

* Time-Variant Checking:

$$y(t-t_0) = x(t-t_0) \cos(\omega_0(t-t_0))$$

Now let's delay $x(t)$ by $t_0 \rightarrow x(t-t_0)$

$$\therefore F[x(t-t_0)] = x(t-t_0) \cos(\omega_0 t) \neq y(t-t_0)$$

So the system is time-varying.

c) $y(t) = 3x(t-4)$

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* Linearity Checking:

$$\text{- Let } y_1(t) = 3x_1(t-4) \quad , \quad y_2(t) = 3x_2(t-4)$$

$$= F[x_1] \quad \quad \quad = F[x_2]$$

$$\therefore F[x_1] + F[x_2] = 3x_1(t-4) + 3x_2(t-4)$$

$$\text{- } x(t) = a_1 x_1(t) + a_2 x_2(t)$$

$$F[a_1 x_1(t) + a_2 x_2(t)] = 3 [a_1 x_1(t-4) + a_2 x_2(t-4)]$$

$$= 3a_1 x_1(t-4) + 3a_2 x_2(t-4)$$

$$= a_1 F[x_1(t)] + a_2 F[x_2(t)]$$

So the system is linear

* Time-Variant Checking:

$$y(t-t_0) = 3x(t-4-t_0)$$

Now let's delay $x(t)$ by $t_0 \rightarrow x(t-t_0)$

$$\therefore F[x(t-t_0)] = 3x(t-4-t_0) = y(t-t_0)$$

So the system is time-invariant.

As a result, this system is LTI system.

3. Response of LTI Systems:

[A] Impulse Response:

The impulse response $h(t)$ of LTI system is defined to be the response of the system when the input is $\delta(t)$.

output

$$h(t) = F[\delta(t)]$$

where $h(t)$ is the impulse response of the system.

* Causal System: The system whose impulse response vanish for $t < 0$:

$$h(t) = 0 \quad \text{for } t < 0$$

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* This means that the causal system does not response before the excitation is applied.

* The response of LTI system to any input signal $x(t)$ in time domain is given by:

$$y(t) = x(t) * h(t) \quad \text{in time domain}$$

$$\therefore y(t) = \int_{-\infty}^{\infty} x(\tau) h(t-\tau) d\tau = \int_{-\infty}^{\infty} h(\tau) x(t-\tau) d\tau$$

B Frequency Response $H(f)$: (Transfer Function)

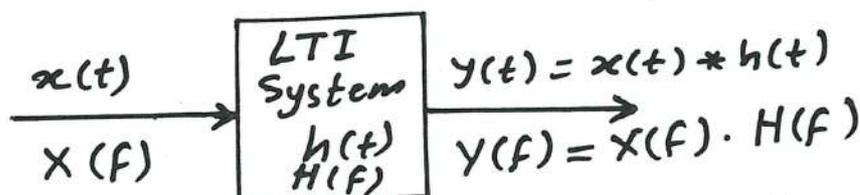
It is the Fourier transform of the impulse response.

$$h(t) \xleftrightarrow{\text{F.T}} H(f) \quad \text{or} \quad h(t) \xleftrightarrow{\text{F.T}} H(\omega)$$

$$y(t) = x(t) * h(t) \quad \text{Take F.T for both sides}$$

$$Y(f) = X(f) \cdot H(f) \quad \{ \text{Freq. Domain} \}$$

$$\therefore \boxed{H(f) = \frac{Y(f)}{X(f)}} \quad \text{OR} \quad H(\omega) = \frac{Y(\omega)}{X(\omega)}$$



- The transfer function $H(f)$ is a complex quantity,

$$H(f) = |H(f)| e^{j\theta_h(f)}$$

$$H^*(f) = H(-f)$$

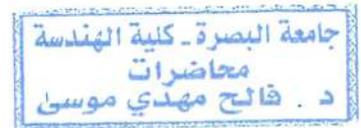
and $|H(f)| = |H(-f)| \rightarrow$ even symmetry

$\theta_h(f) = -\theta_h(-f) \rightarrow$ odd symmetry

$$Y(f) = X(f) H(f)$$

so $|Y(f)| = |X(f)| \cdot |H(f)|$

$$\theta_y(f) = \theta_x(f) + \theta_h(f)$$



* Special Case :

Sinusoidal signal input does not need to be converted to frequency domain to deduce its response in LTI systems.

$$x(t) = A \cos(2\pi f_0 t + \theta_x)$$

$$H(f) = |H(f)| e^{j\theta_h(f)} = |H(f)| \angle \theta_h(f)$$

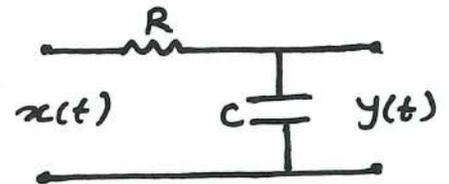
$$\therefore \boxed{y(t) = |H(f_0)| A \cos(2\pi f_0 t + \theta_x + \theta_h(f_0))} \quad \text{For sinusoidal only}$$

Example: For the following RC network, find:

a) The frequency response $H(f)$.

b) The impulse response $h(t)$.

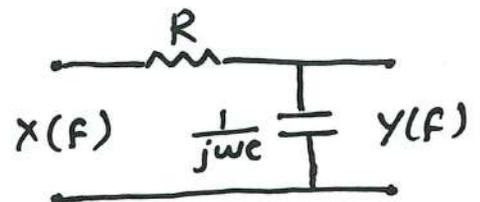
c) Unit-step response.



Sol.

$$a) Y(f) = \frac{1}{R + \frac{1}{j2\pi f C}} X(f)$$

$$Y(f) = H(f) X(f)$$



$$\therefore H(f) = \frac{1}{R + \frac{1}{j2\pi f C}} \Rightarrow H(f) = \frac{1}{1 + j2\pi f RC}$$

$$\text{OR, } H(f) = \frac{\frac{1}{RC}}{\frac{1}{RC} - j2\pi f}$$

$$b) h(t) = F.T^{-1}[H(f)]$$

$$h(t) = F.T^{-1}\left[\frac{1}{RC} \frac{1}{\frac{1}{RC} + j2\pi f}\right] \equiv F.T^{-1}\left[A \frac{1}{a + j2\pi f}\right]$$

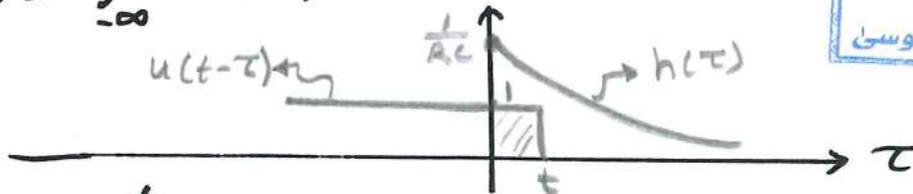
$$h(t) = \frac{1}{RC} e^{-t/RC} u(t)$$

c) Unit step response means $y(t)$ when $x(t) = u(t)$.

The time domain analysis is easier to be used in this problem.

$$y(t) = x(t) * h(t)$$

$$y(t) = \int_{-\infty}^{\infty} h(\tau) u(t-\tau) d\tau$$



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$$y(t) = \int_0^t h(\tau) \times 1 d\tau$$

$$y(t) = \int_0^t \frac{1}{RC} e^{-\frac{\tau}{RC}} d\tau = \frac{-RC}{RC} e^{-\frac{\tau}{RC}} \Big|_0^t$$

$$\therefore y(t) = 1 - e^{-t/RC}$$

4. Distortionless Transmission:

The distortionless channel has an output signal that represents a delayed version of the input signal.

$$y(t) = k x(t - t_0)$$

where,

k - the gain constant. (attenuation if $k < 1$)

t_0 - the time delay.

$$y(t) = x(t) * k \delta(t - t_0) \equiv x(t) * h(t)$$

$$\therefore Y(f) = X(f) \cdot k e^{-j(2\pi t_0)f}$$

$$Y(f) = X(f) \cdot H(f)$$

∴ The impulse response of the distortionless channel is: 8

$$h(t) = k \delta(t - t_0)$$

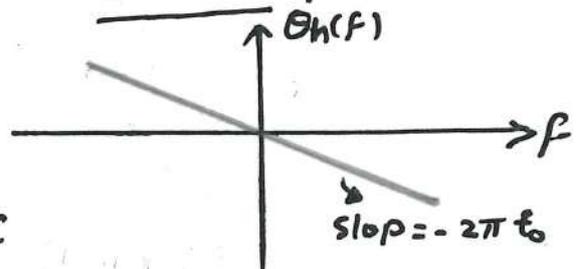
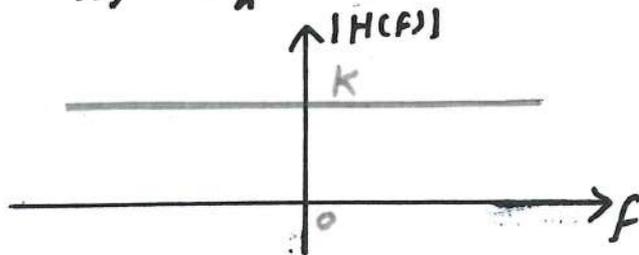
and its frequency response (transfer function) is:

$$H(f) = k e^{-j(2\pi t_0)f} = |H(f)| e^{j\theta_h(f)}$$

In other words, for distortionless channel:

1) $|H(f)| = k$ constant value

2) $\theta_h(f) = -(2\pi t_0)f$ linear phase.



A) Linear Distortion in Linear Systems:

- Amplitude Distortion:

A distortion happens when $|H(f)|$ is not constant within the frequency range of interest. Therefore, the frequency components of the input signal are transmitted with different amount of attenuation.

- Phase Distortion:

A distortion happens when $\theta_h(f)$ is nonlinear within the frequency range of interest. Transmission has different delay for different frequency components

* The signals may undergo both distortions simultaneously.

Group and Phase Delay:

- Group Delay $T_g(f)$: It is the delay that a group of two or more frequency components undergo in passing through a linear system.

$$T_g(f) = \frac{-1}{2\pi} \frac{d\theta_h(f)}{df}$$

where $\theta_h(f)$ is the phase response of the system.

For distortionless system $\theta_h(f) = -2\pi f t_0$

$$\therefore T_g(f) = \frac{-1}{2\pi} \frac{d}{df} (-2\pi f t_0)$$

$$\therefore T_g(f) = \boxed{t_0}$$

- Phase Delay $T_p(f)$: It is the delay of single frequency component.

$$T_p(f) = - \frac{\theta_h(f)}{2\pi f}$$

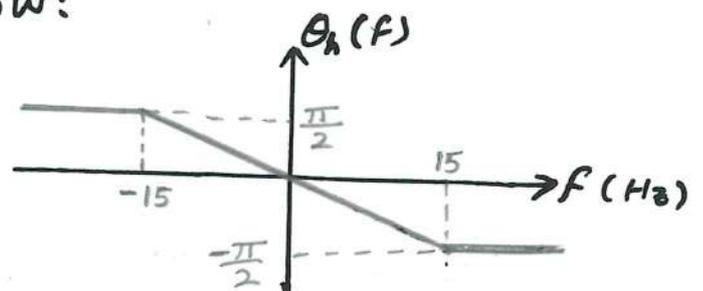
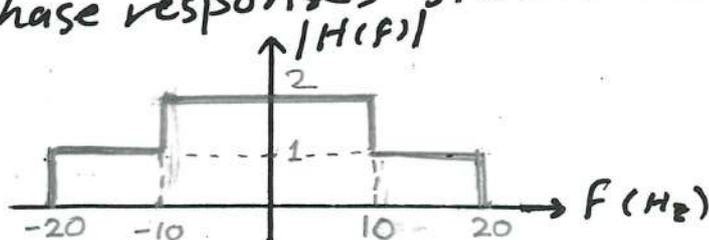
For distortionless system

$$T_p(f) = - \frac{-2\pi f t_0}{2\pi f}$$

$$\therefore T_p(f) = \boxed{t_0}$$

Thus, for distortionless channels
 $T_g = T_d = \text{constant}$.

Example: Consider a system with amplitude and phase responses shown below:



Suppose the following signals are passed through the the above system.

$$a) x_1(t) = \cos(10\pi t) + \cos(12\pi t)$$

$$b) x_2(t) = 3\cos(10\pi t) + 6\cos(26\pi t)$$

$$c) x_3(t) = \cos(26\pi t + 30^\circ) + \sin(34\pi t)$$

Find the response of each signal and the distortion type. Also determine and sketch the group delay and the phase delay.

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Sol.

$$|H(f)| = \begin{cases} 1 & -20 < f < -10 \\ 2 & -10 < f < 10 \\ 1 & 10 < f < 20 \end{cases} \quad \theta_h(f) = \begin{cases} \frac{\pi}{2} & f < -15 \\ -\frac{\pi}{30}f & -15 < f < 15 \\ \frac{\pi}{2} & f > 15 \end{cases}$$

Remember that $y(t) = |H(f_0)| A \cos(2\pi f_0 t + \theta + \theta_h(f_0))$

$$a) x_1(t) = \cos(2\pi(5)t + 0^\circ) + \cos(2\pi(6)t + 0^\circ)$$

$$f_1 = 5 \text{ Hz} \quad f_2 = 6 \text{ Hz}$$

$$\therefore y_1(t) = \underline{2} \times \cos(2\pi(5)t - \frac{\pi}{30}(5)) + \underline{2} \times \cos(2\pi(6)t - \frac{\pi}{30}(6))$$

$$y_1(t) = 2 \cos(2\pi(5)t - \frac{\pi}{6}) + 2 \cos(2\pi(6)t - \frac{\pi}{5})$$

Since both components multiplied by the same gain ($k=2$), and the phase are linear through the both frequencies, then this signal is passed without any distortion (distortionless transmission).

$$b) x_2(t) = 3 \cos(2\pi(5)t + 0^\circ) + 6 \cos(2\pi(13)t + 0^\circ)$$

$$f_1 = 5 \text{ Hz} \quad f_2 = 13 \text{ Hz}$$

$$y_2(t) = \underline{2} \times 3 \cos(2\pi(5)t + \frac{\pi}{30} \times 5) + \underline{1} \times 6 \cos(2\pi(13)t - \frac{\pi}{30} \times 13)$$

$$y_2(t) = 6 \cos(2\pi(5)t - \frac{\pi}{30} \times 5) + 6 \cos(2\pi(13)t - \frac{\pi}{30} \times 13)$$

$$y_2(t) = 6 \cos(2\pi(5)t - \frac{\pi}{6}) + 6 \cos(2\pi(13)t - \frac{13}{30}\pi)$$

The 1st component is multiplied by gain = 2
 The 2nd component is multiplied by gain = 1
 but both components located at the same linear region
 of the phase response, so
 $x_2(t)$ undergoes amplitude distortion only.

$$c) x_3(t) = \cos(2\pi(13)t + 30^\circ) + \cos(2\pi(17)t - 90^\circ)$$

$F_1 = 13 \text{ Hz} \qquad F_2 = 17 \text{ Hz}$

$$y_3(t) = \underset{=}{1} * \cos(2\pi(13)t + 30^\circ - \frac{\pi}{30} * 13) + \underset{=}{1} * \cos(2\pi(17)t - 90^\circ - \frac{\pi}{2})$$

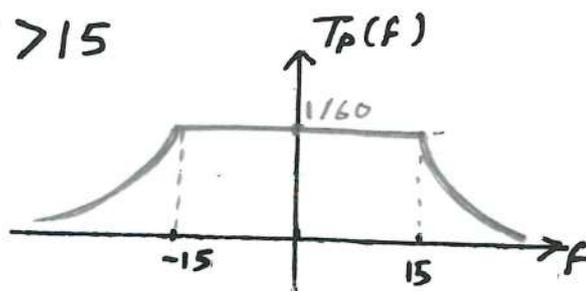
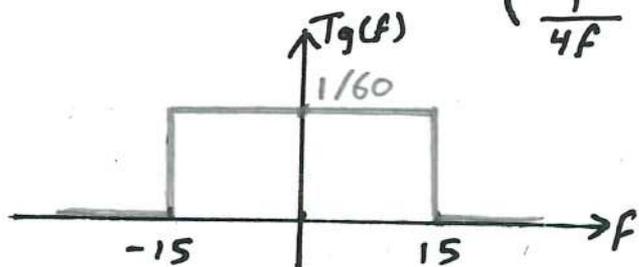
$$y_3(t) = \cos(2\pi(13)t - 48^\circ) + \cos(2\pi(17)t - 180^\circ)$$

Both components ^{are} multiplied by gain = 1
 but they do not located within the same linear phase response
 $x_3(t)$ undergoes phase distortion only.

$$T_g(f) = \frac{-1}{2\pi} \cdot \frac{d\theta_h(f)}{df} = \begin{cases} 0 & f < -15 \\ \frac{1}{60} & -15 < f < 15 \\ 0 & f > 15 \end{cases}$$

$$T_d(f) = \frac{\theta_h(f)}{-2\pi f} = \begin{cases} \frac{-1}{4f} & f < -15 \\ \frac{1}{60} & -15 < f < 15 \\ \frac{1}{4f} & f > 15 \end{cases}$$

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[B] Nonlinear Distortion:

This kind of distortion is introduced by nonlinear systems. Suppose an input signal $x(t)$ given by:

$$x(t) = A_1 \cos(2\pi f_1 t) + A_2 \cos(2\pi f_2 t)$$

is entered to a nonlinear system whose input-output characteristic is given by:

$$y(t) = a_1 x(t) + a_2 x^2(t)$$

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where a_1 & a_2 are constants. Therefore,

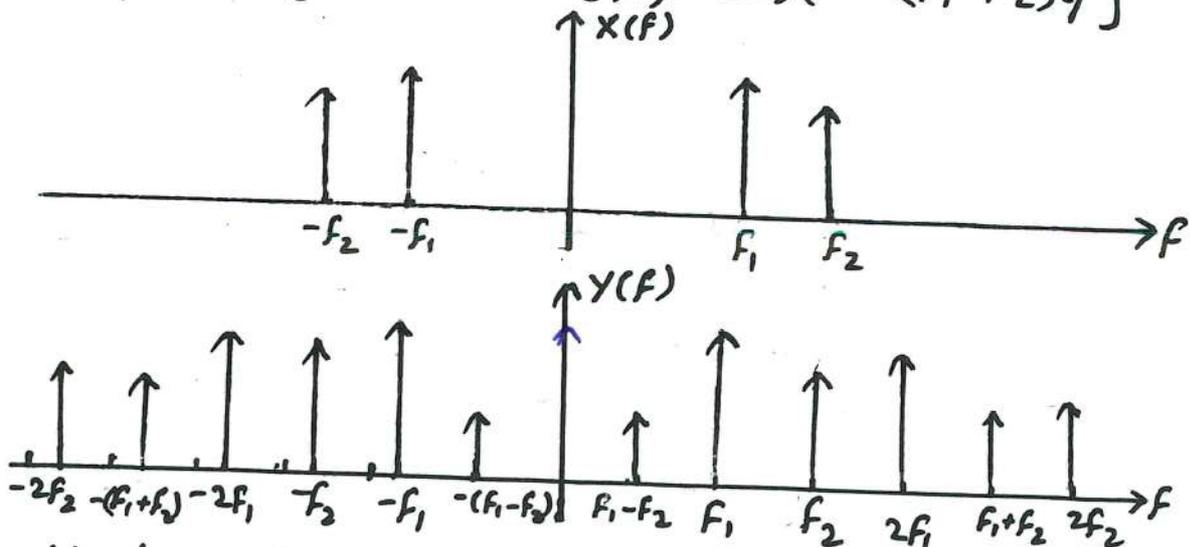
$$y(t) = a_1 [A_1 \cos(2\pi f_1 t) + A_2 \cos(2\pi f_2 t)] + a_2 [A_1 \cos(2\pi f_1 t) + A_2 \cos(2\pi f_2 t)]^2$$

$$y(t) = a_1 [A_1 \cos(2\pi f_1 t) + A_2 \cos(2\pi f_2 t)] + a_2 [A_1^2 \cos^2(2\pi f_1 t) + 2A_1 A_2 \cos(2\pi f_1 t) \cos(2\pi f_2 t) + A_2^2 \cos^2(2\pi f_2 t)]$$

but $\cos^2 \theta = \frac{1}{2} [1 + \cos(2\theta)]$

$$\cos \theta_1 \cos \theta_2 = \frac{1}{2} [\cos(\theta_1 - \theta_2) + \cos(\theta_1 + \theta_2)]$$

$$\begin{aligned} \therefore y(t) &= a_1 [A_1 \cos(2\pi f_1 t) + A_2 \cos(2\pi f_2 t)] + \\ &+ \frac{1}{2} a_2 (A_1^2 + A_2^2) + \\ &+ \frac{1}{2} a_2 [A_1^2 \cos(2\pi (2f_1) t) + A_2^2 \cos(2\pi (2f_2) t)] + \\ &+ a_2 A_1 A_2 [\cos(2\pi (f_1 + f_2) t) + \cos(2\pi (f_1 - f_2) t)] \end{aligned}$$



Note that nonlinear systems has produced frequencies in the output other than the input frequencies.

5. Filters:

The filter is a system that passes some frequency components and suppresses others.

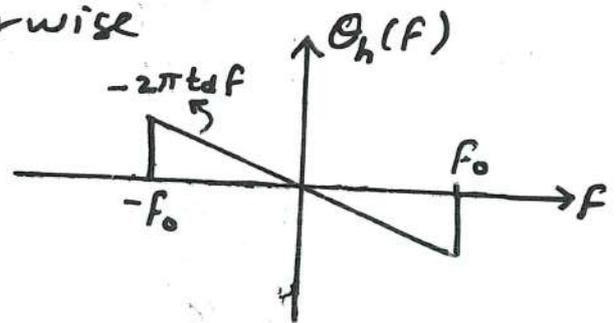
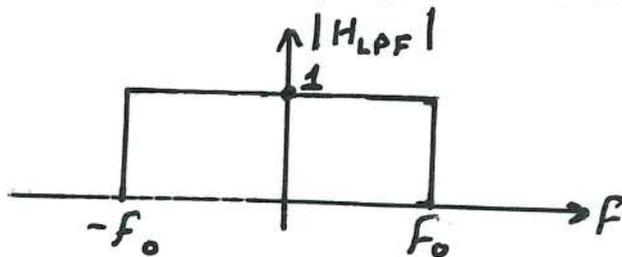
[A] Ideal Filters:

Filters have the characteristics of distortionless transmission over specified frequency band and zero response over all other bands.

- Ideal Low-Pass Filter (LPF):

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$$H_{LPF}(f) = \begin{cases} e^{-j2\pi t_d f} & |f| \leq f_0 \\ 0 & \text{otherwise} \end{cases}$$



It can be re-written as:

$$H_{LPF}(f) = \text{TT}\left(\frac{f}{2f_0}\right) e^{-j2\pi t_d f}$$

Ideal LPF can perfectly pass frequencies below f_0 and perfectly eliminate frequencies above f_0 .

$$h_{LPF}(t) = \mathcal{F.T}^{-1}[H_{LPF}(f)]$$

$$h_{LPF}(t) = 2f_0 \text{Sinc}[2f_0(t - t_d)]$$

