

The University of Texas at Austin
Dept. of Electrical and Computer Engineering
Final Exam

Date: December 10, 2010

Course: EE 313 Evans

Name: _____
Last, First

- The exam is scheduled to last 3 hours.
- Open books and open notes. You may refer to your homework assignments and the homework solution sets.
- Calculators are allowed.
- **Power off all cell phones**
- You may use any standalone computer system, i.e. one that is not connected to a network.
- All work should be performed on the quiz itself. If more space is needed, then use the backs of the pages.
- **Fully justify your answers unless instructed otherwise. If you cite a reference, then please also provide the page number and quote you are using.**

Problem	Point Value	Your score	Topic
1	10		Mathematical Modeling
2	10		Differential Equation Rhythm
3	10		Differential Equation Blues
4	10		Discrete-Time Stability
5	10		System Identification
6	15		Discrete-Time Filter Analysis
7	15		Discrete-Time Filter Design
8	10		Sinusoidal Signal
9	10		Sinusoidal Amplitude Demodulation
Total	100		

Final Exam Problem 1. *Mathematical Modeling.* 10 points

Consider a signal $y(t)$ that is the continuous-time output of a light switch.

The signal $y(t)$ is one when the light switch is “on”, and zero when it is “off”.

- (a) Sketch $y(t)$ when the light switch is “off” before time $t = 0$ and turns “on” at $t = 0$ and stays “on”. Give a mathematical definition of $y(t)$ in terms of the unit step function $u(t)$. Please give the value of $u(0)$ that you are using. 5 points.

- (b) Sketch $y(t)$ when the light switch turns “on” at $t = 0$ s and turns “off” at $t = 1$ s. Give a mathematical definition of $y(t)$ in terms of the unit step function $u(t)$. Please give the value of $u(0)$ that you are using. 5 points.

Final Exam Problem 2. *Differential Equation Rhythm.* 10 points.

Consider a continuous-time system with input $x(t)$ and output $y(t)$ governed by the differential equation

$$\frac{d^2}{dt^2} y(t) + 3 \frac{d}{dt} y(t) + 2y(t) = \frac{d^2}{dt^2} x(t)$$

for $t \geq 0^+$.

- (a) What are the characteristic roots of the differential equation? 2 points.

- (b) Find the zero-input response assuming non-zero initial conditions. Please leave your answer in terms of C_1 and C_2 . 4 points.

- (c) Find the zero-input response for the initial conditions $y(0^+) = 0$ and $y'(0^+) = 1$. 4 points.

Final Exam Problem 3. *Differential Equation Blues.* 10 points.

Consider a continuous-time linear time-invariant (LTI) system with input $x(t)$ and output $y(t)$ governed by the differential equation

$$\frac{d^2}{dt^2} y(t) + 3 \frac{d}{dt} y(t) + 2y(t) = \frac{d^2}{dt^2} x(t)$$

for $t \geq 0^-$.

- (a) Give a formula for the transfer function in the Laplace domain. 2 points.

- (b) Give the values of the poles and zeroes of the transfer function. 2 points.

- (c) Give the region of convergence for the transfer function. 2 points.

- (d) Give a formula for the frequency response of the LTI system. 2 points.

- (e) What is the frequency selectivity of the LTI system? Lowpass, bandpass, bandstop, highpass, notch or all-pass? 2 points.

Final Exam Problem 4. *Discrete-Time Stability.* 10 points.

In this problem, the input signal is denoted by $x[n]$ and the output signal is denoted by $y[n]$.

The input-output relationship of a linear time-invariant (LTI) system is defined as

$$y[n] - 1.6 y[n - 1] + K y[n - 2] = x[n]$$

where K is an adjustable gain that can take any real value. By adjusting K , one can change the time response and frequency response of the system.

- (a) What are the pole locations? Express your answer in terms of K . 3 points.

- (b) For what values of K is the system bounded-input bounded-output stable? 2 points.

- (c) Plot the pole locations in the z -domain as K varies. 2 points.

- (d) Describe the frequency selectivity of the system (lowpass, highpass, bandpass, bandstop, notch, or allpass) for all possible values of K for which the system is bounded-input bounded-output stable. 3 points.

Final Exam Problem 5. *System Identification.* 10 points.

Consider a continuous-time system with input $x(t)$ and output $y(t)$.

You observe the following input-output relationships:

- When $x(t) = u(t)$, the output is $y(t) = u(t)$, assuming that $u(0) = 1$.
- When $x(t) = \cos(2 \pi t)$, the output is $y(t) = \frac{1}{2} + \frac{1}{2} \cos(4 \pi t)$.
- When $x(t) = \cos(4 \pi t)$, the output is $y(t) = \frac{1}{2} + \frac{1}{2} \cos(8 \pi t)$.

(a) Is the system linear and time-invariant? Please justify your answer. 5 points.

(b) Give a formula for the input-output relationship. 5 points.

Final Exam Problem 6. Discrete-Time Filter Analysis. 15 points.

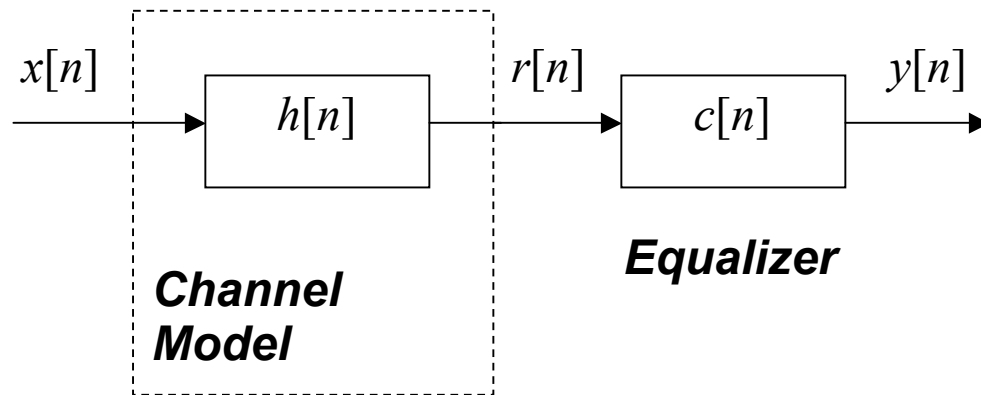
A causal discrete-time linear time-invariant filter with input $x[n]$ and output $y[n]$ is governed by the following difference equation:

$$y[n] = -0.8 y[n-1] + x[n] - x[n-1]$$

- Draw the block diagram for this filter. 3 points.
- What are the initial conditions? What values should they be assigned? 3 points.
- Find the equation for the transfer function in the z -domain including the region of convergence. 3 points.
- Find the equation for the frequency response of the filter. 3 points.
- Describe the frequency selectivity of this filter as lowpass, bandpass, bandstop, highpass, notch, or allpass. Why? 3 points.

Final Exam Problem 7. Discrete-Time Filter Design. 15 points.

Consider the design of a discrete-time LTI filter with impulse response $c[n]$ to equalize a channel modeled as an LTI filter with impulse response $h[n]$:



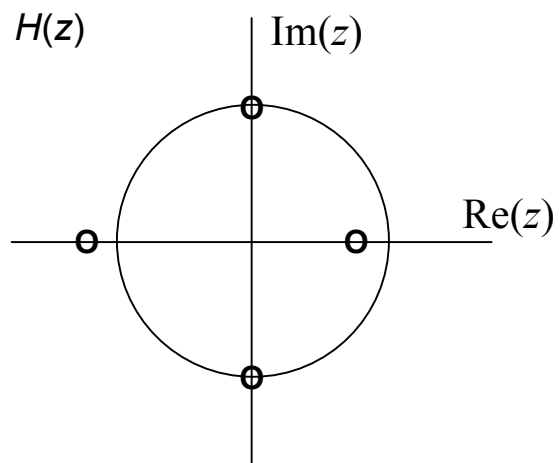
The channel model could represent a communication channel, a biomedical instrument or an audio system that distorts the input signal $x[n]$.

The equalizer is designed to compensate for the magnitude distortion in the channel as best it can. That is, the overall system from $x[n]$ to $y[n]$ should ideally have an all-pass response.

(a) Let $h[n] = 0.9^n u[n]$. What is the transfer function of the equalizer, $C(z)$? 4 points.

(b) Let $h[n] = \delta[n] - 2\delta[n-1]$. What is the transfer function of the equalizer, $C(z)$? 4 points.

(c) Let $H(z)$ have four zeros and no poles. The zeros are shown on the pole-zero diagram on the right. Place poles on the pole-zero diagram to design $C(z)$. You do not have to write the transfer function for $C(z)$. 7 points.



Final Exam Problem 8. Sinusoidal Signal. 10 points.

In practice, we cannot generate a two-sided sinusoid $\cos(2 \pi f_c t)$, nor can we wait until the end of time to observe a one-sided sinusoid $\cos(2 \pi f_c t) u(t)$.

In the lab, we can turn on a signal generator for a short time and observe the output in the time domain on an oscilloscope or in the frequency domain using a spectrum analyzer.

Consider a finite-duration cosine that is on for 1s given by the equation

$$c(t) = \cos(2 \pi f_c t) \text{ rect}(t - 1/2)$$

where f_c is the carrier frequency (in Hz).

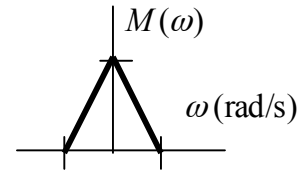
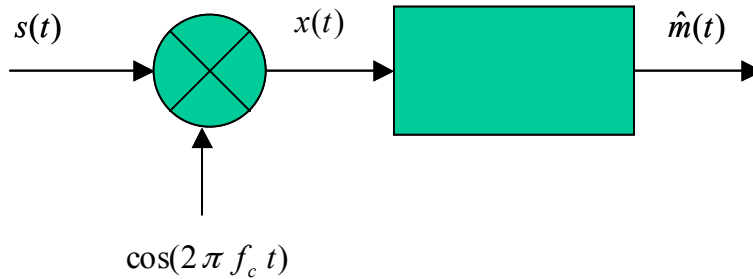
- (a) Give a formula for the Fourier transform of $c(t)$. 3 points.
- (b) Draw the magnitude of the Fourier transform of $c(t)$. 3 points.
- (c) Describe the differences between the magnitude of the Fourier transforms of $c(t)$ and a two-sided cosine of the same frequency. What is the bandwidth of each signal? 4 points.

Final Exam Problem 9. *Sinusoidal Amplitude Demodulation.* ω_m 10 points.

A lowpass, real-valued message signal $m(t)$ with bandwidth f_m (in Hz) is to be transmitted using sinusoidal amplitude modulation

$$s(t) = m(t) \cos(2 \pi f_c t)$$

where f_c is the carrier frequency (in Hz) and $f_c \gg f_m$. The receiver processes the transmitted signal $s(t)$ to obtain an estimate of the message signal, $\hat{m}(t)$, as follows:



Hence, $x(t) = s(t) \cos(2 \pi f_c t)$. $M(\omega)$ is plotted above to the upper right.

(a) Plot the Fourier transform of $s(t)$, i.e. $S(\omega)$. 4 points.

(b) Plot the Fourier transform of $x(t)$, i.e. $X(\omega)$. 4 points.

(c) Give the smallest passband frequency and the largest stopband frequency for the lowpass filter to recover $m(t)$. 2 points.