Total No.	of (Questions	:	10]
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SEAT No.:	
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P2963

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[5460]-552

T.E. (E & TC) (End Semester) DIGITAL SIGNAL PROCESSING DSP (Digital Signal Processing) (2015 Pattern)

Time : 2½ *Hours*]

[Max. Marks: 70

Instructions to the candidates:

- 1) Neat diagram must be drawn whenever necessary.
- 2) Figures to the right side indicate full marks.
- 3) Your answers will be valued as a whole.
- 4) Assume suitable data if necessary.
- Q1) a) Show the mapping between analog frequencies and digital frequencies.[4]
 - b) Explain the concept of Eigen values and Eigen vector, Find the Eigen values of given matrix A as given below: [6]

$$A = \begin{bmatrix} 1 & 1 & 2 \\ 0 & 1 & 3 \\ 0 & 0 & 2 \end{bmatrix}$$

OR

- Q2) a) Explain the cyclic property of twiddle factor for 8 point DFT. [4]
 - b) Find linear convolution using overlap save method of the following sequences: [6]

$$x(n) = \{1, 2, -1, 2, 3, -2, -3, -1, 1, 1, 2, -1\}$$
 and $h(n) = \{1, 2, 3\}$

- Q3) a) Explain how ROC is important to determine the causality and stability of LTI discrete time system.[4]
 - b) Draw signal flow graph of radix-2 DIF FFT algorithm for N=4. [6]

- **Q4)** a) State and prove the convolution property of Z transform. [6]
 - b) Show relation between Fourier Transform and Z-Transform. [4]
- Q5) a) Comparison between Impulse invariance and bilinear transformation method. What is prewarping? [9]
 - b) A Digital filter has frequency specification as:

Pass band Frequency =
$$\omega_p = 0.2\pi$$

Stop band Frequency = $\omega_s = 0.3\pi$

What are the corresponding specifications for pass band and stop band frequencies in analog domain if,

- i) Impulse invariance technique is used for designing.
- ii) BLT method is used for Designing.

OR

Q6) a) Obtain direct form I and Direct form II realization of a LTI system described difference equation as given below: [9]

$$3y(n) - 2y(n-1) + y(n-2) = 4x(n) - 3(n-1) + 2x(n-2)$$

- b) Give the properties and characteristics of chebyshev and Butterworth filter, Give salient features of Low Pass Butterworth Filter. [9]
- Q7) a) What is Gibbs Phenomenon? Explain Importance of windowing functions to design FIR filter in details.[8]
 - b) Distinguish between IIR and FIR filter, Why ideal filter cannot be realized practically? [8]

OR

Q8) a) Design an FIR filter with Hamming window for desired impulse response given below: [8]

$$H_d(w) = e^{-3jw}; -\frac{\pi}{4} \le w \le \frac{\pi}{4}$$

$$=0:\frac{\pi}{4} \le w \le \pi$$

- b) Explain finite word length effect in Digital FIR filter. What do You understand by linear phase response? [8]
 Q9) a) Explain different types audio crossover systems? Why digital crossover
 - is preferred? [8]
 - b) Explain compact disc recording system with the help of block schematic.

8]

OR

Q10) Write short notes: (Any Two)

[16]

- a) Vibration analysis for Defective Gear Teeth.
- b) Voice Coders (Vocoders)
- c) Speech noise Reduction
- d) Explain how DSP is very useful to suppress the interference in ECG.

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