

CBCS SCHEME

USN

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

15EE63

Sixth Semester B.E. Degree Examination, July/August 2021

Digital Signal Processing

Time: 3 hrs.

Max. Marks:80

Note: Answer any FIVE full questions.

- 1 a. Define DFT and IDFT. State and prove linearity property. (05 Marks)
b. Find the 4 point DFT of the sequence $x(n) = \cos\left(\frac{n\pi}{4}\right)$. (06 Marks)
c. Determine the IDFT of 4 point sequence $X(k) = [4, -j2, 0, j2]$. (05 Marks)
- 2 a. Find the output $y(n)$ of filter whose impulse response is $h(n) = [1, 1, 1]$ and the input signal to the filter is $x(n) = [3, -1, 0, 1, 3, 2, 0, 1, 2, 1]$ using overlap save method. (08 Marks)
b. The two response $x_1(n)$ and $x_2(n)$ are given as follows $x_1(n) = [2, 1, 2, 1]$ and $x_2(n) = [1, 2, 3, 4]$ find out the sequence $x_3(m)$ which is equal to circular convolution of above two sequences $x_3(m) = x_1(n) \otimes x_2(n)$. (08 Marks)
- 3 a. Develop an 8 point DIT-FFT algorithm using signal flow graph for the following sequence $x(n) = [1, 1, 1, 1, 0, 0, 0, 0]$, show all the intermediate results on the signal flow graph. (10 Marks)
b. Explain Radix-2 DIF-FFT algorithm. (06 Marks)
- 4 a. The DFT $X(K)$ of sequence is given as $X(K) = \{0, 2\sqrt{2}(1-j), 0, 0, 0, 0, 0, 2\sqrt{2}(1+j)\}$. Determine the corresponding time sequence $x(n)$ and write its signal flow graph. (10 Marks)
b. Explain computational efficiency. (06 Marks)
- 5 a. The system function of an analog filter is given as $H_a(s) = \frac{1}{(s+1)(s+2)}$. Obtain $H(z)$ using impulse invariant method. Take sampling frequency of 5 samples/sec. (08 Marks)
b. Design an analog filter with maximally flat response in the passband and an acceptable attenuation of -2dB at 20 radians/sec. The attenuation in the stop band should be more than 10dB beyond 30 radians/sec. (08 Marks)
- 6 a. Distinguish between Butterworth and Chebyshev filter. (04 Marks)
b. Compare impulse in variance and Bilinear transformation method. (04 Marks)
c. Design a unit band width 3dB digital Butterworth of filter of first order by using bilinear transformation. (08 Marks)
- 7 a. Determine the order of a Chebyshev digital lowpass filter to meet the following specifications. In the passband extending from 0 to 0.25π , a ripple of not more than 2dB is allowed. In the stopband extending from 0.4π to π attenuation can be more than 40dB. Use bilinear transportation method. (08 Marks)
b. Explain direct form structure for IIR systems and cascade form structure for IIR systems. (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.

- 8 a. A difference equation describing a filter is given below :
$$y(n) - \frac{3}{4}y(n-1) + \frac{1}{8}y(n-2) = x(n) + \frac{1}{2}x(n-1)$$

Draw direct form I and direct form II structures. (08 Marks)
- b. Realize the following system function :
- i) $H(z) = \left(\frac{1}{2} + z^{-1} + \frac{z^{-2}}{2} \right) \left(1 + \frac{z^{-1}}{3} + z^{-2} \right)$ by cascade realization. (08 Marks)
- ii) $H(z) = 3 + \frac{4z}{z - \frac{1}{2}} + \frac{2}{z - \frac{1}{4}}$ by parallel realization. (08 Marks)
- 9 a. Explain different types of windows. (08 Marks)
- b. Realize a linear phase FIR filter with the following impulse response give necessary equations : $h(n) = \delta(n) + \frac{1}{2}\delta(n-1) - \frac{1}{4}\delta(n-2) + \delta(n-4) + \frac{1}{2}\delta(n-3)$. (08 Marks)
- 10 a. Design linear phase FIR filters using windows. (08 Marks)
- b. Bring out the comparison between IIR and FIR filters. (08 Marks)

CMRIT LIBRARY
BANGALORE - 560 037