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Total number of printed pages – 3

B. Tech
PCEC 4304

Sixth Semester Regular / Back Examination – 2015

DIGITAL SIGNAL PROCESSING

BRANCH(S) : AEIE, IEE

QUESTION CODE : J 203

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

- Express the transfer function of a 1st order digital FIR filter using z-transform.
- State the scaling property of the z transform.
- What is paley-weiner theorem ? What is its importance ?
- State the magnitude response of the system described by
$$y(n) = 0.5Y(n - 1) + X(n - 2)$$
- How many real multiplication and addition is required for computation of N-point DFT ?
- What are the two conditions that must be satisfied for a filter to behave as linear phase characteristics ?
- How ripples in the pass band of FIR filters can be eliminated ?
- State the limitations of bilinear transformation method for realizing IIR filter.
- State two practical difficulties that are generally encountered during designing of FIR filter.
- What is adaptive in adaptive filter ? How it is different from classical filter ?

P.T.O.

2. (a) Find the Z-transform of the following signal using property. 4

$$x(n) = 1 \quad \text{for } 0 < n < 11$$

$$= 0 \quad \text{otherwise}$$
- (b) Determine the response of the system 6

$$y(n) = 0.8y(n - 1) + 0.2Y(n - 2) + X(n)$$
to the input signal $x(n) = \delta(n) - \delta(n - 1)$
3. (a) Explain the Pole –Zero pattern of FIR filters with suitable example. 5
(b) Find the impulse response of LTI system whose frequency response is described as 5
- $$H(e^{j\omega}) = 1 \quad \text{For } |\omega| < \pi/4$$
- $$= 0 \quad \text{otherwise}$$
4. (a) Establish the relation between ω and Ω using bilinear transformation. And then, bring out a mapping between them. 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. (a) Consider the casual system 6

$$Y(n) = 0.75y(n - 1) - 0.125y(n - 2) + x(n) + 0.3x(n - 1)$$
Obtain direct form I and form-II structure.
- (b) Explain the frequency sampling structure of FIR filter. 4
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) What is N-point DFT ? Find 4-point DFT of the discrete signal,
 $X(n) = \{0, 1, 2, 1\}$. 5
7. (a) Explain Decimation in frequency FFT algorithm 5
(b) What is adaptive filter ? How it can be used to identify an unknown system ? 5
8. Write short notes on any **two** of the following : 5×2
- (a) Adaptive Noise Cancelling
 - (b) Stability of LTI system
 - (c) Windows used in designing FIR filter
 - (d) Use of DFT in linear filtering.

