Total No.	of Questions	:	10
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## TE (E & TC) (Engg.) (Semester - I) DIGITAL SIGNAL PROCESSING (2012 Pattern)

*Time* : 2½ *Hours*]

[Max. Marks:70

Instructions to the candidates:

- 1) Answer O1 or O2, O3 or O4, O5 or O6, O7 or O8, O9 or O10
- 2) Assume suitable data, if necessary.
- 3) Neat diagrams must be drawn wherever necessary.
- 4) Figures to the right indicate full marks.
- **Q1**) a) What are the advantages of Digital signal processing over analog signal processing. [4]
  - b) Check the orthogonality of the vectors.

[4]

$$A1 = (1,3,2)^T A2 = (3,-1,0)^T A3 = \left(\frac{1}{3},1,-\frac{5}{3}\right)^T$$

c) State and prove any two properties of Z transform

[2]

OR

- **Q2)** a) Compute 4 point DFT of a sequence  $x(n) = \{0,1,2,3\}$  using Decimation In Time FFT algorithm [4]
  - b) Compute the circular convolution of the following sequences  $x(n) = \{4,3,2,1\}$   $h(n) = \{2,1,2,1\}$
  - c) Write a note on, "Overlap and save method"

[2]

[4]

Q3) a) What is the relationship between DFT and DTFT

[3]

[4]

- b) Compute the IDFT of the following sequence  $X(k) = \{4,1-j,-2, 1+j\}$
- c) By using partial fraction method find the Inverse Z transform of [3]

$$x(z) = \frac{z}{(z+2)(z-1)}$$

Show that the computational complexity is reduced if 64 point DFT is

computed using Radix -2 DIT FFT algorithm [3] Compute the z transform and draw ROC of the following sequences [4] b)  $x(n) = a^{n}u(n)$  for  $n \ge 0$   $x(n) = 2^{n}u(n) + 3^{n}u(-n-1)$ State the properties of Region of convergence [3] c) **Q5)** a) The system transfer function of analog filter is given by [8]  $H(s) = \frac{s + 0.1}{(s + 0.1)^2 + 9}$ using Impulse Invariance method determine H(z) Assume T = 1Sec Compare Bilinear transformation and Impulse Invariance method b) [6] Show the mapping between analog and Digital frequencies c) [4] Obtain direct form I, Direct Form II, Cascade and Parallel form realization **Q6)** a) of a system described by [10] y(n) = 0.75y(n-1) - 0.25y(n-2) + x(n) + 2x(n-1) - x(n-2)A digital filter has specifications as: b) Passband frequency =  $Wp = 0.4\Pi$ , Stopband frequency =  $Ws = 0.6\Pi$ What the corresponding specifications are for pass band and stop frequencies in analog domain if [6] Impulse Invariance Technique is used for designing i) Bilinear Transformation method is used for designing Compare Rectangular window with Hanning window [2] c) State the characteristics of FIR filter **Q7**) a) [6] Design a linear phase FIR low pass filter with cut off frequency fo 0.5 b) rad/sample by taking 11 samples of ideal frequency response [10]

*04*) a)

The frequency characteristics of ideal low pass filter is given as *Q8*) a) [10]

$$H(w) = 1 for -\frac{\pi}{2} \le w \le \frac{\pi}{2}$$
$$= 0 for \frac{\pi}{2} \le w \le \pi$$

Design digital FIR filter using fourier series method

- Find the value of h (n) for N = 9i)
- Find the system function H(Z)ii)
- Realize the linear phase FIR Filter [6] b) y(n) = x(n) + 2x(n-1) + 2x(n-2) + x(n-3)
- Explain the application of DSP in Voice processing **Q9**) a) [6]
  - Design a two stage decimator for the following specifications: [10] b) Sampling rate of an input signal = 20 KHZ

Down sampler D = 80

Passband = 0 to 40 Hz

Transition band = 40 to 50 H

Passband ripple = 0.02

Stopband ripple = 0.002

- Explain important salient features of TMS320C 6713 DSP processor *Q10*)a) and draw its functional block diagram. [8]
  - ors, and the second sec Explain the necessity of Barrel shifter b) [4]
  - Compare DSP processors with Microprocessors c) [4]