

Name :	
Roll No.:	
Invigilator's Signature :	

CS/M.Tech (ECE)/SEM-1/MCE-103/2010-11 2010-11 ADVANCED DIGITAL SIGNAL PROCESSING

Time Allotted: 3 Hours Full Marks: 70

The figures in the margin indicate full marks.

Candidates are required to give their answers in their own words as far as practicable.

Answer Question No. 1 and any four from the rest.

1. Answer *all* questions :

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- a) Do you compute energy and power of a discrete signal simultaneously? Justify your answer with an example.
- b) Define the necessary and sufficient condition that must be satisfied for the digital filter to be causal and realizable.
- c) What are the significance of the position of poles and zeros to design a filter in *Z*-domain?

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d) Among the following systems point out which one has minimum, maximum and mixed phase :

i)
$$H_1(z) = 6 + z^{-1} - z^{-2}$$

ii)
$$H_2(z) = 1 + 5z^{-1}/3 - 2z^{-2}/3$$

iii)
$$H_3(z) = 1 - 5z^{-1}/2 - 3z^{-2}/2$$

iv)
$$H_4(z) = 1 - z^{-1} - 6z^{-2}$$
.

- e) A signal has 1024 samples. How many levels of Haar transform are possible? What is the significance of fluctuating part in Wavelet transform?
- f) What would be the problem if your system is not a LTI system?
- g) What is Kalman filter and what can it do?
- 2. a) The following input-output pairs have been observed during a time-invariant system:

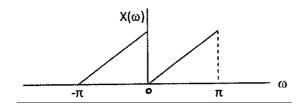
$$x_1[n] = \{1, 0, 2\} \longleftrightarrow y_1[n] = \{0, 1, 0, 2\}$$

$$x_2[n] = \{0, 0, 3\} \xrightarrow{T} y_2[n] = \{0, 1, 0, 2\}$$

$$x_3[n] = \{0, 0, 0, 1\} \xrightarrow{T} y_3[n] = \{1, 2, 1\}$$

Can you draw any conclusion regarding the linearity of the system? What is the impulse response?

- b) Show that a relaxed LTI system is causal if and only if h[n] = 0 for n < 0.
- c) Determine the signal x[n] if its Fourier transform is as given below:



- d) Show that the Fourier transform of the correlation between the signal $x_1[n]$ and $x_2[n]$ is $X_1(\omega)X_2(-\omega)$. 4+3+4+3
- 3. a) What is the difference between Steady state and

 Transient response to sinusoidal input signals?
 - b) Give an idea how a LTI system can be used as a frequency selective filter.

- c) Consider the filter : y[n] = 0.9y[n-1] + bx[n].
 - i) Determine b so that $|H(\omega)| = 0$.
 - ii) Determine the frequency at which $|H(\omega)| = 1/\sqrt{2}$.
 - iii) Is this a LPF, HPF or BPF? Justify your answer.
- d) Knowing $\left|H\left(\omega\right)\right|^{2}$, do you determine H(z) ? Explain.

$$2 + 4 + (1\frac{1}{2} + 1\frac{1}{2} + 2) + 3$$

- 4. a) Compute the DFT of the 4-point sequence $x[n] = \{0, 1, 2, 3\} \text{ using matrix method.}$
 - b) Determine the output response y[n], if $h[n] = \{1, 1, 1\}$; $x[n] = \{1, 2, 3, 1\}$ by using Circular convolution. Under what condition is the result of Circular convolution same as Linear convolution?
 - c) Compute a 4-point DFT of a sequence $x[n] = \{0, 1, 2, 3\}$ using DIT-FET. 4 + (3 + 2) + 5

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- 5. a) What kind of extra features you can get from Wavelet transform over Fourier?
 - b) How conservation and compaction of energy are supported in Haar Wavelet transform with respect to the signal f = (4, 6, 10, 12, 8, 6, 5, 5)?
 - c) What is the significance of MRA? Derive an expression of 2 level Haar transform in terms of second average signal (A^2) and first and second detail signal D^1 and D^2 respectively. 2 + 5 + 7
- 6. a) For the desired frequency response $H_d(\omega)$, explain the method of designing the linear phase filter of odd length by frequency sampling method with symmetric unit impulse response.

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b) Determine the coefficients of linear phase FIR filter of length 15 and symmetric unit impulse response and a frequency response that satisfies

$$H_r(2\pi k/15) = 1$$
 for $k = 0, 1, 2, 3$
= 0.4 for $k = 4$
= 0 for $k = 5, 6, 7. $9 + 5$$

- 7. a) Define covariance of two random variables x_1 and x_2 .
 - b) Consider an object falling under a constant gravitational field. Let y (t) denotes the height of object. Show that only the estimated state from the previous time step and the current measurement are needed to compute the estimate of current state.
 - c) If the state of the recursive estimation is represented by following two parameters.
 - i) $\hat{X}_{k/k}$, the posterior establishment at time k given observation up to and including at time k, and
 - ii) $P_{k/k}$, the posterior error covariance matrix (a measure of accuracy of the state estimate),

Explain the two phase 'Predict' and 'Update' of the estimator. 2 + 5 + 7

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- 8. a) Explain the energy density spectrum of a continuous signal $x_a(t)$ sampled at a uniform sampling rate F_S to result the sequence x[n].
 - b) Compute energy density spectrum of $x_a(t)$ from its samples x[n]. Consider x[n] to be of finite duration. 5+9

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