## EEE 391

Basics of Signals and Systems
Midterm Exam
25 March 2010, Thursday
closed book and notes
no calculators

Given Time: 120 min

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Last Name	:
First Name	:
Section	:
ID number	:
Signature	:

Exam	Total	Points
Part	Points	Received
Q1	25	
Q2	25	
Q3	20	
Q4	30	
Total	100	

## Allocation of points:

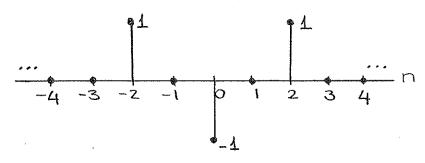
- 1) 25 pts (a) 6 pts (b) 6 pts (c) 6 pts (d) 7 pts
- 2) 25 pts (a) 4 pts (b) 8 pts (c) 9 pts (d) 4 pts
- 3) 20 pts (a) 5 pts (b) 5 pts (c) 5 pts (d) 5 pts
- 4) 30 pts (a) 3 pts (b) 4 pts (c) 5 pts (d) 10 pts (e) 8 pts

## Attention:

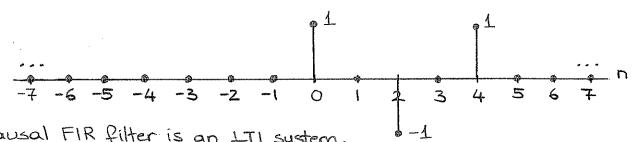
Read all the questions carefully and show your work or justify your answer for full or partial credit.

Please put your answer for each part in the given spaces.

(1) A causal FIR filter produces the following output sequence when S[n+z] is given to it as input:



a) Find the impulse response of the filter and plot it below: (h[n]):



causal FIR filter is an LTI system. -1Therefore,  $S[n+2] \mapsto h[n+2]$  (This was shown in class)

S[n] > h[n] (shift the given output by 2 units to the right)

b) Identify and list all of the filter coefficients & bk3.

$$\{b_k\} = \{1,0,-1,0,1\}$$

.What is the order of the filter?

order = 4

c) Write the difference equation of the filter.

$$y[n] = \times [n] - \times [n-2] + \times [n-4]$$

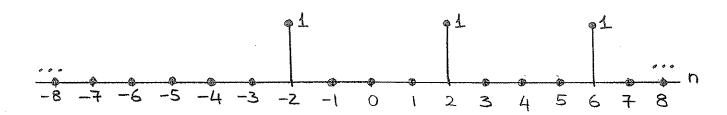
$$y[n] = b_{0} \times [n] + b_{1} \times [n-1] + b_{2} \times [n-2] + b_{3} \times [n-3] + b_{4} \times [n-4]$$

$$= \times [n] - \times [n-2] + \times [n-4] \qquad b_{0} = b_{4} = 1 \quad b_{2} = -1$$

$$b_{1} = b_{3} = 0$$

(1) d) Find the output of the filter when the input is:  $\times [n] = \delta[n+2] + \delta[n] + \delta[n-2]$  and plot it below:

$$y[n] = S[n+2] + S[n-2] + S[n-6]$$



Since the system is LTI, you can use superposition to solve this part:

$$h[n+2] = S[n+2] - S[n] + S[n-2]$$

$$h[n] = S[n] - S[n-2] + S[n-4]$$

$$h[n-2] = S[n-2] - S[n-4] + S[n-6]$$

$$y[n] = h[n+2] + h[n] + h[n-2]$$

$$= S[n+2] + S[n-2] + S[n-6]$$

$$x(t)$$
 square-law  $y(t) = [x(t)]^2$  input  $(\cdot)^2$  output

The signal x(t) = 2 cos(wot) + sin (2wot) is given as input to the square-law system shown above.

a) What is the average value of x(t)?

average (dc) value = 
$$\frac{1}{10} \int_{0}^{10} \left[ 2\cos(\omega_{s}t) + \sin(2\omega_{s}t) \right] dt = \frac{2\sin(\omega_{s}t) - \cos(2\omega_{s}t)}{\omega_{s}T_{0}} = 0$$

b) Find the output of the system for the given input and express it in the form:

 $y(t) = A_0 + \sum_{k=1}^{N} A_k \cos(\omega_0 kt + \emptyset_k)$ 

$$y(t) = \left[ \times (t) \right]^{2}$$

$$= \left[ 2\cos(\omega_{0}t) + \sin(2\omega_{0}t) \right]^{2}$$

$$= 4\cos^{2}(\omega_{0}t) + 4\cos(\omega_{0}t)\sin(2\omega_{0}t) + \sin^{2}(2\omega_{0}t)$$

$$= \frac{2}{4} \left[ \frac{1 + \cos(2\omega_{0}t)}{2} \right] + \frac{2}{4} \left[ \frac{\sin(3\omega_{0}t) + \sin(\omega_{0}t)}{2} \right] + \left[ \frac{1 - \cos(4\omega_{0}t)}{2} \right]$$

$$= \frac{5}{2} + 2\cos(2\omega_{0}t) + 2\sin(3\omega_{0}t) + 2\sin(3\omega_{0}t) - \frac{1}{2}\cos(4\omega_{0}t)$$

$$= \frac{5}{2} + 2\sin(\omega_{0}t) + 2\cos(2\omega_{0}t) + 2\sin(3\omega_{0}t) - \frac{1}{2}\cos(4\omega_{0}t)$$

$$= \frac{5}{2} + 2\cos(\omega_{0}t - \frac{\pi}{2}) + 2\cos(2\omega_{0}t) + 2\cos(3\omega_{0}t - \frac{\pi}{2}) + \frac{1}{2}\cos(4\omega_{0}t + \pi)$$

$$(k=0)$$

$$(k=1)$$

$$(k=2)$$

$$(k=3)$$

$$(k=4)$$

$$A_1 = 2$$

$$A_z = Z$$

$$A_3 = 2$$

$$\phi_0 = 0$$

$$Q' = -\overline{x}$$

$$\phi_z = 0$$

$$\emptyset_4 = \pi$$

(2) c) Is y(t) a periodic signal? If so find all of its Fourier series coefficients. If not, explain why.

y(t) is a periodic signal with fundamental period To = 211 was

Fourier series coefficients:

$$y(t) = \frac{5}{2} + 2\cos(\omega_{0}t - \frac{\pi}{2}) + 2\cos(2\omega_{0}t) + 2\cos(3\omega_{0}t - \frac{\pi}{2}) + \frac{1}{2}\cos(4\omega_{0}t + \pi)$$

$$= \frac{5}{2} + 2\left[\frac{e^{j(\omega_{0}t - \frac{\pi}{2})} + e^{-j(\omega_{0}t - \frac{\pi}{2})}}{2}\right] + 2\left[\frac{e^{j(\omega_{0}t - \frac{\pi}{2})} + 2\left[\frac{e^{j(\omega_{0}t - \frac{\pi}{2})}}{2}\right] + 2\left[\frac{e^{j(\omega_{0}t + \frac{\pi}{2})}}{2}\right] + 2\left[\frac{e^{j(\omega_{0}t + \frac{\pi}{2})} + e^{-j(\omega_{0}t + \frac{\pi}{2})}}{2}\right]$$

$$= \frac{1}{4}e^{-j\pi}e^{-j4\omega_{0}t} + e^{j\frac{\pi}{2}}e^{-j3\omega_{0}t} + e^{j\frac{\pi}{2}}e^{-j\omega_{0}t} + e^{j\frac{\pi}{2}}e^{-j\omega_{0}t}$$

$$= \frac{1}{4}e^{-j\pi}e^{-j4\omega_{0}t} + e^{j\frac{\pi}{2}}e^{-j3\omega_{0}t} + e^{j\frac{\pi}{2}}e^{-j\omega_{0}t} + e^{j\frac{\pi}{2}}e^{-j\omega_{0}t}$$

$$a_{-4} = \frac{1}{4}e^{-jT} = -\frac{1}{4}$$
 $a_{-3} = e^{-j\frac{T}{2}} = j$ 
 $a_{-3} = e^{-j\frac{T}{2}} = -j$ 
 $a_{-2} = 1$ 
 $a_{-2} = 1$ 
 $a_{-3} = e^{-j\frac{T}{2}} = -j$ 

$$\alpha_{1} = e^{i\frac{\pi}{2}} = \overline{j} \qquad \alpha_{1} = e^{-i\frac{\pi}{2}} = -\overline{j}$$

$$a_0 = \frac{5}{2}$$

d) What is the average value of y(t)?

$$y_{av}(t) = a_0 = \frac{5}{2}$$

Note that the average value of the input signal is zero whereas the average value of the output signal is not. This occurs because of the squaring operation involved.

(3) A system is defined by the following equation where  $\times [n]$  is the input to the system and y[n] is its output:

$$y[n] = \begin{cases} x[n] + (n+1)^2 & \text{for } 0 \le n \le 10 \\ 0 & \text{otherwise} \end{cases}$$

- a) Is the system linear? Justify.
- b) Is the system time invariant? Justify.
- c) Is the system causal? Justify.
- d) Find the (unit) impulse response of the system.
- a) The system is nonlinear.

$$\times [n] \mapsto y[n]$$

$$\propto \times [n] \mapsto \begin{cases} \alpha \times [n] + (n+1)^{2} & \text{for } 0 \leq n \leq 10 \\ 0 & \text{otherwise} \end{cases}$$

b) The system is not time invariant (it is time varying)

$$\times [n] \mapsto y[n]$$

$$\times [n-n_0] \mapsto \begin{cases} \times [n-n_0] + (n+1)^2 & \text{for } 0 \le n \le 10 \\ 0 & \text{otherwise} \end{cases}$$

$$= w[n]$$

$$y[n-n_0] = \begin{cases} x[n-n_0] + (n-n_0+1)^2 & \text{for } 0 \le n-n_0 \le 10 \\ n_0 \le n \le n_0+10 \end{cases}$$

c) The system is <u>causal</u>. It does not use any future values of the input.

d) 
$$h[n] = \begin{cases} 5[n] + (n+1)^2 & \text{for } 0 \le n \le 10 \\ 0 & \text{otherwise} \end{cases}$$

4) The signal  $x(t) = -3 + 2\cos(80\pi t + II)$  is sampled uniformly at a rate of  $f_s = 60 \, \text{Hz}$  to obtain a discrete-time signal  $\times [n]$ .

a) What is the Nyquist rate of the given signal?

$$\omega_0 = 80 \, \text{T rad/s}$$
.  
 $f_0 = 40 \, \text{Hz} = f_{\text{max}}$   
 $f_{\text{Nyquist}} = 2 f_{\text{max}} = 80 \, \text{Hz}$ .

b) At the given sampling rate, what is the maximum frequency that can be reconstructed perfectly by an ideal reconstructor?

$$W = \hat{W}f_{s} = \frac{\hat{W}}{T_{s}}$$

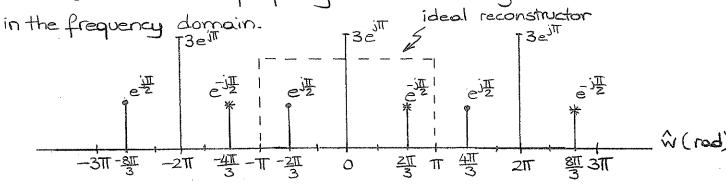
$$= \pm \pi f_{s}$$

$$f = \frac{\pm \pi f_{s}}{2\pi} = \pm \frac{f_{s}}{2}$$

c) Find the sampled signal x[n].

$$\times [n] = \times (nT_s) = -3 + 2\cos\left(\frac{4\pi}{3}n + \frac{\pi}{2}\right)$$
$$= \times (n/p_s)$$

d) Plot the digital spectrum of the signal  $\times$  [n] over the interval  $-3\pi$   $\leqslant 3\pi$ . Include all aliases in this interval and also their complex amplitudes. Draw a rectangular window using dashed lines to indicate the frequency window used by the ideal reconstructor in the frequency domain.



e) X[n] is converted to an analog signal by an ideal D-to-C converter with fs = 60 Hz. Find the signal at the output of the reconstructor. Indicate whether this is a case of undersampling or oversampling. If undersampling, indicate which components are aliased and whether folding occurs or not.

e) 
$$y[n] = 3e^{j\pi} + e^{-j\frac{\pi}{2}} e^{j\frac{\pi}{3}n} + e^{j\frac{\pi}{2}} e^{-j\frac{\pi}{3}n}$$
  
 $= -3 + 2\cos(\frac{\pi}{3}n - \frac{\pi}{2})$   
 $y(t) = -3 + 2\cos(\frac{\pi}{3}f_5t - \frac{\pi}{2})$   $\hat{\omega} = \omega T_5 = 0$ 

$$y(t) = -3 + 2\cos\left(\frac{2\pi}{3}f_{5}t - \frac{\pi}{2}\right)$$

$$= -3 + 2\cos\left(\frac{3\pi}{3}60t - \frac{\pi}{2}\right)$$

$$= -3 + 2\cos\left(40\pi t - \frac{\pi}{2}\right)$$
 (reconstructed signal)

The d.c. component of x(t) is not aliased and can be reconstructed perfectly. The 40 Hz sinusoidal component is aliased because it requires a minimum sampling rate of 80 Hz. Therefore, this is a case of undersampling of the second component. Since  $\hat{w} = \pm 4\pi$  rad,  $\pi < 1 \hat{w} < 2\pi$ , folding occurs, resulting in the phase reversal of the reconstructed signal. The reconstructed signal has two components: a d.c. component and a 20 Hz component.