

--	--	--	--	--	--	--	--	--	--

Fifth Semester B.E. Degree Examination, May/June 2010
Digital Signal Processing

Time: 3 hrs.

Max. Marks:100

**Note: Answer any FIVE full questions, selecting
at least TWO questions from each part.**

PART - A

- 1 a. For the following sequences, find: (12 Marks)
 - i) N-point DFT of $x(n) = \cos \frac{2\pi}{N} K_0 n$
 - ii) 5-point DFT of $x(n) = \{1, 1, 1\}$
- b. Find IDFT for the sequence : $x(k) = \{5, 0, (1-j), 0, 1, 0, (1+j), 0\}$ (08 Marks)
- 2 a. State and prove circular frequency shift property of DFT. (04 Marks)
- b. Compute the circular convolution of the sequences $x_1(n) = \{2, 1, 2, 1\}$ and $x_2(n) = \{1, 2, 3, 4\}$ using DFT and IDFT method. (08 Marks)
- c. Find the output $y(n)$ of a filter whose impulse response is $h(n) = \{1, 2\}$ and the input signal to the filter is $x(n) = \{1, 2, -1, 2, 3, -2, -3, -1, 1, 2, -1\}$ using overlap-save method. (08 Marks)
- 3 a. Determine the number of complex multiplications, complex additions and trigonometric functions, required for direct computation of N-point DFT, (10 Marks)
- b. How many complex multiplications and additions are required for 64-point DFT in FFT? (04 Marks)
- c. Prove : i) Symmetry and ii) Periodicity property of a twiddle factor. (06 Marks)
- 4 a. Develop Radix-2 N-point DIT-FFT algorithm and draw the signal flow graph. (12 Marks)
- b. Obtain 8-point DFT of the sequence, $x(n) = \{2, 1, 2, 1, 0, 0, 0, 0\}$. Using Radix-2 DIF-FFT algorithm. Show clearly all the intermediate results. (08 Marks)

PART - B

- 5 a. Given $|H_a(j\Omega)|^2 = \frac{1}{1+16\Omega^4}$, determine the analog filter system function $H_a(S)$. (08 Marks)
- b. Derive an expression for 'N' and Ω_{cp} of Butterworth filter if passband and stopband attenuations are in dB. (08 Marks)
- c. Let $H(s) = \frac{1}{s^2 + s + 1}$ represents transfer function of a low pass filter with a passband of 1 rad/sec. Use frequency transformation to find the transfer function of the following analog filters:
 - i) A LPF with $\Omega'p = 10$ rad/sec
 - ii) A HPF with $\Omega'p = 100$ rad/sec. (04 Marks)
- 6 a. Derive an expression for frequency response of a symmetric impulse response for N-odd. (08 Marks)

Important Note : 1. On completing your answers, compulsorily draw diagonal cross lines on the remaining blank pages.
2. Any revealing of identification, appeal to evaluator and /or equations written eg, 42+8 = 50, will be treated as malpractice.

- 6 b. A lowpass filter is to be designed with the following desired frequency response:

$$H_d(e^{j\omega}) = H_d(\omega) = \begin{cases} e^{-j\omega}, & |\omega| < \pi/4 \\ 0, & \pi/4 < |\omega| < \pi \end{cases}$$

Determine the filter coefficients $h_d(n)$ and $h(n)$ if $\omega(n)$ is a rectangular window defined as follows:

$$W_R(n) = \begin{cases} 1, & 0 \leq n \leq 4 \\ 0, & \text{otherwise} \end{cases}$$

Also, find the frequency response, $H(\omega)$ of the resulting FIR filter. (12 Marks)

- 7 a. Derive the expression for the bilinear transformation, to transform an analog filter to a digital filter, by trapezoidal rule and explain the mapping from s-plane to z-plane. (08 Marks)
- b. Convert the analog filter with system function $H_a(s) = \frac{(s+0.1)}{(s+0.1)^2+9}$ into a digital filter (IIR) by means of impulse invariance method. (08 Marks)
- c. Given the analog transfer function, $H(s) = \frac{(s+2)}{(s+1)(s+3)}$. Find $H(z)$, using matched z-transform design. The system uses sampling rate of 10Hz ($T = 0.1$ sec). (04 Marks)
- 8 a. Obtain direct form I, direct form II, cascade and parallel structure for the system described by $y(n) = -0.1y(n-1) + 0.72y(n-2) + 0.7x(n) - 0.252x(n-2)$. (16 Marks)
- b. Obtain the direct form realization of linear phase FIR system given by $H(z) = 1 + \frac{2}{3}z^{-1} + \frac{15}{8}z^{-2} + \frac{2}{3}z^{-3} + z^{-4}$. (04 Marks)

* * * * *