

EECS 360  
Semester Review

- 1) Classification of signals:
  - a) Periodic
  - b) Aperiodic
  - c) Energy
  - d) Power
  - e) Continuous time
  - f) Discrete time
  - g) Deterministic
  - h) Random
- 2) Phasor representation of sin & cos and complex numbers, magnitude and phase
- 3) Classification of Systems
  - a) Linear/Nonlinear
  - b) Time varying/Time invariant
  - c) Causal/non-causal
  - d) Continuous time/Discrete time
- 4) Special functions
  - a)  $\delta(t)$
  - b)  $\Lambda(t)$
  - c)  $u(t)$
  - d)  $\pi(t)$
- 5) Convolution
  - a) Continuous time
  - b) Discrete time
- 6) Impulse response of linear time invariant systems,  $h(t)$
- 7) Step response of systems  $a(t)$
- 8) Impulse response of cascaded linear time invariant systems,  $h(t)$  and  $h(n)$ .
- 9) Bounded input/Bounded output stability and constraints on  $h(t)$  and  $h(n)$  for stability
- 10) Causality and constraints on  $h(t)$  for stability
- 11) Fourier Series
  - a) Complex exponential
  - b) Sin/Cos
  - c) Cos
- 12) Power-Parsaval's theorem
- 13) Spectral plots for periodic signals: one sided and double sided
- 14) Fourier Transform
- 15) Fourier Transform theorems and properties
- 16) Energy-Parsaval's theorem for aperiodic signals
- 17) Transfer Function of linear time invariant systems - $H(f)$
- 18) Amplitude and phase response of linear time invariant systems,  $H(\omega)$  and  $H(z)$ .
- 19) Transfer Function of cascaded linear time invariant systems

- 20) Criteria for an ideal linear time invariant systems
  - a) ILPF
  - b) IBPF
- 21) Distortion for linear time invariant systems
  - a) Amplitude distortion
  - b) Phase distortion
- 22) Distortion for non-linear time or time varying systems-Total Harmonic distortion
- 23) Signals and systems
  - a) Impact of systems on signals
  - b) Bandwidth and its definitions
  - c) Inverse time duration-bandwidth relationship
  - d) Inverse rise time and bandwidth relationship
  - e) If  $B_h \gg B_s$  then minimal distortion,  
where  $B_h$ =system bandwidth and  $B_s$ =signal bandwidth
- 24) Sampling theorem
  - a) Minimum sample rate
  - b) Spectrum of a sampled signal
  - c) Aliasing
- 25) Discrete Fourier Transform
  - a) Definition
  - b) DFT is the Fourier Series of a periodic sampled (discrete) signal
  - c) Parameters of the DFT, frequency resolution, record length, and sample rate
  - d) Picket fence effect
  - e) Spectral leakage
  - f) Circular convolution
- 26) Z-Transform and digital filters
  - a) Definition
  - b) Frequency response is  $H(z)$  evaluated on the unit circle
  - c) Properties of Z-transforms
    - Linearity
    - Delay
    - Multiplication by  $a^n$
    - Multiplication by  $k$
    - Convolution
  - d) Poles and zeros for the general difference equation
  - e) Relationship of pole and zero locations to the frequency response
  - f) Analysis-given the general difference equation find:
    - Location of poles and zeros
    - Frequency response
  - g) Design of digital filters
    - Position the poles and zeros in the z-plane to get desired freq. response
    - Form  $H(z)$
    - Determine the coefficients for the difference equation
  - h) All poles of  $H(z)$  must be inside the unit circle for the system to be stable