

7. IIR filters are one of two primary types of digital filters used in Digital Signal Processing applications. Express the property of Infinite Impulse Response(IIR) filter (Recursive/Non recursive)?
(CO3) [Knowledge]
8. Frequency transformation techniques are used to generate High pass filter, Bandpass and bandstop filter from the lowpass filter system function. The transformation technique in which there is one to one mapping from s-domain to z-domain is _____.
(CO3) [Knowledge]
9. Parellel realization of filtering gaining importance for parellel processing of informations. For parellel realization, partial fraction method is followed. For such a structre, The degree of the numerator must be less than the degree of the denominator. Check the statement truthfulness.
(CO3) [Knowledge]
10. Compare to analog filter digital filters are operated with wide range of frequencies. To design a digital filter ,analog filter transfer function is needed (TRUE/FALSE)?
(CO3) [Knowledge]
11.
$$H_d(e^{j\omega}) = \begin{cases} e^{-j2\omega} & (\frac{\pi}{4}) < |\omega| < \pi \\ 0 & otherwise \end{cases}$$
 for the given desired frequency response, Identify the type of filter.
(CO4) [Knowledge]
12. A filter is described by the following difference equation, $y(n)=2x(n)+3x(n-1)+x(n-2)+4x(n-3)+x(n-4)+3x(n-5)+2x(n-6)$. Write the transfer function H(z) for the above filter.
(CO4) [Knowledge]
13. A particular filter operated over a wide range of frequencies and able to pass the frequency less than given cutoff frequency and attenuating the information of higher frequency. Which is functioning as _____ filter.
(CO4) [Knowledge]
14. Linear phase is a property of a filter, where the phase response of the filter is a linear function of frequency. Then related condition of linear phase is _____
(CO4) [Knowledge]
15. Identify the type of filter which is described by the following desired frequency response.
$$H_d(e^{j\omega}) = \begin{cases} e^{(-j\omega)} & |\omega| \leq (\frac{\pi}{4}) \\ 0 & otherwise \end{cases}$$

(CO4) [Knowledge]

PART B

ANSWER ALL THE TWO QUESTIONS

2 X 15 = 30M

16. In signal processing, the Overlap–Save method is used to evaluate the convolution of a very long signal with a finite impulse. Find $y(n)$ for the given $x(n)$ and $h(n)$ by using the Overlap Save method, where $x(n)=\{1,2,-1,2,3,-2,-3,-1,1,1,2,-1\}$ and $h(n)=\{1,2,3\}$.
(CO1) [Comprehension]
17. A digital system should be designed to produce circular convolution of two equal length sequence. But we are having a system which is able to produce Discrete Fourier Transform and its inverse. Determine the circular convoluted value for the two sequences of $x_1(n)=\{1,2,-2,-1\}$ and $x_2(n)=\{-3,-2,1,0\}$ using FFT and IFFT algorithm.
(CO2) [Comprehension]

PART C

ANSWER ALL THE TWO QUESTIONS

2 X 20 = 40

18. a) A Butterworth filter is a type of signal processing filter designed to have a frequency response as flat as possible in the passband. Design an Analog Low Pass Butterworth Filter which has a 2db Passband Attenuation at a frequency of 20 rad/sec and a 10db Stopband Attenuation at 30 rad/sec.
- b) Realisation of a system means obtaining a network corresponding to the equation or transferfunction.
- The Direct Form-I realization is the simplest and most straightforward structure for the realization of any system.
- For the given difference equation $y(n)+2y(n-1)+3y(n-2)=x(n)+5x(n-1)$, Obtain the Direct Form-I realization structure. (CO3) [Application]

19. The window method for digital filter design is fast, convenient, and robust, but generally suboptimal. The window method consists of simply windowing a theoretically ideal FIR by some suitably chosen window function. By using a Rectangular Window $W(n)$, Design FIR filter for a given desired response

$$H_d(\omega) = \begin{cases} e^{-j3\omega} & \text{for } \left(-\frac{\pi}{4}\right) \leq \omega \leq \left(\frac{\pi}{4}\right) \\ 0 & \text{otherwise} \end{cases}$$

Where $W(n)$ is define as

$$W(n) = \begin{cases} 1 & \text{for } 0 \leq n \leq 6 \\ 0 & \text{otherwise} \end{cases}$$

Further derive the Frequency Response $H(e^{j\omega})$ and realize the derived Filter $H(z)$ by using the minimum number of multipliers (ie., Linear Phase Structure).

(CO4) [Application]
