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Total Number of Pages: 02

**B.TECH**  
**PCEC4304**

**6<sup>th</sup> Semester Regular / Back Examination 2016-17**

**DIGITAL SIGNAL PROCESSING**

**BRANCH(S): CSE, ELECTRICAL, MECH**

**Time: 3 Hours**

**Max Marks: 70**

**Q. CODE: Z683**

**Answer Question No.1 which is compulsory and any five from the rest.  
The figures in the right hand margin indicate marks.**

**Q1 Answer the following questions: (2 x 10)**

- a) Which standard discrete signal is usually used to analyze discrete time system? Justify with necessary mathematical expression using Z-transform
- b) Derive the relation between Z-transform and Laplace transform.
- c) Determine the pole-Zero plot for the signal  
$$x(n) = a^n u(n)$$
- d) A signal  $x(t) = 22 \sin(1500\pi t)$  is sampled with a sampling period of 120 micro second and passed through a ideal low pass filter with 3-dB frequency of 10 kHz.. Find out the frequencies that will appear at the output of low pass filter?
- e) Find out the real multiplication and real additions that are required to compute 16 point DFT using direct computation and FFT algorithm?
- f) What is meant by radix-2 FFT?
- g) Are the zeros of a stable linear phase FIR filters lie outside the unit circle of Z-plane? Justify
- h) Why IIR filter do not have linear phase?
- i) Why FIR filters are inherently stable?
- j) What is Gibbs phenomenon?

**Q2 a) Determine the phase and magnitude of the system (5)**

$$Y(n) - y(n - 1) = 0.25y(n - 2) + x(n) - x(n - 1)$$

**b) Plot pole zero pattern and stability test of the following impulse response system. (5)**

$$Y(n) = y(n - 1) - 0.52y(n - 2) + x(n) + x(n - 1)$$

- Q3 a)** The impulse response of LTI system is expressed as **(5)**

$$h(n) = 0.2^n u(n)$$

Find the value of A such that  $h(n) - A h(n - 1) = \delta(n)$

- b)** Prove that if  $x(n)$  is a real sequence and  $x(n) = -x(N - n)$  then its DFT is purely imaginary **(5)**

- Q4 (a)** Convert the analog filter with system function **(5)**

$$H(s) = \frac{s + 0.1}{(s + 0.1)^2 + 9}$$

Into a digital IIR filter using impulse invariance transformation method. The digital filter is to have resonant frequency of  $\pi/2$ .

- (b)** Obtain  $H(z)$  using bilinear transformation of the following transfer functions with  $T=1$  sec. **(5)**

$$h(s) = \frac{s}{s^2 + 2s + 1}$$

- Q5 a)** Consider the casual system **(6)**

$$Y(n) = -0.5y(n - 1) - 0.12y(n - 2) + 0.7x(n) - 0.252x(n - 2)$$

Justify whether the system is FIR or IIR and then obtain a transpose structure of the system

- b)** Find the impulse response of LTI system whose frequency response is described as **(4)**

$$H(e^{j\omega}) = 1 \quad \text{For } |\omega| < \pi/4 \\ = 0 \quad \text{otherwise}$$

Is such LTI system practically realizable? Justify your answer.

- Q6 a)** Compare FIR with IIR filter with suitable example. **(5)**

- b)** Prove that convolution of two signal in discrete time domain is equal to multiplication in discrete frequency domain using DFT. **(5)**

- Q7 a)** Explain Decimation in time FFT algorithm **(5)**

- b)** What are the practical difficulties in designing FIR filter? Explain with a suitable example. **(5)**

- Q8** **Write short notes on (Any two)** **(5 x 2)**

- a)** Circular convolution  
**b)** Symmetric and asymmetric condition of FIR filter  
**c)** Adaptive filter  
**d)** FIR filter using windowing technique