

## B.E. INSTRUMENTATION AND ELECTRONICS ENGINEERING THIRD YEAR SECOND SEMESTER – 2018

SUBJECT: Digital Signal Processing

Time: Three hours

Full Marks 100

Each question carries 10 marks.

**Module:1**

1. a) Express the sequence

$$x(n) = \begin{cases} 1 & n = 0 \\ 2 & n = 1 \\ 3 & n = 2 \\ 0 & \text{else} \end{cases}$$

as a sum of scaled and shifted unit steps.

- b) Consider the discrete time sequence

$$x(n) = \cos\left(\frac{n\pi}{8}\right)$$

Find two different continuous time signals that would produce this sequence when sampled at  $f_s = 10\text{kHz}$ .**Or**

- a) Consider a sinusoidal signal
- $\sin(\omega_0 t)$
- of period
- $T_0 = 2\pi/\omega_0$
- . It is sampled with period
- $T_s$
- . Find the condition to be satisfied from the sampled signal to be periodic.

- b) Find whether i)
- $\sin(0.1\pi k)$
- and ii)
- $\sin(0.3\pi k)$
- are periodic or not. Also compare the properties of these two signals in analog domain and digital domain.

**Module:2**

2. a) The input to a linear shift-invariant system is
- $x(n) = \cos(n\omega_0)$
- . Find the output if the real valued unit sample response of the system is
- $h(n)$
- .

- b) Determine the response of the system with impulse response

$$h(n) = \left(\frac{1}{2}\right)^n u(n)$$

to the input signal

$$x(n) = 5 - 5\sin\left(\frac{\pi}{2}n\right) + 10\cos(\pi n), -\infty < n < \infty$$

**Or**

- a) Determine whether the system is shift-invariant:

$$y(n) = x(n) + x(n-1) + x(n-2).$$

- b) Determine whether the system is linear:

$$y(n) = \log(x(n)).$$

- c) Determine whether the system is causal:

$$y(n) = x(n) + x(n-1) + x(n-2).$$

- d) Determine whether the system is stable:

$$y(n) = \cos(x(n)).$$

**Module:3: Answer any three**

3. a) Find the 10-point IDFT of

$$X(k) = \begin{cases} 3, & k = 0 \\ 1, & 1 \leq k \leq 9 \end{cases}$$

- b) Consider the finite length sequence

$$x(n) = \delta(n) + 2\delta(n-5)$$

Find the 10 point discrete Fourier transform.

4. a) Determine the z-transform of

$$x(n) = -\left(\frac{1}{2}\right)^n u(-n-1) + 2^n u(-n-1)$$

and depict the ROC and the locations of poles and zeros in z-plane.

- b) Determine the z-transform of

$$x(n) = \left(\frac{1}{2}\right)^n u(n) + 2^n u(-n-1)$$

and depict the ROC and the locations of poles and zeros in z-plane.

- c) Determine the z-transform and the pole-zero plot for the signal

$$x(n) = \begin{cases} 1, & 0 \leq n \leq N-1 \\ 0, & \text{elsewhere} \end{cases}$$

5. Write short notes on (any two)

- a) Gibb's phenomenon
- b) Phase delay and group delay
- c) Limitations of DFT.

6. Find the transfer function and then pole zero pattern of the following sequence:

	h(-2)	h(-1)	h(0)	h(1)	h(2)	h(3)	h(4)	h(n) otherwise
a)	1	-2.5	5.25	-2.5	1	0	0	0
b)	0	1	-.5	1	0	0	0	0
c)	0	0	1	-2.5	5.25	-2.5	1	0

Comment on the locations of the singularities and type of sequences, its specialties.

**Module:4: Answer any five**

7. A second order continuous time filter has a system function

$$H_a(s) = \frac{1}{s-a} + \frac{1}{s-b}$$

where  $a < 0$  and  $b < 0$  are real.

- a) Determine the locations of the poles and zeros of  $H(z)$  if the filter is designed using the bilinear transformation with  $T_s=2$ .
- b) Repeat part (a) for the impulse invariance technique.
- c) Comment on the locations of singularities.

8. Design a BPF of unity gain to pass frequencies in the range 1-4 rad/sample using Hanning window  $M=7$ . The causal Hanning window is given as

$$w(n) = \begin{cases} 0.5 - 0.5 \cos\left(\frac{2\pi n}{M-1}\right) & 0 \leq n \leq M-1 \\ 0 & \text{otherwise} \end{cases}$$

9. Design an IIR low-pass filter for the following specification:

Pass-band gain required : -1dB

Frequency up to which pass-band gain must remain more or less steady : 25Hz

Amount of attenuation required : -30dB

Frequency from which the attenuation must start : 75 Hz

Assume the sampling frequency to be 300Hz.

10. Determine cut off frequency/frequencies for a FIR filter whose transfer function is
- $H(z) = \frac{1}{2}(1 - z^{-2})$
  - $H(z) = \frac{1}{2}(1 - z^{-1})$
  - $H(z) = \frac{1}{2}(1 + z^{-2})$
  - $H(z) = \frac{1}{2}(1 + z^{-1})$
11. a) Show that the bilinear transformation maps the  $j\Omega$ -axis in the  $s$ -plane *onto* the unit circle,  $|z|=1$ , and maps the left-half  $s$ -plane,  $\text{Re}(s) < 0$  *inside* the unit circle,  $|z| < 1$ .
- b) What order Butterworth filter is necessary to design a digital low pass filter that has a passband cut off frequency  $\omega_p = 0.375\pi$  with  $\delta_p = 0.01$  and stopband cut off frequency  $\omega_s = 0.5\pi$  with  $\delta_s = 0.01$ .
12. a) An FIR linear phase filter has a unit sample response that is real with  $h(n)=0$  for  $n < 0$  and  $n > 7$ . If  $h(0) = 1$  and the system function has a zero at  $z=0.4 e^{j\pi/3}$  and a zero at  $z=3$ , what is  $H(z)$ ?
- b) The relationship between input and output of an FIR system is

$$y(n) = \sum_{k=0}^N b(k)x(n-k)$$

Find the coefficients  $b(k)$  of the smallest order filter that satisfies following condition:

- The filter has (generalized) linear phase
- It completely rejects a sinusoid of frequency  $\pi/3$
- The magnitude of the frequency response is equal to unity at  $\omega=0$  and  $\omega=\pi$ .