

B.E./B.TECH. DEGREE EXAMINATION, DECEMBER 2020

Fifth Semester

EC16502- PRINCIPLES OF DIGITAL SIGNAL PROCESSING

(Regulation 2016)

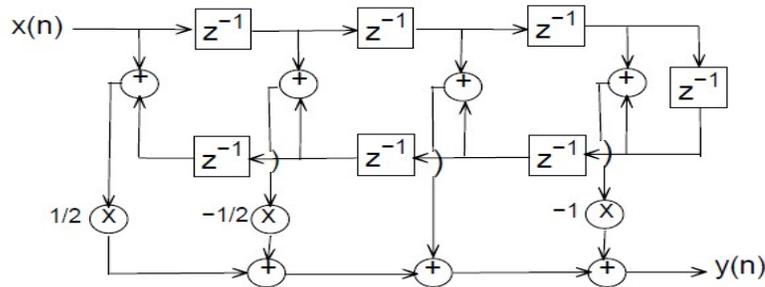
Time: Three hours

Maximum : 80 Marks

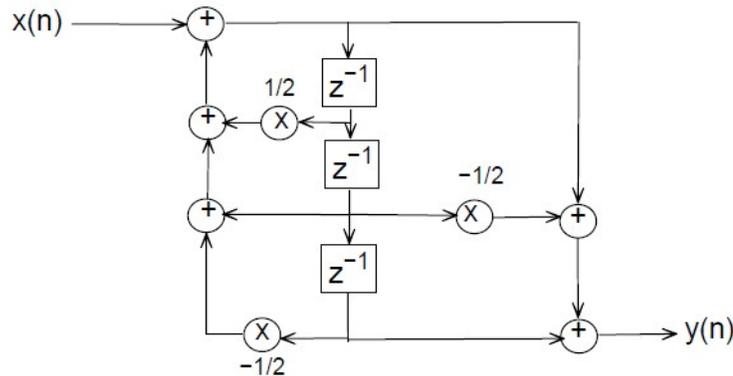
Answer ALL questions

PART A - (8 X 2 = 16 marks)

- Discrete Fourier Transform of a sequence $x(n]$ is given by $X(k) = \{4, (1+2j), j, (1-3j)\}$, hence Discrete Fourier Transform of $x^*(n]$ is:
 - $\{4, (1+3j), -j, (1-2j)\}$
 - $\{4, (1-2j), j, (1+3j)\}$
 - $\{4, (1-3j), j, (1+2j)\}$
 - $\{4, (1+2j), -j, (1-3j)\}$
- Determine the filter coefficients for the given Linear phase realization structure



- $1 -1 2 -2 -2 2 -1 1$
 - $1/2, -1/2, 1, -1$
 - $1/2, -1/2, 1, -1, -1, 1, -1/2, 1/2$
 - $0, 0, 0, 0, -1/2, 1/2, 1, -1$
- For the IIR filter shown below, what is the input-output relation?



- $y(n) - \frac{1}{2}y(n-1) - y(n-2) + \frac{1}{2}y(n-3) = x(n) - \frac{1}{2}x(n-2) + x(n-3)$
 - $y(n) + \frac{1}{2}y(n-1) + y(n-2) - \frac{1}{2}y(n-3) = x(n-1) + \frac{1}{2}x(n-2) - x(n-3)$
 - $y(n) + \frac{1}{2}y(n-1) - y(n-2) + \frac{1}{2}y(n-3) = x(n) - \frac{1}{2}x(n-2) + x(n-3)$
 - $y(n) - \frac{1}{2}y(n-1) + y(n-2) + \frac{1}{2}y(n-3) = x(n) + \frac{1}{2}x(n-2) + x(n-3)$
- The error in the filter output that results from rounding or truncating calculations within the filter is called
 - Coefficient quantization error
 - Adder overflow limit cycle
 - Round off noise
 - Limit cycles

5. Give the significance of anti-imaging filter in sampling rate conversion.
6. Realize the following system using minimum number of multipliers:

$$H(z) = 1 + (3/4)z^{-1} + (17/8)z^{-2} + (3/4)z^{-3} + z^{-4}$$
7. Given the specification $\alpha_p = 1\text{dB}$; $\alpha_s = 30\text{dB}$; $\Omega_p = 200 \text{ rad/sec}$; $\Omega_s = 600 \text{ rad/sec}$. Determine the order of the Butterworth filter.
8. Express the fraction $-7/8$ in sign magnitude, 2's complement and 1's complement.

PART B - (4 X16 = 64 marks)

09. (a) (i) Consider the finite sequence $x[n] = \{1, 2, 2, 1\}$. The 5-point DFT of $x[n]$ is denoted by $X(k)$. Plot the sequence whose DFT is $Y(k) = e^{-j4\pi k/5} X(k)$. (6)
- (ii) Find the output $y[n]$ of a filter whose impulse response $h[n] = \{1, 2\}$ and input signal is $x[n] = \{1, 2, -1, 2, 3, -2, -3, -1, 1, 1, 2, -1\}$ using Overlap add method. (10)

(OR)

- (b) Find the DFT of a sequence $x[n] = \{1, 2, 3, 4, 4, 3, 2, 1\}$ using DIT algorithm. (16)
10. (a) Design a High pass filter using Hamming window to meet the following specifications. (16)
 Cut-off frequency = 250 Hz
 Sampling frequency = 1 KHz
 Filter length = 7
 Also draw its Direct form structure.

(OR)

- (b) (i) Using frequency sampling technique, design a linear phase FIR low pass filter with a cut-off frequency of $0.5\pi \text{ rad/sec}$. Assume order of the filter, $M=9$ (10)
 - (ii) Determine the frequency response of FIR filter described by, (6)
 $y(n) = 0.25x(n) + x(n-1) + 0.25x(n-2)$ and also check the linear phase condition.
11. (a) Design a digital Butterworth filter satisfying the constraint (16)

$$0.707 \leq |H(e^{j\omega})| \leq 1, \quad 0 \leq \omega \leq \pi/2$$

$$|H(e^{j\omega})| \leq 0.2, \quad 3\pi/4 \leq \omega \leq \pi$$
 With $T=1\text{s}$ using Impulse Invariant Technique.

(OR)

- (b) Design a digital Chebyshev filter to satisfy the constraint (16)

$$0.707 \leq |H(e^{j\omega})| \leq 1, \quad 0 \leq \omega \leq 0.2\pi$$

$$|H(e^{j\omega})| \leq 0.1, \quad 0.5\pi \leq \omega \leq \pi$$

Using bilinear transformation and assuming $T = 1$ s. Realize the filter using minimum number of delay units.

12. (a) Two first order filters are connected in cascade whose system functions of the (16)

individual sections are $H_1(z) = \frac{1}{1-0.6z^{-1}}$ and $H_2(z) = \frac{1}{1-0.9z^{-1}}$.

Determine the overall output noise power. Assume $B=8$ bits.

(OR)

- (b) A digital system is characterized by the difference equation $y(n)=0.875y(n-1)+x(n)$. Determine the dead band of the system when $x(0)=0.75$ and $B=4$ (16)

bits.