UNIVERSITY OF TEXAS AT AUSTIN Dept. of Electrical and Computer Engineering

Quiz#2

Date: November 3, 1999

Course: EE 313

Name: _____

Last,

First

- The exam is scheduled to last 50 minutes.
- Open books and open notes. You may refer to your homework and solution sets.
- Calculators are allowed.
- 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.

Problem	Point Value	Your Score	Topic
1	30		Differential Equation
2	20		Step-To-Pulse Response
3	30		Tapped Delay Line
4	20		Sigma-Delta Modulation
Total	100		

Problem 2.1 Differential Equation. 30 points.

Solve the following differential equation

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

with the initial conditions y(0) = 1 and y'(0) = 1 by using the Laplace transform.

Problem 2.2 Step-To-Pulse Response. 20 points.

The step response of a linear time-invariant system is $4(1 - e^{-5t})u(t)$. Suppose that an input

$$x(t) = \begin{cases} 5 & \text{if } 0 \le t \le 2\\ 0 & \text{otherwise} \end{cases}$$

is applied to the system. Find the corresponding output y(t) for $-\infty < t < \infty$.

Problem 2.3 Tapped Delay Line. 30 points.

The **discrete-time** tapped delay line is also known as a digital finite impulse response filter. For input x[k], the output y[k] is given by

$$y[k] = \sum_{m=0}^{N-1} a_m x[k-m]$$

(a) Compute the transfer function H(z). 7 points.

(b) Compute the frequency response $H(\omega)$. 7 points.

- (c) For each of the following sets of coefficients, state whether the filter is lowpass, highpass, bandpass, or bandstop.
 - i. $N = 2, a_0 = 1$, and $a_1 = 1$. 4 points.
 - ii. $N = 2, a_0 = 1$, and $a_1 = j$. 4 points.
 - iii. $N = 2, a_0 = 1$, and $a_1 = -1$. 4 points.
 - iv. N = 2, $a_0 = 1$, and $a_1 = -j$. 4 points.

Problem 2.4 Sigma-Delta Modulation. 20 points.

Shown below is a type of sigma-delta modulator called a noise-shaping feedback coder.



We can approximate the effect of the quantizer as a gain K, which would make the overall system linear and time-invariant. Replace the quantizer with a gain of K and derive the transfer function from input x[n] to output y[n].