R - 1007

Seat No.

T.E. (Electrical Engg.) (Semester - V) Examination, 2009 DIGITAL SIGNAL PROCESSING (New)

Day and Date: Saturday, 5-12-2009 Time: 10.30 a.m. to 1.30 p.m.

Max. Marks: 100

Note: Attempt any three questions from each Section.

SECTION-1

i. a) Determine the response of the system defined by:

$$y(n) = \frac{5}{6}y((n-1) - \frac{1}{6}y(n-2) + x(n)$$

to the input signal

$$x(n) = \delta(n) - \frac{1}{3}\delta(n-1)$$
 where

 $\delta(n)$ is the impulse signal.

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b) Derive the condition of the stability for system and check the stability property of system described as

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$$y(n) = e^{-X(n)}$$

2. a) Derive the relation between DFT and Z-transform for N-point sequence.

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b) If
$$X_1(n) = [1,2,1,2]$$
 and $X_2(n) = [2,1,2,1]$

obtain the circular convolution using DFT - IDFT property.

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- 3. a) How computation time is reduced in case of fast fourier algorithms? Show that for N-pointing DFT equation how it reduced down for multiplication and for addition?
 - b) Explain in detail the process of decimation of signal X (n) by a factor M.

: M. 8

4. a) Using DIT-FFT algorithm explain the signal flow graph for 8-point DFT sequence and solve for 8-point DFT using DIT-FFT if

$$X(n) = [1, 2, 3, 4, 3, 2, 1, 0].$$

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b) Explain the requirement of changing sampling rate and comment on the benefits of up sampling and down sampling.

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SECTION-II

5. a) Obtain the direct form II, cascade and parallel realization of:

$$H(z) = \frac{1 + \frac{1}{4}z^{-1}}{\left(1 + \frac{1}{2}z^{-1}\right)\left(1 + \frac{1}{2}z^{-1} + \frac{1}{4}z^{-2}\right)}$$
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- b) Explain with block diagram the architecture of DSP TMS 320 controller.
- 6. a) Design a second order Band Pass Butterworth filter that meets following specifications:

Lower out off frequency = 210 Hz

Upper cut off frequency = 330 Hz

Sampling frequency = 960 sps.

Also realize the resulting filter structure in direct form 1.

- b) Explain quantization error in Rounding and truncation.
- 7. a) Design an FIR low pass filter for the specifications given as:

 $H_d(f) = 1$ for $0 \le t \le 5$ Hz

= 0 Elsewhere

The sampling frequency is 20 sps and the impulse response is to have duration of 1 sec.

- b) What is window function? Explain the need of windowing and different window functions used.
- 8. a) Explain effect of finite word length on the stability and frequency response of digital filters.
 - b) Design a Digital filter using Bilinear transformation method if

$$M(s) = \frac{5s+1}{s^2+0.4s+1}$$

and the cutoff frequency is 10 Hz with the sampling frequency of 60 Hz.