



B.Tech III Year II Semester Supplementary Examinations, June 2022
Digital signal processing
(ECE)

Maximum Marks: 70

Date: 20.06.2022 Duration: 3 hours

- Note:
1. This question paper contains two parts A and B.
 2. Part A is compulsory which carries 20 marks. Answer all questions in Part A.
 3. Part B consists of 5 Units. Answer any one full question from each unit.
 4. Each question carries 10 marks and may have a, b, c, d as sub questions.

Part-A

All the following questions carry equal marks

(10x2M=20 Marks)

- 1 What is an LTI system.
- 2 Define the frequency response of a discrete-time system.
- 3 Define discrete Fourier series.
- 4 Obtain the circular convolution of the sequence $x(n)=\{1,2,3\}$, $h(n)=\{1,1,1\}$.
- 5 What is meant by bilinear transformation technique.
- 6 Distinguish between Analog filters and Digital filters.
- 7 What are the disadvantages of Fourier series method.
- 8 What are the desirable characteristics of the window.
- 9 What is the need for anti-imaging filter after up sampling a signal.
- 10 What are the effects of Dead band.

Part-B

Answer All the following questions.

(10M X 5=50Marks)

- 11 Explain the block diagram of DSP and its applications. [10]

OR

- 12 Determine the impulse response for the given system by using time response analysis

$$y(n) - \frac{1}{6}y(n-1) - \frac{1}{6}y(n-2) = x(n). [10]$$

- 13 Determine 8-point DFT of the sequence $x(n)=\{1,1,0,0,1,1,1,0\}$ [10]

OR

- 14 Find the DFT of a given signal $x(n)=\{1,2,3,4\}$ by using DIT-FFT algorithm. [10]

- 15 a) What are the advantages and disadvantages of digital filters. [5]
b) Design Butterworth low pass filter with the given specifications $\alpha_p = 1\text{dB}$, $\alpha_s = 30\text{dB}$ and $\Omega_p = 200\text{rad/sec}$, $\Omega_s = 600\text{rad/sec}$. Convert this filter into digital filter by using step invariant method. [5]

OR

- 16 Realize the following system [10]
 $y(n) = -0.1y(n-1) + 0.72y(n-2) + 0.7x(n) - 0.252x(n-2)$ by using
i) Direct form-I ii) Direct form-II

- 17 Design an ideal low pass filter whose frequency response [10]

$$H_d(e^{j\omega}) = 1 \text{ for } -\pi/2 \leq \omega \leq \pi/2$$
$$= 0 \text{ for } \pi/2 \leq \omega \leq \pi.$$

Find the values of $h(n)$ for $N=11$. Find $H(Z)$

OR

- 18 a) Explain the Design of FIR digital filter by using fourier series method. [5]
b) What do you understand by linear phase response. [5]
- 19 a) Explain the process of decimation using relevant expressions and block diagram. [5]
b) What is the need of Multirate signal processing. [5]

OR

- 20 a) Compare fixed point and floating point arithmetic operations. [5]
b) Explain the methods used to prevent overflow. [5]