

**328651(28)**

**BE (6<sup>th</sup> Semester)  
Examination, Nov.-Dec., 2017  
(New Scheme)**

**Digital Signal Processing**

**Time Allowed : 3 hours**

**Maximum Marks : 80**

**Minimum Pass Marks : 28**

- Note :** (i) Part (a) of each question is compulsory.  
(ii) The figures in the right-hand margin indicate marks.

**Unit-I**

Attempt any Two parts from (b), (c) and (d)

1. (a) Write a short note on discrete cosine transformation. [2]  
(b) Explain discrete Fourier transform and calculate DFT for the input signal  $x(n) = \{0, 1, 2, 3\}$ . [7]

- (c) Let  $x_1(n)$  and  $x_2(n)$  be the following two 4-point sequences :

$x_1(n) = \{1, 2, 2, 1\}$

$x_2(n) = \{1, -1, -1, 1\}$

- (i) Determine their linear convolution  $x_3(n)$ .  
(ii) Compute the circular convolution  $x_4(n)$  so that it is equal to  $x_3(n)$ . [7]  
(d) Explain discrete time Fourier transform (DTFT). Also explain its properties. [7]

**Unit-II**

Attempt any One part from (b) and (c)

2. (a) Design transfer function for the 1<sup>st</sup> and 2<sup>nd</sup> order of Butterworth filter. [2]  
(b) Obtain direct form I, II and cascade form realization of a system transfer function - described by

$y(n) - \frac{3}{4}y(n-1) + \frac{1}{8}y(n-2)$

$= x(n) + \frac{1}{2}x(n-1)$  [14]

- (c) Write short notes on signal flow graph and frequency sampling. Draw cascade canonical IIR filter for the transfer function shown below : [14]

$H(z) = (1 + Z^{-1}) / (1 - Z^{-1} + \frac{1}{2}Z^{-2})(1 - Z^{-1} + Z^{-2})$

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## Unit-III

Attempt any One part from (b) and (c)

3. (a) What is Gibbs phenomenon? [2]  
 (b) Design an ideal low-pass filter with a frequency response

$$H_d(e^{j\omega}) = 1 \quad \text{for } -\pi/2 \leq \omega \leq \pi/2$$

$$= 0 \quad \text{for } \pi/2 \leq \omega \leq \pi$$

Find the values of  $h(n)$  for the order of filter  $N=9$ . Find  $H(z)$  using rectangular window method. Also find  $H(e^{j\omega})$ . [14]

- (c) Design a filter with

$$H_d(e^{j\omega}) = e^{-j3\omega} \quad -\pi/4 \leq \omega \leq \pi/4$$

$$= 0 \quad \text{otherwise}$$

Find  $H(z)$  using Hamming and Hamming window for the order of filter  $N=7$ . [14]

## Unit-IV

Attempt any Two parts from (b), (c) and (d)

4. (a) Define impulse invariance method of transformation. [2]  
 (b) An analog filter has a transfer function  $H(s) = 1/(s^2 + 6s + 9)$ . Design a digital filter using BLT. [7]  
 (c) Prove that  $\Omega_c = \Omega_p / (10^{0.1 A_{p-1}})^{1/2N}$   
 $= \Omega_s / (10^{0.1 A_{s-1}})^{1/2N}$ . [7]

- (d) Design a Chebyshev filter with a maximum passband attenuation of 2.5 dB, at  $\Omega_p = 20$  rad/sec and stopband attenuation of 30 dB at  $\Omega_s = 50$  rad/sec. [7]

## Unit-V

Attempt any Two parts from (b), (c) and (d)

5. (a) What is the need for anti-imaging filter after upsampling a signal? [2]  
 (b) Assume that the input signal  $x(n) = \{1, 2, 3, 2, 3, -1, 3, 5, 6, 9, 1, -1, \dots\}$ . Find the output  $y(n)$  for the following cases and give comments : [7]  
 (i) Up sampled by 2-fold then down sample by 3-fold  
 (ii) Down sample by 3-fold then up sample by 2-fold  
 (iii) Up sample by 4-fold then down sample by 2-fold  
 (iv) Down sample by 2-fold then up sample by 4-fold  
 (c) Plot the signals and their corresponding spectra for rational sampling rate conversion by (i)  $1/D=5/3$  and (ii)  $1/D = 3/5$ . Assume that the spectrum of the input signal  $x(n)$  occupies the entire range  $-\pi \leq \omega \leq \pi$ . [7]  
 (d) Answer the following : [7]  
 (i) Show that the Up-sampler and Down-sampler are time-variant systems.  
 (ii) Give any five cases where the multirate DSP is used.