega})\Big|_{\omega = 2\pi k/R}, \quad 0 \le k \le R-1$$

• The normalized discrete-time angular frequency ω_k corresponding to the DFT bin number k (DFT frequency) is given by

$$\omega_k = \frac{2\pi k}{R}$$

• The continuous-time angular frequency corresponding to the DFT bin number *k* (DFT frequency) is given by

$$\Omega_k = \frac{2\pi k}{RT}$$

• To interpret the results of DFT-based spectral analysis correctly we first consider the frequency-domain analysis of a sinusoidal signal

- Consider $g[n] = \cos(\omega_o n + \phi), -\infty < n < \infty$
- It can be expressed as $g[n] = \frac{1}{2} \left(e^{j(\omega_o n + \phi)} + e^{-j(\omega_o n + \phi)} \right)$
- Its DTFT is given by $G(e^{j\omega}) = \pi \sum_{\substack{\ell = -\infty \\ \ell = -\infty}}^{\infty} e^{j\phi} \delta(\omega - \omega_o + 2\pi\ell) + \pi \sum_{\substack{\ell = -\infty \\ \ell = -\infty}}^{\infty} e^{-j\phi} \delta(\omega - \omega_o + 2\pi\ell)$

- $G(e^{j\omega})$ is a periodic function of ω with a period 2π containing two impulses in each period
- In the range $-\pi \le \omega \le \pi$, there is an impulse at $\omega = \omega_o$ of complex amplitude $\pi e^{j\phi}$ and an impulse at $\omega = -\omega_o$ of complex amplitude $\pi e^{-j\phi}$
- To analyze g[n] using DFT, we employ a finite-length version of the sequence given by $\gamma[n] = \cos(\omega_o n + \phi), \ 0 \le n \le N 1$

- Example Determine the 32-point DFT of a length-32 sequence g[n] obtained by sampling at a rate of 64 Hz a sinusoidal signal $g_a(t)$ of frequency 10 Hz
- Since $F_T = 64$ Hz is much larger than the Nyquist frequency of 20 Hz, there is no aliasing due to sampling

• DFT magnitude plot

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• Since $\gamma[n]$ is a pure sinusoid, its DTFT $\Gamma(e^{j2\pi f})$ contains two impulses at $f = \pm 10$ Hz and is zero everywhere else

- Its 32-point DFT is obtained by sampling $\Gamma(e^{j2\pi f})$ at $f = 64 \times 2/32 = 2k \ Hz, 0 \le k \le 31$
- The impulse at *f* = 10 Hz appears as Γ[5] at the DFT frequency bin location

$$k = \frac{fR}{F_T} = \frac{10 \times 32}{64} = 5$$

and the impulse at f = -10 Hz appears as $\Gamma[27]$ at bin location k = 32 - 5 = 27

• <u>Note</u>: For an *N*-point DFT, first half DFT samples for k = 0 to k = (N/2) - 1corresponds to the positive frequency axis from f = 0 to $f = F_T/2$ excluding the point $f = F_T/2$ and the second half for k = N/2 to k = N - 1 corresponds to the negative frequency axis from $f = -F_T/2$ to f = 0excluding the point f = 0

- Example Determine the 32-point DFT of a length-32 sequence $\gamma[n]$ obtained by sampling at a rate of 64 Hz a sinusoidal signal $x_a(t)$ of frequency 11 Hz
- Since $\gamma[n]$ is a pure sinusoid, its DTFT $\Gamma(e^{j2\pi f})$ contains two impulses at $f = \pm 11$ Hz and is zero everywhere else

• Since $\frac{fR}{F_T} = \frac{11 \times 32}{64} = 5.5$

the impulse at f = 11 Hz of the DTFT appear between the DFT bin locations k = 5and k = 6

• Likewise, the impulse at f = -11 Hz of the DTFT appear between the DFT bin locations k = 26 and k = 27

• DFT magnitude plot



• <u>Note</u>: Spectrum contains frequency components at all bins, with two strong components at *k* = 5 and *k* = 6, and two strong components at *k* = 26 and *k* = 27

- The phenomenon of the spread of energy from a single frequency to many DFT frequency locations is called **leakage**
- To understand the cause of leakage, recall that the *N*-point DFT $\Gamma[k]$ of a length-*N* sequence $\gamma[n]$ is given by the samples of its DTFT $\Gamma(e^{j\omega})$:

$$\Gamma[k] = \Gamma(e^{j\omega_k})\Big|_{\omega_k = 2\pi k/N}, \ 0 \le k \le N-1$$

 Plot of the DTFT of the length-32 sinusoidal sequence of frequency 11 Hz sampled at 64 Hz is shown below along with its 32-point DFT



- The DFT samples are indeed obtained by the frequency samples of the DTFT
- Now the sequence

 $\gamma[n] = \cos(\omega_o n + \phi), \ 0 \le n \le N - 1$

has been obtained by windowing the infinite-length sinusoidal sequence g[n]with a rectangular window w[n]:

$$w[n] = \begin{cases} 1, & 0 \le n \le N-1 \\ 0, & \text{otherwise} \end{cases}$$

Spectral Analysis of
Sinusoidal Signals
The DTFT
$$\Gamma(e^{j\omega})$$
 of $\gamma[n]$ is given by

$$\Gamma(e^{j\omega}) = \frac{1}{2\pi} \int_{-\pi}^{\pi} G(e^{j\varphi}) \Psi(e^{j(\omega-\varphi)}) d\varphi$$
where $G(e^{j\omega})$ is the DTFT of $g[n]$:

$$G(e^{j\omega}) = \pi \sum_{\ell=-\infty}^{\infty} e^{j\phi} \delta(\omega - \omega_o + 2\pi\ell)$$

$$\ell = -\infty + \pi \sum_{\ell=-\infty}^{\infty} e^{-j\phi} \delta(\omega + \omega_o + 2\pi\ell)$$
and $\Psi(e^{j\omega})$ is the DTFT of $w[n]$:

$$\Psi(e^{j\omega}) = e^{-j\omega(N-1)/2} \frac{\sin(\omega N/2)}{\sin(\omega/2)}$$

• Hence

$$\Gamma(e^{j\omega}) = \frac{1}{2}e^{j\phi}\Psi(e^{j(\omega-\omega_o)}) + \frac{1}{2}e^{-j\phi}\Psi(e^{j(\omega+\omega_o)})$$

- Thus, $\Gamma(e^{j\omega})$ is a sum of frequency shifted and amplitude scaled DTFT $\Psi(e^{j\omega})$ with the amount of shifts given by $\pm \omega_o$
- For the length-32 sinusoid of frequency 11 Hz sampled at 64 Hz, normalized angular frequency of the sinusoid is 11/64 = 0.172

- Its DTFT $\Gamma(e^{j\omega})$ is obtained by frequency shifting the DTFT $\Psi(e^{j\omega})$ to the right and to the left by the amount $0.172 \times 2\pi = 0.344\pi$, adding both shifted versions and scaling the sum by a factor of 1/2
- In the frequency range $0 \le \omega \le 2\pi$, which is one period of the DTFT, there are two peaks, one at 0.344π and the other at $2\pi(1-0.172)$ = 1.656π

• Plot of $|\Gamma(e^{j\omega})|$ and the 32-point DFT $|\Gamma[k]|$



- The two peaks of |Γ[k]| located at bin locations k = 5 and k = 6 are frequency samples on both sides of the main lobe located at 0.172
- The two peaks of |Γ[k]| located at bin locations k = 26 and k = 27 are frequency samples on both sides of the main lobe located at 0.828

- All other DFT samples are given by the samples of the sidelobes of Ψ(e^{jω}) causing the leakage of the frequency components at to other bin locations with the amount of leakage determined by the relative amplitudes of the main lobe and the sidelobes
- Since the relative sidelobe level $A_{s\ell}$ of the rectangular window is very high, there is a considerable amount of leakage to the bin locations adjacent to the main lobes

- Problem gets more complicated if the signal being analyzed has more than one sinusoid
- We now examine the effects of the length *R* of the DFT, the type of window being used, and its length *N* on the results of spectral analysis
- Consider

$$x[n] = \frac{1}{2}\sin(2\pi f_1 n) + \sin(2\pi f_2 n), \quad 0 \le n \le N - 1$$

Spectral Analysis of Sinusoidal Signals • Example - N = 16, $f_1 = 0.22$, $f_2 = 0.34$



From the above plot it is difficult to determine whether there is one or more sinusoids in x[n] and the exact locations of the sinusoids



 An increase in the DFT length to R = 32 leads to some concentrations around k = 7 and k = 11 in the normalized frequency range from 0 to 0.5



• There are two clear peaks when R = 64



 An increase in the accuracy of the peak locations is obtained by increasing DFT length to R = 128 with peaks occurring at k = 27 and k = 45

- However, there are a number of minor peaks and it is not clear whether additional sinusoids of lesser strengths are present
- <u>General conclusion</u> An increase in the DFT length improves the sampling accuracy of the DTFT by reducing the spectral separation of adjacent DFT samples

- <u>Example</u> $N = 16, R = 128, f_2 = 0.34$
 - f_1 varied from 0.28 to 0.31 $f_1 = 0.28, f_2 = 0.34$ $f_1 = 0.29, f_2 = 0.34$ 10 10 8 8 6 6 X[k] X[k] 2 2 50 50 100 100k k
- The two sinusoids are clearly resolved in
 both cases

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• The two sinusoids cannot be resolved in both cases

- Reduced resolution occurs when the difference between the two frequencies becomes less than 0.4
- The DTFT $\Gamma(e^{j\omega})$ is obtained by summing the DTFTs of the two sinusoids
- As the difference between the two frequencies get smaller, the main lobes of the individual DTFTs get closer and eventually overlap

- If there is significant overlap, it will be difficult to resolve the two peaks
- Frequency resolution is determined by the width Δ_{ML} of the main lobe of the DTFT of the window
- For a length-*N* rectangular window $\Delta_{ML} = \frac{4\pi}{2M+1}$
- In terms of normalized frequency, for N = 16, main lobe width is 0.125

- Thus, two closely spaced sinusoids windowed by a length-16 rectangular window can be resolved if the difference in the frequencies is about half the main lobe width, i.e., 0.0625
- Rectangular window has the smallest main lobe width and has the smallest frequency resolution

- But the rectangular window has the largest sidelobe amplitude causing considerable leakage
- From the previous two examples, it can be seen that the large amount of leakage results in minor peaks that may be identified falsely as sinusoids
- Leakage can be reduced by using other types of windows

• Example -

 $x[n] = 0.85 \sin(2\pi \times 0.22) + \sin(2\pi \times 0.26)$

windowed by a length-*R* Hamming window



• Leakage has been reduced considerably, but it is difficult to resolve the two sinusoids



• An increase in the DFT length results in a substantial reduction of leakage, but the two sinusoids still cannot be resolved

- The main lobe width Δ_{ML} of a length-*N* Hamming window is $8\pi/N$
- For *N* = 16, normalized main lobe width is 0.25
- Two frequencies can thus be resolved if their difference is of the order of half of the main lobe width, i.e., 0.125
- In the example considered, the difference is 0.04, which is much smaller

• To increase the resolution, increase the window length to R = 32 which reduces the main lobe width by half



• There now appears to be two peaks

- An increase of the DFT size to *R* = 64 clearly separates the two peaks
- Separation is more visible for R = 256



- <u>General conclusions</u> Performance of DFTbased spectral analysis depends on three factors: (1) Type of window, (2) Window length, and (3) Size of the DFT
- Frequency resolution is increased by using a window with a very small main lobe width
- Leakage is reduced by using a window with a very small relative sidelobe level

- Main lobe width can be reduced by increasing the window length
- An increase in the accuracy of locating peaks is obtained by increasing the DFT length
- It is preferable to use a DFT length that is a power of 2 so that very efficient FFT algorithms can be employed
- An increase in DFT size increases the computational complexity

- DFT can be employed for spectral analysis of a length-*N* sinusoidal signal composed of sinusoidal signals as long as the frequency, amplitude and phase of each sinusoidal component are time-invariant and independent of *N*
- There are situations where the signal being analyzed is instead nonsationary, for which these parameters are time-varying

• An example of a time-varying signal is the chirp signal $x[n] = A\cos(\omega_o n^2)$ and shown below for $\omega_o = 10\pi \times 10^{-5}$



• The instantaneous frequency of x[n] is $2\omega_o n$

- Other examples of such nonstationary signals are speech, radar and sonar signals
- DFT of the complete signal will provide misleading results
- A practical approach would be to segment the signal into a set of subsequences of short length with each subsequence centered at uniform intervals of time and compute DFTs of each subsequence

- The frequency-domain description of the long sequence is then given by a set of short-length DFTs, i.e., a time-dependent DFT
- To represent a nonstationary x[n] in terms of a set of short-length subsequences, x[n] is multiplied by a window w[n] that is stationary with respect to time and move x[n] through the window

• Four segments of the chirp signal as seen through a stationary length-200 rectangular window



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 Short-time Fourier transform (STFT), also known as time-dependent Fourier transform of a signal x[n] is defined by

$$X_{\text{STFT}}(e^{j\omega}, n) = \sum_{m=-\infty}^{\infty} x[n-m]w[m]e^{-j\omega m}$$

where *w*[*m*] is a suitably chosen window sequence

• The STFT is also defined as given below:

$$X_{\text{STFT}}(e^{j\omega}, n) = \sum_{m=-\infty}^{\infty} x[m] w[n-m] e^{-j\omega m}$$

 Here, if w[n] = 1 for all values of n, the STFT reduces to DTFT of x[n]

- Even though DTFT of x[n] exists under certain well-defined conditions, windowed x[n] being of finite length ensures the existence of any x[n]
- Function of w[n] is to extract a finite-length portion of x[n] such that the spectral characteristics of the extracted section are approximately stationary

- X_{STFT} (e^{jω}, n) is a function of 2 variables: integer variable time index n and continuous frequency variable ω
- $X_{\text{STFT}}(e^{j\omega}, n)$ is a periodic function of ω with a period 2π
- Display of $X_{\text{STFT}}(e^{j\omega}, n)$ is usually referred to as **spectrogram**
- Display of spectrogram requires normally three dimensions

- Often, STFT magnitude is plotted in two dimensions with the magnitude represented by the darkness of the plot
- Plot of STFT magnitude of chirp sequence $x[n] = A\cos(\omega_o n^2)$ with $\omega_o = 10\pi \times 10^{-5}$ for a length of 20,000 samples computed using a Hamming window of length 200 shown next



• STFT for a given value of *n* is essentially the DFT of a segment of an almost sinusoidal sequence

- Shape of the DFT of such a sequence is similar to that shown below
- Large nonzero-valued DFT samples around the frequency of the sinusoid
- Smaller nonzero-valued DFT samples at other frequency points



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- In the spectrogram plot, large-valued DFT samples show up as narrow very short dark vertical lines
- Other DFT samples show up as points
- As the instantaneous frequency of the chirp signal increases linearly with *n*, short dark line move up in the vertical direction
- Because of aliasing, dark line starts moving down in the vertical direction
- Spectrogram appears in a triangular shape

- In practice, the STFT is computed at a finite set of discrete values of $\boldsymbol{\omega}$
- The STFT is accurately represented by its frequency samples as long as the number of frequency samples *N* is greater than window length *R*
- The portion of the sequence *x*[*n*] inside the window can be fully recovered from the frequency samples of the STFT

• Sampling $X_{\text{STFT}}(e^{j\omega}, n)$ at *N* equally spaced frequencies $\omega_k = 2\pi k / N$, with $N \ge R$ we get

$$X_{\text{STFT}}[k,n] = X_{\text{STFT}} \left(e^{j\omega}, n \right)_{\omega = 2\pi k / N}$$

R-1

$$= \sum_{m=0}^{N-1} x[n-m]w[m]e^{-j2\pi km/N}, \ 0 \le k \le N-1$$

• If $w[m] \neq 0$, $X_{\text{STFT}}[k,n]$ is simply the *R*point DFT of x[n-m]w[m]

- $X_{\text{STFT}}[k,n]$ is a 2-D sequence and periodic in k with a period N
- Applying the IDFT we get

$$x[n-m]w[m] = \frac{1}{N} \sum_{k=0}^{N-1} X[k,n] e^{j2\pi km/N}, \ 0 \le m \le R-1$$

or

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$$x[n-m] = \frac{1}{Nw[m]} \sum_{k=0}^{N-1} X[k,n] e^{j2\pi km/N}, \ 0 \le m \le R-1$$

• Thus the sequence inside the window can be fully recovered from

• The sampled STFT for a window defined in the region $0 \le m \le R - 1$ is given by

$$X_{\text{STFT}}[k, \ell L] = X_{\text{STFT}}(e^{j2\pi k/N}, \ell L)$$

$$= \sum_{m=0}^{R-1} x [\ell L - m] w [m] e^{-j2\pi km/N}$$

where ℓ and k are integers such that $-\infty < \ell < \infty$ and $0 \le k \le N - 1$

• Figure below shows lines in the (ω,n) -plane corresponding to $X_{\text{STFT}}(e^{j\omega},n)$ for N = 9 and L = 4



• Figure below shows the grid of sampling points in (ω, n) -plane for N = 9 and L = 4



 As we have shown it is possible to uniquely reconstruct the original signal from such a 2-D discrete representation provided
 L < R < N

where N is the DFT length, R is the window length and L is the sampling period in time

- The function of the window w[n] is to extract a portion of the signal x[n] and ensure that the extracted section is approximately stationary
- To the end, the window length *L* should be small, in particular for signals with widely varying spectral parameters

- A decrease in the window length increases the time resolution property of the STFT
- On the other hand, the frequency resolution property of the STFT increases with an increase in the window length
- A shorter window provides a wideband spectrogram, whereas, a longer window results in a narrowband spectrogram

- Parameters characterizing the DTFT of a window are the main lobe width Δ_{ML} and the relative sidelobe amplitude $A_{s\ell}$
- Δ_{ML} determines the ability of the window to resolve two sinusoidal components in the vicinity of each other
- $A_{s\ell}$ controls the degree of leakage of one component into a nearby signal component

In order to obtain a reasonably good estimate of the frequency spectrum of a time-varying signal, the window should be chosen to have very small A_{sℓ} with a length *R* chosen based on the acceptable accuracy of the frequency and time resolutions

STFT Computation Using MATLAB

- The M-file specgram can be used to compute the STFT of a signal
- The application of specgram is illustrated next



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STFT Computation Using MATLAB

 Using Program 11_4 we compute the narrowband spectrogram of this speech signal



STFT Computation Using MATLAB

• The **wideband spectrogram** of the speech signal is shown below



• The frequency and time resolution tradeoff between the two spectrograms can be seen