

LECTURE #1 OBJECTIVES

- Write general formula for a “sinusoidal” waveform, or signal
- From the formula, plot the sinusoid versus time
- What’s a signal?
 - It’s a function of time, $x(t)$
 - in the mathematical sense

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LECTURE #2 OBJECTIVES

- Relate TIME-SHIFT to PHASE
- Introduce an ABSTRACTION:
 - Complex Numbers represent Sinusoids
 - Complex Exponential Signal

$$z(t) = Ze^{j\omega t}$$

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LECTURE #3 OBJECTIVES

- Phasors = Complex Amplitude
 - Add Sinusoids = Complex Addition
 - PHASOR ADDITION THEOREM
- $$z(t) = Ze^{j\omega t} = (Ae^{j\phi})e^{j\omega t}$$
- Develop the ABSTRACTION:
 - Complex Numbers represent Sinusoids

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LECTURE #4 OBJECTIVES

- Sinusoids with DIFFERENT Frequencies
 - Add Sinusoids
- $$x(t) = \sum_{k=1}^N A_k \cos(2\pi f_k t + \phi_k)$$
- SPECTRUM Representation
 - Graphical Form shows Different Freqs

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LECTURE #5 OBJECTIVES

- Signals with HARMONIC Frequencies
 - Add Sinusoids with $f_k = kf_0$
- $$x(t) = A_0 + \sum_{k=1}^N A_k \cos(2\pi kf_0 t + \phi_k)$$
- ANALYSIS via Fourier Series
 - For PERIODIC signals: $x(t+T) = x(t)$

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LECTURE #6 OBJECTIVES

- Frequency can change vs. time
 - Basis of Frequency Modulation (FM)
 - Define “instantaneous frequency”
 - Chirp Signals (LFM)
 - Quadratic phase
- $$x(t) = A \cos(\alpha t^2 + 2\pi f_0 t + \phi)$$

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LECTURE #7 OBJECTIVES

- SAMPLING can cause ALIASING
 - ▮ Sampling Theorem
 - ▮ Sampling Rate > 2(Highest Frequency)
- Spectrum for digital signals, $x[n]$
 - ▮ Normalized Frequency

$$\hat{\omega} = \omega T_s = \frac{2\pi f}{f_s}$$

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LECTURE #8 OBJECTIVES

- DIGITAL-to-ANALOG CONVERSION is
 - ▮ Reconstructing $x(t)$ from its samples
 - ▮ SAMPLING THEOREM applies
 - ▮ Smooth Interpolation
- Mathematical Model of D-to-A
 - ▮ SUM of SHIFTED PULSES
 - ▮ Linear Interpolation example

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LECTURE #9 OBJECTIVES

- INTRODUCE FILTERING IDEA
 - ▮ Weighted Average
 - ▮ Running Average
- FINITE IMPULSE RESPONSE FILTERS
 - ▮ FIR Filters

$$y[n] = \sum_{k=0}^M b_k x[n-k]$$
 - ▮ Show how to compute the output $y[n]$ from the input signal, $x[n]$

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LECTURE #10

- BLOCK DIAGRAM REPRESENTATION
 - ▮ Components for Hardware
 - ▮ Connect Simple Filters Together to Build More Complicated Systems
- GENERAL PROPERTIES of FILTERS
 - ▮ LINEARITY
 - ▮ TIME-INVARIANCE
 - ▮ ==> CONVOLUTION

LTI SYSTEMS

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LECTURE #11

- SINUSOIDAL INPUT SIGNAL
 - ▮ DETERMINE FIR OUTPUT
- FREQUENCY RESPONSE of FIR
 - ▮ MAGNITUDE vs. Frequency
 - ▮ PHASE vs. Freq
 - ▮ PLOTTING:

$$H(\hat{\omega}) = |H(\hat{\omega})| e^{j\varphi(\hat{\omega})}$$

MAG

PHASE

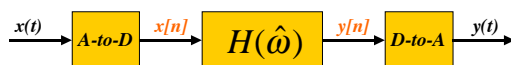
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LECTURE #12

- DIGITAL PROCESSING of ANALOG SIGNALS
- UNIFICATION:
 - ▮ How does Frequency Response affect $x(t)$ to produce $y(t)$?



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LECTURE #13

- INTRODUCE the Z-TRANSFORM
 - Give Mathematical Definition
 - Show how $H(z)$ POLYNOMIAL simplifies analysis
 - CASCADE EXAMPLE
- Z-Transform can be applied to
 - FIR Filter: $h[n] \rightarrow H(z)$
 - Signals: $x[n] \rightarrow X(z)$

$$H(z) = \sum_n h[n]z^{-n}$$

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LECTURE #14

- Relate $H(z)$ to FREQUENCY RESPONSE

$$H(\hat{\omega}) = H(z) \Big|_{z=e^{j\hat{\omega}}}$$

- Zeros of $H(z)$
- THREE DOMAINS:
 - Show Relationship for FIR:

$$h[n] \leftrightarrow H(z) \leftrightarrow H(e^{j\hat{\omega}})$$

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LECTURE #15

- INFINITE IMPULSE RESPONSE FILTERS
 - IIR Filters
 - Have FEEDBACK: PREVIOUS OUTPUTS

$$y[n] = \sum_{\ell=1}^N a_{\ell} y[n-\ell] + \sum_{k=0}^M b_k x[n-k]$$

- Show how to compute the output $y[n]$
 - FIRST-ORDER CASE ($N=1$)
 - $h[n] \leftrightarrow H(z)$

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LECTURE #16

- FREQUENCY RESPONSE of IIR

- Get $H(z)$ first

$$H(e^{j\hat{\omega}}) = H(z) \Big|_{z=e^{j\hat{\omega}}}$$

- $H(z)$ has POLES and ZEROS
- THREE-DOMAIN APPROACH
 - Get $h[n]$ from $H(z)$
 - Get STRUCTURES from $H(z)$

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LECTURE #17

- SECOND-ORDER IIR FILTERS
 - TWO FEEDBACK TERMS

$$y[n] = a_1 y[n-1] + a_2 y[n-2] + \sum_{k=0}^2 b_k x[n-k]$$

- $H(z)$ can have COMPLEX POLES & ZEROS
- THREE-DOMAIN APPROACH
 - UNIFIES $h[n]$ & FREQUENCY RESPONSE in terms of POLES and ZEROS

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LECTURE #18

- THREE-DOMAIN APPROACH
 - EXHIBIT BANDPASS FILTERS
- RE-UNIFICATION:
 - How does Frequency Response affect $x(t)$ to produce $y(t)$?



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