



### FIFTH SEMESTER B.TECH. (ELECTRICAL & ELECTRONICS ENGINEERING) END SEMESTER EXAMINATIONS, NOVEMBER-DECEMBER 2023

#### DIGITAL SIGNAL PROCESSING [ELE 3152]

REVISED CREDIT SYSTEM

Time: 3 Hours

Date: 08 December 2023

Max. Marks: 50

#### Instructions to Candidates:

- ❖ Answer **ALL** the questions.
- ❖ Missing data may be suitably assumed.

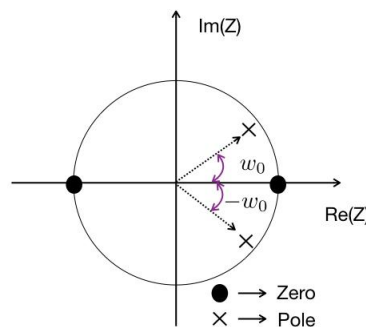
- | Q.No       | Questions                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                              | Marks       |
|------------|------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|-------------|
| <b>1A.</b> | <p>Given the discrete-time (DT) sequence <math>x(n) =  n </math> for <math>-3 \leq n \leq 3</math>, compute, sketch and label the following:</p> <ul style="list-style-type: none"> <li>i) <math>x(n)</math></li> <li>ii) <math>x(2n)</math></li> <li>iii) <math>x(2 - 2n)</math></li> <li>iv) <math>x(3n - 1)\delta(n - 1)</math></li> </ul>                                                                                                                                                                                                                                                                                                                          | <b>(02)</b> |
| <b>1B.</b> | <p>Determine the output of a filter whose impulse response is <math>h[n] = \delta[n] - 2\delta[n - 1] + \delta[n - 2]</math> and the input signal <math>x[n] = 3\delta[n + 2] - 2\delta[n] + 3\delta[n - 1] + 4\delta[n - 2] - \delta[n - 4] + 5\delta[n - 5] + 4\delta[n - 6] + 3\delta[n - 7]</math>. Use overlap-add method for linear filtering. Take sub-frame length of 3.</p>                                                                                                                                                                                                                                                                                   | <b>(04)</b> |
| <b>1C.</b> | <p>Calculate the 8-point DT sequence using Radix-2, DIT-FFT algorithm for the given DFT:</p> $X[k] = [20, -5.828 - j2.414, 0, -0.172 - j0.414, 0, -0.172 + j0.414, 0, -5.828 + j2.414].$ <p>Draw the Butterfly diagram. Indicate the result at each stage.</p>                                                                                                                                                                                                                                                                                                                                                                                                         | <b>(04)</b> |
| <b>2A.</b> | <p>Consider an analog signal <math>x(t) = 15 \cos\left(60\pi t - \frac{\pi}{4}\right) - 12 \cos(150\pi t)</math> applied to a sampling and reconstruction system. Determine</p> <ul style="list-style-type: none"> <li>i) Minimum value of sampling rate <math>F_s</math> to ensure <math>y(t) = x(t)</math>, where <math>y(t)</math> is the reconstructed signal.</li> <li>ii) Discrete-time signal <math>x(n)</math> for <math>F_s = 450</math> Hz.</li> <li>iii) Value of <math>F_s</math> so that <math>y(t) = A + 15 \cos\left(60\pi t - \frac{\pi}{4}\right)</math>. Where <math>A</math> is a constant. Also, determine the value of <math>A</math>.</li> </ul> | <b>(02)</b> |
| <b>2B.</b> | <p>Given the linear phase FIR filter <math>h(n) = \{1, 4, -8, a, b\}</math>. Find the values of <math>a</math> and <math>b</math> for type-1 system (symmetric and odd-length). Realize the filter with minimum number of multipliers.</p>                                                                                                                                                                                                                                                                                                                                                                                                                             | <b>(03)</b> |
| <b>2C.</b> | <p>Realize the given filter <math>H(z) = \frac{1+z^{-1}}{(1-0.5z^{-1})(1-3z^{-1}+2z^{-2})}</math> using</p> <ul style="list-style-type: none"> <li>i) Direct form-II</li> <li>ii) Parallel form structure</li> </ul>                                                                                                                                                                                                                                                                                                                                                                                                                                                   | <b>(05)</b> |

- 3A.** Consider the z-transform  $H(z) = \frac{1}{(2z^{-1}-1)((1/2)z^{-1}-1)}$ .
- Sketch the poles, zeros and the possible ROCs.
  - For each ROC, determine if the impulse response is stable or unstable.
  - Compute  $h[n]$  considering  $H(z)$  with ROC corresponding to the causal impulse response.
- (03)**

- 3B.** Consider a filter described by the difference equation as below:  
 $y[n] = 0.1 x[n] - 0.9 y[n - 1]$ .  
 Determine the frequency at which  $|H[0]| = 0.707$ . Is this a low-pass or high-pass filter?
- (04)**

- 3C.** Write short note on FIR and IIR all-pass filters. Mention the system function and pole-zero map for both.
- (03)**

- 4A.** Derive the system function and frequency response equations for the DT system characterized by the pole-zero map shown below:



Given: ' $r$ ' is the distance between the location of the pole and the origin in the complex Z-plane. The circle shown in figure is an unit radius circle.

**(03)**

- 4B.** A low-pass filter has the desired frequency response

$$H_d(e^{j\omega}) = \begin{cases} e^{-j4\omega}; & 0 < \omega < \frac{\pi}{2} \\ 0; & \text{elsewhere} \end{cases}$$

Design a suitable filter using frequency sampling technique.

**(03)**

- 4C.** Design a normalized linear phase FIR low pass filter with a cut-off frequency of 1 radian/sample, having a group delay  $\tau = 4$  and maximum attenuation in the stop band using window technique (select an appropriate window to meet the specifications). Draw a suitable structure to implement the same.
- (04)**

- 5A.** List any four differences between all-zero and pole-zero filters.
- (02)**

- 5B.** The transfer function of an analog filter is given by:  $H[s] = \frac{1}{s+2}$ . Design a digital filter  $H[z]$  from  $H[s]$  using impulse invariance technique. Use a sampling frequency of 1 Hz. Also, draw the pole-zero plot.
- (03)**

- 5C.** Design a Butterworth filter using Bilinear transformation technique to meet the following specifications:

$$0.75 \leq |H(e^{j\omega})| \leq 1; \quad 0 \leq \omega \leq 0.25\pi$$

$$|H(e^{j\omega})| \leq 0.23; \quad 0.63\pi \leq \omega \leq \pi$$

Take a sampling frequency of 8 KHz.

**(05)**