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Total number of printed pages – 2

B. Tech
PCEC 4304

Sixth Semester Regular Examination – 2014

DIGITAL SIGNAL PROCESSING

BRANCH(S) : EC, ETC

QUESTION CODE : F 215

Full Marks – 70

Time : 3 Hours

Answer Question No. 1 which is compulsory and any **five** from the rest.

The figures in the right-hand margin indicate marks.

1. Answer the following questions :

2 × 10

(a) The following analog signal is sampled at 8,000 samples per second :

$$x(t) = \sin(1250t) + 2\cos(1000\pi t)$$

What is corresponding discrete time signal after sampling?

(b) State the time shifting properties of Z-transform.

(c) What is approximate transition width of main lobe in the rectangular window? What happens to it if you double the filter length?

(d) State the magnitude response of the system described by

$$y(n) = 0.2Y(n-2) + X(n).$$

(e) Draw the basic structure of 1st order digital IIR filter.

(f) What do you mean by linear phase characteristics of FIR filter? Why IIR filters does not have linear phase characteristics?

(g) What is condition of stability of a LTI system in Z-domain?

(h) State the limitations of impulse invariance method for realizing IIR filter.

(i) State circular frequency shifting property of DFT.

(j) Establish the relation between DFT and Z-transform.

P.T.O.

2. (a) Find the Z-transform of the following signal : 6
 (i) $x(n) = a^n u(n)$
 (ii) $x(n) = n^2 u(n-1)$
- (b) Using convolution property, determine Z-transform of the signal 4
 $x(n) = (n+1) u(n)$
3. (a) Explain various windows used for design of linear phase FIR filter. 5
 (b) Find the impulse response of LTI system whose frequency response is described as 5

$$H(e^{j\omega}) = \begin{cases} 1 & \text{For } |\omega| < \pi/4 \\ 0 & \text{otherwise} \end{cases}$$
4. (a) Compare the merits and limitations of impulse invariance and bilinear transformation methods of IIR filter design. 5
 (b) Design a single pole low-pass digital filter with 3-dB bandwidth of 0.2π , using bilinear transformation applied to the analog filter. 5
- $$H(s) = \frac{\Omega}{s + \Omega}$$
- Where Ω is the 3-dB bandwidth of an analog filter. 5
5. (a) Consider the casual system 5
 $Y(n) = 0.75y(n-1) - 0.125y(n-2) + x(n) + 0.3x(n-1)$
 Obtain direct form-I and form-II structure.
- (b) Explain linear phase and stability property of FIR and IIR filters. 5
6. (a) The DFT of $x(n)$ is described as $X(k) = \{1, -1 + 2j, -1, 1 + 2j\}$. Find the DFT of $x^2(n)$. 5
 (b) Explain how DFT can be used in linear filtering the discrete signal. 5
7. (a) Explain Decimation in frequency FFT algorithm. 5
 (b) What is N in N-point DFT ? Find 4-point DFT of the discrete signal, $X(n) = \{0, 1, 2, 3\}$. 5
8. Write short notes on any **two** : 5×2
 (a) Linear phase FIR filter by frequency sampling method
 (b) Stability of LTI system
 (c) Symmetric FIR filter
 (d) Use of DFT in linear filtering.

