D 51340

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SEVENTH SEMESTER B.TECH. (ENGINEERING) EXAMINATION, DECEMBER 2008

EE 04 702-DIGITAL SIGNAL PROCESSING

(2004 Admissions)

Time : Three Hours

Maximum : 100 Marks

I. (a) Consider the system defined by the relation :

y(n) = ay(n-1) + x(n)

Given the initial condition y(0) = 1, find whether the system is (i) linear; (ii) time variant.

(b) Convolve the following signals :

$$x(n) = \{1, 1, 1, 1, 1\}$$

$$\uparrow$$

$$h(n) = \delta(n) - \delta(n-1) + \delta(n-2) + \delta(n-3).$$

(c) Prove that convolution in frequency domain is equal to multiplication in time domain.

(d) Find the circular convolution of :

 $x_1(n) = \{..., 1, 2, 1, ...\}$ and $x_2(n) = \{..., 1, 3, 1, ...\}, N = 3.$

(e) Draw the direct form implementation of the FIR filter:

 $H(z) = 1 + 2z^{-1} + 3z^{-2} - yz^{-4} + 5z^{-5}.$

- (f) Give the features of floating point format.
- (g) Distinguish between Passive and Active filters.
- (h) What is frequency warping?

 $(8 \times 5 = 40 \text{ marks})$

II. (a) (i) Test whether the following signal is periodic or not :

$$x(n) = 5\cos(n\pi/4) + 2\sin(3n\pi/4).$$

(ii) Find the circular slip $x[((n+1))_3]$ of the sequence $x(n) = \{\dots, 1, 1, 1, 1, \dots\}$ N = 4.

(15 marks)

Or

Turn over

(b) Determine the zero input response of the system described by the homogeneous second-order differential equation y(n) - 3y(n-1) - 4y(n-2) = 0.

(15 marks)

III. (a) With the help of an example illustrate the overlap and same method of block convolution. (15 marks)

Or

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- (b) Explain the following :---
 - (i) Decimation in time FFT algorithms.
 - (ii) Decimation in frequency FFT algorithms.

(15 marks)

IV. (a) Obtain the filter coefficients of a 3-stage lattice structure FIR filter, from its equivalent thirdorder direct form structure whose coefficients are $a_0 = 1$, $a_1 = 11/8$, $a_2 = 7/8$, $a_3 = 1/4$.

(15 marks)

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(b) Discuss the applications of DSP in : (i) damage processing ; (ii) speech processing.

(15 marks)

V. (a) Compare the advantages and disadvantages of various window functions. (15 marks)

Or

(b) Design a low pass filter for the following specifications (Use bauser window):

Frequency of pass band edge	=	300 Hz	
Gain in pass band H ₁	=	– 2dB	
Frequency from which stop band begins	=	800 Hz	
Gain (attenuation) in stop band, H_2	=	– 25 dB	
Sampling frequency	=	3000 Hz	
			115

(15 marks) [4 × 15 = 60 marks]