

END TERM EXAMINATION

FIFTH SEMESTER [B.TECH./M.TECH.] DECEMBER 2015

Paper Code: IT307

Subject: Digital Signal Processing

Time : 3 Hours

Maximum Marks :60

Note: Attempt any five questions. Use of calculator is permitted.

- Q1. Let $x[n]$, $y[n]$ and $w[n]$ denote three arbitrary sequences. Show that: (6+6)
- (a) Discrete convolution is commutative, i.e.,
 $x[n] * y[n] = y[n] * x[n]$
- (b) Discrete convolution is associative, i.e.,
 $x[n] * (y[n] * w[n]) = (x[n] * y[n]) * w[n]$
- Q2. For each of the following systems, determine whether or not the system is (1) stable, (2) casual, (3) linear, and (4) shift-invariant: (2x6)
- (a) $y[n] = g[n] x[n]$ (d) $y[n] = x[n-n_0]$
- (b) $y[n] = \sum_{k=n_0}^n x[k]$ (e) $y[n] = e^{x[n]}$
- (c) $y[n] = \sum_{k=n-n_0}^{n+n_0} x[k]$ (f) $y[n] = ax[n] + b$ where $a, b > 0$
- Q3. Find the z-transform of the following: (6+6)
- (a) $x[n] = a^n \sin(\omega n) u[n]$ (b) $x[n] = a^n u[n] - b^n u[-n-1]$, (a and b) < 1 , $b > a$
- Q4. Determine the impulse response of the FIR filter whose impulse response is (6+6)
- $h[n] = \{1, -2, 3\}$ and the input signal is $x[n] = \{1, -2, 3, -4, 5, -6, 7, -8, 9\}$. Use any method for calculation of the concerned DFT and then use the following method for calculation of the linear convolution:
- (a) Overlap Save (b) Overlap Add
- Q5. A system is described by the difference equation (4x3)
- $y[n] - (3/4)y[n-1] + (1/2)y[n-2] = x[n] + (1/2)x[n-1]$
- Draw a signal flow graph to implement this system in each of the following forms:
- (a) Direct form I,
 (b) Direct form II,
 (c) Cascade and
 (d) Parallel
- Q6. Design a digital lowpass Butterworth filter with a passband magnitude characteristic that (12)
- is constant within 0.75 dB for a frequency below $\omega = 0.2613\pi$ and stopband attenuation of at least 20dB for frequencies between $\omega = 0.4018\pi$ and π . Use the Impulse Invariant Design Method.
- Q7a. Determine the DFT of the signal $x[n] = \{2, 1, 4, 6, 5, 8, 3, 9\}$ by decimation in time FFT. (8)
- Q7b. What is the time complexity of the (naive) DFT algorithm, and the time complexity of the radix-2 Decimation in time FFT algorithm. (4)
- Q8. Write short notes on any two of the following: (6+6)
- (a) Sampling Theorem
 (b) FIR filter design with windows
 (c) Bi-linear transformation for Filter design

P

147