

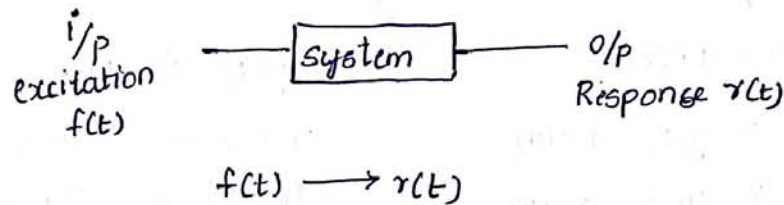
UNIT-3

SIGNAL TRANSMISSION THROUGH LINEAR SYSTEMS

System: A system is defined as set of rules that associates an o/p time function to every i/p time function.

(or)

A system is an interconnection of elements which produces expected o/p for available i/p.



→ System is an mathematical operator which maps i/p into o/p

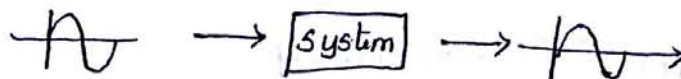
classification of system.



- ↓
1. Static & Dynamic systems
 2. Linear & Non-Linear
 3. Time invariant & Time variant
 4. Linear TIV & LTIV
 5. Stable system
 6. casual & non-causal systems.

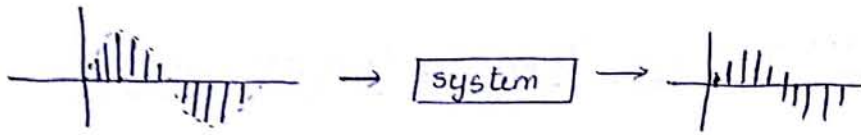
(i) continuous time systems

→ A continuous time system operates on a continuous time i/p signal to produce a continuous time o/p signal



(ii) Discrete time systems:

A discrete time system operates on a discrete time i/p signal to produce a discrete time o/p signal.



classifications:

(i) static and Dynamic Systems:

→ A static system or system is said to be static if its o/p at any instant depends only on present values of i/p.

Ex: $y(t) = ax(t)$

at $t=0$ $y(0) = ax(0)$

at $t=1$ $y(1) = ax(1)$

(ii) $y(t) = a^t x(t)$

at $t=0$ $y(0) = a^0 x(0)$

at $t=1$ $y(1) = a^1 x(1)$

→ A system is said to be dynamic if its o/p depends on present & past values of i/p.

Ex: $y(t) = x(t-1) + x(t-2) + x(t)$

at $t=2$

$$y(2) = x(2-1) + x(2-2) + x(2) = x(1) + x(0) + x(2)$$

$\begin{array}{ccc} \swarrow & & \downarrow \\ & \text{past} & \text{present} \end{array}$

(ii) Linear and Non Linear Systems:

→ A system is said to be linear if it satisfies the superposition principle.

→ It states that the response of the system to a weighted sum of signals be equal to the corresponding weighted sum of o/p's of the system to each of the individual i/p signal.

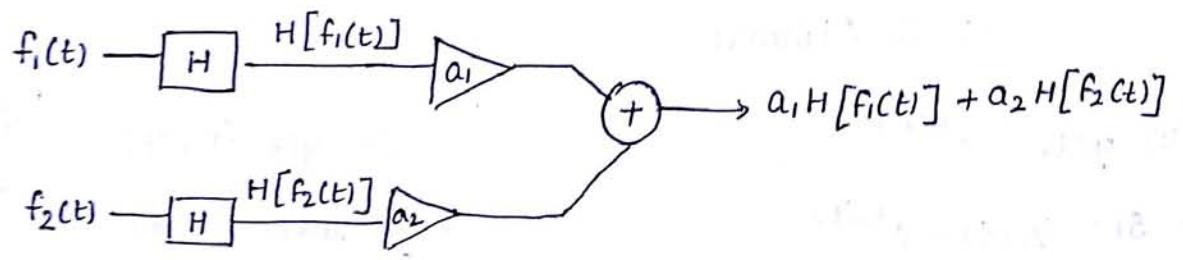
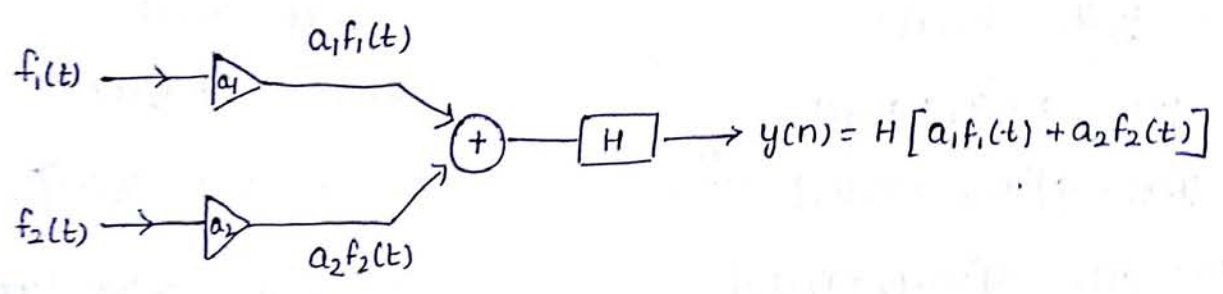
$$H[a_1 f_1(t) + a_2 f_2(t)] = a_1 H[f_1(t)] + a_2 H[f_2(t)]$$

where a_1, a_2 are weighted constants.

$$a_1 f_1(t) \xrightarrow{\text{Response}} a_1 H[f_1(t)]$$

$$a_2 f_2(t) \xrightarrow{\text{Response}} a_2 H[f_2(t)]$$

$$H[a_1 f_1(t) + a_2 f_2(t)] \rightarrow a_1 H[f_1(t)] + a_2 H[f_2(t)]$$



Block diagram.

→ Any system which does not obey the above principle is called as non-linear systems.

check for Linearity:

Procedure:

1. Apply different i/p's separately and get the o/p.
2. Apply different i/p's simultaneously and get the output.
3. If both outputs are same it is linear otherwise non-linear.

Ex:

(i) $y(t) = 4 \sin t x(t)$

Step 1: $y_1(t) = 4 \sin t x_1(t)$

$y_2(t) = 4 \sin t x_2(t)$

$y_1(t) + y_2(t) = 4 \sin t [x_1(t) + x_2(t)]$

Step 2: $y(t) = 4 \sin t [x_1(t) + x_2(t)]$

$S_1 = S_2$

(2) $y(t) = ax(t)$

sol $S_1: y_1(t) = ax_1(t)$

$y_2(t) = ax_2(t)$

$y(t) = ax_1(t) + ax_2(t)$

$y(t) = a[x_1(t) + x_2(t)]$

$S_2: y(t) = a[x_1(t) + x_2(t)]$

$S_1 = S_2$ (Linear)

(4) $y(t) = e^{x(t)}$

sol $S_1: y_1(t) = e^{x_1(t)}$

$y_2(t) = e^{x_2(t)}$

$y(t) = e^{x_1(t)} + e^{x_2(t)}$

$y(t) = e^{[x_1(t) + x_2(t)]}$

$S_2: e^{x_1(t)} + e^{x_2(t)}$

$S_1 \neq S_2$ (Non-Linear)

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

sol $S_1: y_1(t) = x_1(t-t_0)$

$y_2(t) = x_2(t-t_0)$

$y(t) = x_1(t-t_0) + x_2(t-t_0)$

$S_2: y(t) = x_1(t-t_0) + x_2(t-t_0)$

$S_1 = S_2$ (Linear)

(8) $y(t) = x(t+1)e^{-t}$

sol $S_1: y_1(t) = x_1(t+1)e^{-t}; y_2(t) = x_2(t+1)e^{-t}$

$y(t) = e^{-t}[x_1(t+1) + x_2(t+1)]$

$S_2: y(t) = e^{-t}[x_1(t+1) + x_2(t+1)]$

$S_1 = S_2$ (Linear)

(9) $y(t) = 4x(t) + 2 \frac{dx(t)}{dt} \rightarrow$ linear.

(3) $y(t) = x^y(t)$

sol $S_1: y_1(t) = x_1^y(t)$

$y_2(t) = x_2^y(t)$

$y(t) = x_1^y(t) + x_2^y(t)$

$S_2: y(t) = [x_1(t) + x_2(t)]^y \rightarrow$

$S_1 \neq S_2$ (Non-Linear)

(5) $y(t) = tx(t)$

sol $y_1(t) = tx_1(t)$

$S_1: y_2(t) = tx_2(t)$

$y(t) = t[x_1(t) + x_2(t)]$

$S_2: y(t) = t[x_1(t) + x_2(t)]$

$S_1 = S_2$ (Linear)

(11) $y(t) = x(t^y)$

sol $S_1: y_1(t) = x_1(t^y)$

$y_2(t) = x_2(t^y)$

$y(t) = x_1(t^y) + x_2(t^y)$

$S_2: y(t) = [x_1(t) + x_2(t)]^y$

$S_1 \neq S_2$
(Linear)

(7) $y(t) = 3x(t+3)$

sol $S_1: y_1(t) = 3x_1(t+3)$

$y_2(t) = 3x_2(t+3)$

$y(t) = 3[x_1(t+3) + x_2(t+3)]$

$S_2: 3[x_1(t+3) + x_2(t+3)]$

$S_1 = S_2$ (Linear)

(8) $y(t) = Ax(t) + B$

sol $y_1(t) = Ax_1(t) + B$

$y_2(t) = Ax_2(t) + B$

$y(t) = A[x_1(t) + x_2(t)] + 2B$

$y(t) = A[x_1(t) + x_2(t)] + B$

$S_1 \neq S_2$ (Non-linear)

(9) $y(t) = \cos[x(t)]$

sol $S_1: y_1(t) = \cos[x_1(t)]; y_2(t) = \cos[x_2(t)]$

$y(t) = \cos[x_1(t)] + \cos[x_2(t)]$

$S_2: \cos[x_1(t) + x_2(t)]$

$S_1 \neq S_2$ (Non-Linear)

(10) $y(t) = k\Delta x(t)$ where $\Delta x(t) = [x(t+1) - x(t)]$

sol $S_1: y_1(t) = k[x_1(t+1) - x_1(t)]; y_2(t) = k\Delta x_2(t)$

$y(t) = [x_1(t+1) - x_1(t) + x_2(t+1) - x_2(t)] \cdot k$

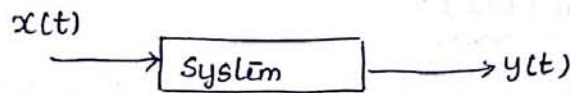
$S_2: y(t) = k[x_1(t+1) - x_1(t) + x_2(t+1) - x_2(t)]$

$S_1 = S_2$ (Linear)

Time Variant And Time Invariant Systems

→ A system is said to be time invariant if the system does not depend on time i.e. system delay is not function of time.

Ex:



$$x(t) \longrightarrow y(t)$$

$$x(t-t_0) \longrightarrow y(t-t_0)$$

→ A time shift t_0 in the input results in the same amount of time shift in the o/p but the waveshape does not change.

i.e. the i/p and/ to o/p characteristics does not change with time.

→ Any system which does not obey the above principle is called as time varying system.

→ An electrical system is said to be time invariant if its component values (R, L, C) does not change with time.

Check for Time Invariant:

1. Shift the i/p only and get the o/p.
2. Shift the entire system and get the o/p.
3. If both steps are identical for o/p then it is time invariant system.

Ex:

(1) $y(t) = 4x(t)$

sol $S_1: y(t) = 4x(t-1)$
 $S_2: y(t-1) = 4x(t-1)$ } $\rightarrow S_1 = S_2$
 (TIV)

(2) $y(t) = 4t x(t)$

sol $S_1: y(t) = 4t x(t-1)$
 $S_2: y(t-1) = 4(t-1) x(t-1)$ } $\rightarrow S_1 \neq S_2$
 (TV)

(3) $y(t) = ax(t)$

sol $S_1: y(t) = ax(t-1)$
 $S_2: y(t-1) = ax(t-1)$ } $\rightarrow S_1 = S_2$
 (TIV)

(4) $y(t) = ax(t) + b$

sol $S_1: y(t) = ax(t-1) + b$
 $S_2: y(t-1) = ax(t-1) + b$ } $S_1 = S_2$
 (TW)

(5) $y(t) = 5t [x(t)]^2$

sol $S_1: y(t) = 5t [x(t-1)]^2$
 $S_2: y(t-1) = 5(t-1) [x(t-1)]^2$ } $S_1 \neq S_2$
 (TV)

(6) $y(t) = x(t+1)e^{-t}$

sol $S_1: y(t) = x(t+1-1)e^{-t} = x(t)e^{-t}$
 $S_2: y(t-1) = x(t+1-1)e^{-(t-1)}$
 $= x(t) \cdot e^{-t} \cdot e^1 \rightarrow \text{constant}$
 $S_1 = S_2$ (TIV)

$$\textcircled{7} \quad y(t) = x(t+3)$$

$$\begin{aligned} \text{Sol } S_1: y(t) &= x(t+3-1) \\ S_2: y(t-1) &= x(t+3-1) \\ &= x(t+2) \end{aligned} \quad \left. \vphantom{\begin{aligned} S_1 \\ S_2 \end{aligned}} \right\} S_1 = S_2 \text{ (TIV)}$$

$$\textcircled{8} \quad y(t) = x^2(t)$$

$$\begin{aligned} \text{Sol } S_1: y(t) &= x^2(t-1) \\ S_2: y(t-1) &= x^2(t-1) \\ \therefore S_1 &= S_2 \text{ (TIV)} \end{aligned}$$

$$\textcircled{9} \quad y(t) = e^{x(t)}$$

$$\begin{aligned} \text{Sol } S_1: y(t) &= e^{x(t+1)} \\ S_2: y(t-1) &= e^{x(t-1)} \\ S_1 &= S_2 \text{ (TIV)} \end{aligned}$$

Linear Time Invariant System (LTI):

→ Any system which obeys the linearity and time invariant property is called as LTI system.

Linear Time Variant System (LTV):

→ Any system which obeys the linearity and does not obey time invariant property is called LTV system.

Ex: $y(t) = ax(t)$

Linearity: $y_1(t) = ax_1(t)$; $y_2(t) = ax_2(t)$

$$y(t) = ax_1(t) + ax_2(t)$$

$$y(t) = a[x_1(t) + x_2(t)]$$

$$S_2: y(t) = a[x_1(t) + x_2(t)]$$

$$\therefore S_1 = S_2$$

T.I: $y(t) = ax(t)$

$$S_1: y(t) = ax(t-1)$$

$$S_2: y(t-1) = ax(t-1) \quad \left. \vphantom{S_1, S_2} \right\} S_1 = S_2 \text{ (TIV)}$$

∴ It is a linear time invariant system (LTI)

|| 4 :

$$(2) \quad y(t) = tx(t) \rightarrow \text{NLTI}$$

$$(3) \quad y(t) = ax(t) + b \rightarrow \text{NLTI}$$

$$(4) \quad y(t) = ax^2(t) \rightarrow \text{NLTI}$$

$$(5) \quad y(t) = e^{x(t)} \rightarrow \text{NLTI}$$

$$(6) \quad y(t) = x(t-t_0) \rightarrow \text{LTI}$$

Stable System:

→ System is absolutely integrable

$$\int_{-\infty}^{\infty} |f(t)| dt < \infty$$

Causal And Non Causal Systems:

→ A system is said to be causal if o/p $y(t_0)$ depends only on the values of i/p $x(t)$ at $t < t_0$ { present, i/p, past i/p, past o/p's } $\begin{cases} x(t) = 0, \text{ for } t < 0 \\ \text{noncausal } t \leq 0, \text{ or } t \leq 0 \text{ and } t > 0 \\ x(t) \neq 0 \text{ for } t < 0 \end{cases}$

Ex: $y(t) = 4x(t-1)$

$$y(2) = 4x(2-1) \Rightarrow 4x(1)$$

$$y(t) = 4x(t-1) + x(t)$$

$$y(2) = 4x(1) + x(2)$$

→ A system is said to be non-causal if the o/p depends on future values of i/p i.e. future i/p's & o/p's.

Ex: $y(t) = 4x(t+1)$

$$y(2) = 4x(3)$$

Examples whether it is causal & Non causal:

(1) $y(t) = k[x(t+1) - x(t)]$

$$y(0) = k[x(1) - x(0)] \rightarrow \text{Noncausal}$$

(2) $y(t) = 3x(t+3)$

$$y(0) = 3x(3) \rightarrow \text{Non causal}$$

(3) $y(t) = (t+3)x(t-3)$

$$y(0) = (0+3)x(0-3) \\ = 3x(-3) \rightarrow \text{causal}$$

(6) $y(t) = x(2t) \rightarrow \text{Noncausal}$

(7) $y(t) = x(t) - x(t-1) \rightarrow \text{causal}$

(8) $y(t) = x(t) + \int_0^t x(\lambda) d\lambda \\ = x(t) + z(\lambda) \Big|_0^t \Rightarrow \text{Causal.}$
At $t=0, t=1, t=2$

(4) $y(t) = x(t) + 3x(t+4)$

when $t=0, y(0) = x(0) + 3x(4)$

when $t=1, y(1) = x(1) + 3x(5)$

So here response at $t=0, y(0)$

depends on the present i/p & future i/p

here system is noncausal.

(5) $y(t) = x(t^2)$

$t=-1, y(-1) = x(1) \rightarrow \text{future}$

$t=0, y(0) = x(0) \rightarrow \text{present}$

$t=1, y(1) = x(1) \rightarrow \text{present}$

$t=2, y(2) = x(4) \rightarrow \text{future}$

Noncausal.

Except at $t=0, t=1$, the response of any value of t depends on future i/p.

Impulse Response:

The response of a system for an impulse i/p is called a impulse response of the system and it is denoted by $h(t)$

$$\delta(t) \longrightarrow \boxed{\text{system}} \longrightarrow h(t)$$

$$\delta(t) \rightarrow h(t)$$

→ Every system is characterised by its impulse response.

$$\begin{array}{ccc} \text{i/p} & \longrightarrow & \boxed{h(t)} \longrightarrow \text{o/p} \\ f(t) & & r(t) \end{array}$$

Response of a System for an arbitrary i/p:

Response of $\delta(t) \rightarrow h(t)$

$$\delta(t-t_0) \rightarrow h(t-t_0)$$

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

The response of a system for a given i/p $f(t)$ is determined by using superposition principle.

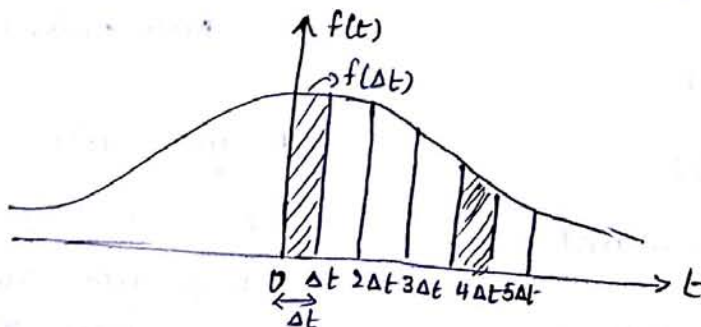
step 1: Resolve the i/p function in terms of impulse functions.

step 2: Determine individually the response of LTI system for impulse function.

step 3: Find the sum of individual responses which will become the overall response $r(t)$.

Representation of a function $f(t)$ in terms of an impulse function:

Here the function $f(t)$ is a impulse train function.



$$\text{area } f(5\Delta t) \times \Delta t$$

$$f(t) = \sum_{n=-\infty}^{\infty} f(n\Delta t) \Delta t \cdot \delta(t-n\Delta t)$$

$$f(t) = \lim_{\Delta t \rightarrow 0} \sum_{n=-\infty}^{\infty} f(n\Delta t) \Delta t \delta(t-n\Delta t)$$

The rectangular of width Δt & height $f(n\Delta t)$ and area under the rectangles is $\Delta t \cdot f(n\Delta t)$ and this n^{th} element approached a delta function of strength $f(n\Delta t) \Delta t$ located at $t=n\Delta t$. and this delta function is represented as $f(n\Delta t) \Delta t \delta(t-n\Delta t)$

$$f(t) = \lim_{\Delta t \rightarrow 0} \sum_{n=-\infty}^{\infty} f(n\Delta t) \cdot \Delta t \cdot \delta(t-n\Delta t)$$

As $\Delta t \rightarrow 0$, the n^{th} element may be considered.

2) Determination of $r(t)$ for the input $f(t)$:

Let $h(t)$ be the impulse response of the system.

$$\delta(t) \rightarrow \boxed{\text{system}} \rightarrow h(t)$$

$$\text{then } \delta(t) \rightarrow h(t)$$

$$\delta(t-n\Delta t) \rightarrow h(t-n\Delta t)$$

$$f(n\Delta t) \delta(t-n\Delta t) \rightarrow f(n\Delta t) \cdot h(t-n\Delta t)$$

$$f(n\Delta t) \cdot \Delta t \delta(t-n\Delta t) \rightarrow f(n\Delta t) \cdot \Delta t h(t-n\Delta t)$$

$$\lim_{\Delta t \rightarrow 0} \sum_{n=-\infty}^{\infty} f(n\Delta t) \cdot \Delta t \delta(t-n\Delta t) \rightarrow \lim_{\Delta t \rightarrow 0} \sum_{n=-\infty}^{\infty} f(n\Delta t) \cdot \Delta t h(t-n\Delta t)$$

$$f(t) \rightarrow \boxed{\text{system}} \rightarrow r(t)$$

$$r(t) = \lim_{\Delta t \rightarrow 0} \sum_{n=-\infty}^{\infty} f(n\Delta t) \cdot \Delta t h(t-n\Delta t)$$

$\Delta t \rightarrow 0$ means summation becomes integration.

$$r(t) = \int_{-\infty}^{\infty} f(\tau) \cdot h(t-\tau) d\tau$$

$$r(t) = f(t) \otimes h(t)$$

$$f(t) \rightarrow \boxed{h(t)} \rightarrow r(t) \rightarrow f(t) \otimes h(t)$$

i.e if the response of a system is known for www.jntuworldupdates.org then response to any other function $f(t)$ can be obtained from the above eqn.
 → An unit impulse function is called as a Test function and it is used to characterise a system.

$$r(t) = f(t) \otimes h(t)$$

In frequency domain

$$r(t) \xrightarrow{FT} R(\omega)$$

$$f(t) \xrightarrow{FT} F(\omega)$$

$$h(t) \xrightarrow{FT} H(\omega)$$

Using convolution property

$$f(t) \otimes h(t) = F(\omega) \odot H(\omega)$$

$$R(\omega) = F(\omega) \cdot H(\omega)$$

$$H(\omega) = \frac{R(\omega)}{F(\omega)}$$

- When $F(\omega) = 1$; i.e i/p is unit impulse $H(\omega) = R(\omega)$
- Transfer function $H(\omega)$ of a system is defined as the transform of the response of a system where the i/p is unit impulse function.

$$H(\omega) = |H(\omega)| e^{j\theta(\omega)}$$

↳ phase response of the system.

↳ Amplitude response of the system.

$$\begin{aligned} \ln[H(\omega)] &= \ln[|H(\omega)| + j\theta(\omega)] \\ &= L(\omega) + j\theta(\omega) \end{aligned}$$

Gain of the system

phase shift introduced by system.

Note: An impulse function contains all frequencies in equal amount so we can use it as a test function.

$$H(\omega) = \frac{R(\omega)}{F(\omega)} \rightarrow \text{Transfer fn of LTI system.}$$

↙

$$F.T [h(t)] \quad h(t) = I.F.T [H(\omega)]$$

FILTER CHARACTERISTICS OF LINEAR SYSTEMS :IDEAL LOW PASS FILTERS :

- It transmits all the signals below certain frequency 'B' Hz without any distortion.
- The range of frequencies from 0 Hz to 'B' Hz is called passband of lowpass filter.
- The frequency 'B' Hz is called cut-off frequency of the ideal lowpass filter.
- The transfer function of ideal low pass filter can be written as

$$H(f) = k e^{-j2\pi f t_0} ; -B \leq f < B$$

$$= 0 ; |f| > B$$

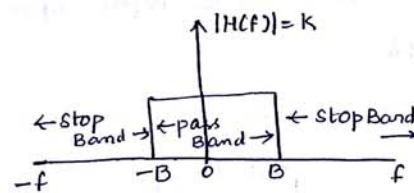
k = amplitude is assumed to be unity.

- By k=1 in above eqn

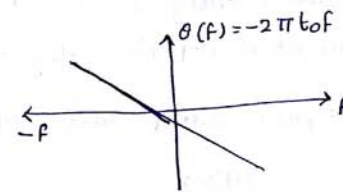
$$H(f) = e^{-j2\pi f t_0} ; -B \leq f < B$$

$$= 0 ; |f| > B$$

- By inverse fourier transform, h(t) can be obtained for ideal LPF



a) magnitude response.



b) phase response.

$$h(t) = \int_{-B}^B e^{-j2\pi f t_0} \cdot e^{j2\pi f t} df$$

$$= \int_{-B}^B [e^{j2\pi f (t-t_0)}] df = \frac{1}{j2\pi(t-t_0)} [e^{j2\pi f (t-t_0)}]_{-B}^B$$

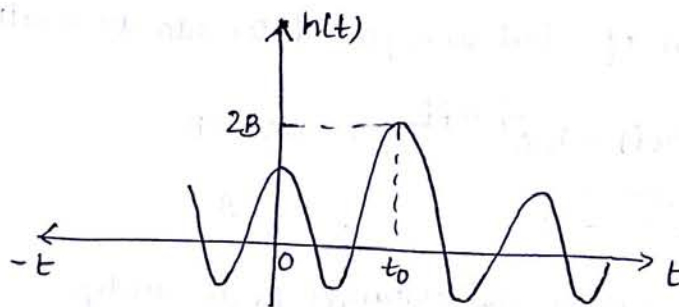
$$= \frac{1}{j2\pi(t-t_0)} [e^{j2\pi B(t-t_0)} - e^{-j2\pi B(t-t_0)}]$$

$$= \frac{1}{\pi(t-t_0)} \left[\frac{e^{j2\pi B(t-t_0)} - e^{-j2\pi B(t-t_0)}}{2j} \right]$$

$$= \frac{1}{\pi(t-t_0)} \sin [2\pi B(t-t_0)]$$

$$h(t) = 2B \left(\frac{\sin [2\pi B(t-t_0)]}{2\pi B(t-t_0)} \right) = 2B \text{sinc} [2B(t-t_0)]$$

Response



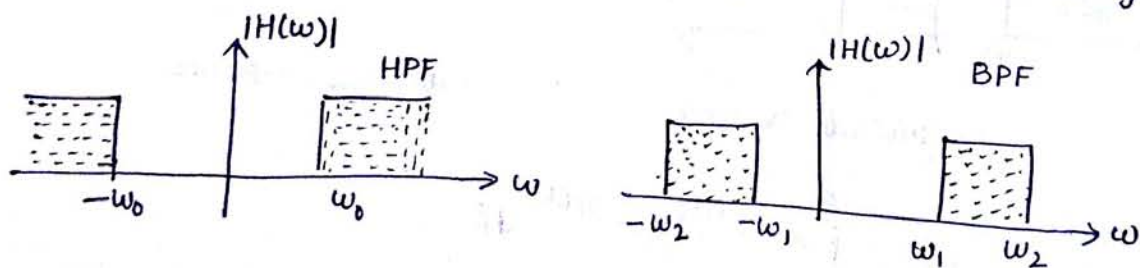
→ Figure shows that impulse response exists for negative values of 't'. But actually unit impulse is applied at $t=0$ always.

→ Practically it is impossible to implement such a system.

OTHER IDEAL FILTERS SUCH AS HPF, BPF etc.,

→ In realizability of ideal LPF its response begins before input is applied and hence it is not physically realizable.

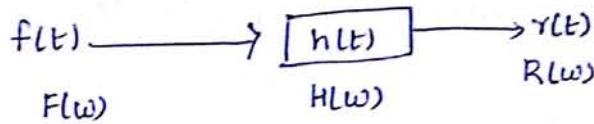
→ ^{ly} HPF, BPF ideal have frequency response as shown in figure



→ These have sharp transition in frequency response.

→ All ideal filters are physically not realizable since their impulse response is non-causal.

INTRODUCTION FOR FILTER CHARACTERISTICS:

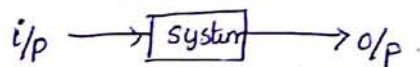


$$R(\omega) = F(\omega) \cdot H(\omega)$$

- The spectrum of o/p is $F(\omega) \cdot H(\omega)$ i.e. the system acts as a kind of filter to various frequency components.
- Some frequency components are boosted in strength and some are attenuated and some remain unaffected.
- \forall each freq. component undergoes a different amount of phase shift i.e. the modification is carried out according to $H(\omega)$.
↳ acts as waiting fn for two different frequencies.

DISTORTIONLESS TRANSMISSION THROUGH SYSTEM:

→ It means output signal is an exact replica of the i/p signal.



→ The difference between i/p and o/p of such system is that

1. Amplitude of the o/p signal may increase or decrease by some factor w.r. to i/p.
2. The o/p sgl may be delayed in time w.r. to i/p sgl because of system delay.

→ o/p sgl $y(t)$ can be written in terms of i/p $x(t)$ as

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

\downarrow constant Represents change in amplitude \rightarrow time delay in transmission of signal through a system.

By taking fourier transform

$$Y(f) = F[y(t)] = F\{k x(t - t_0)\}$$

From time shifting property of FT

$$Y(f) = k X(f) e^{-j2\pi f t_0}$$

$$\text{Transfer fn } H(f) = \frac{Y(f)}{X(f)}$$

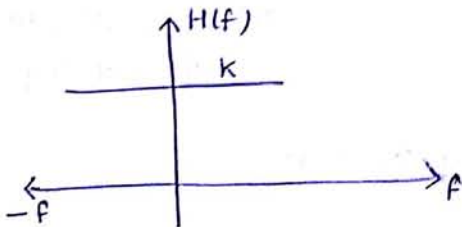
$$\therefore H(f) = \frac{Y(f)}{X(f)} = k \cdot e^{-j2\pi f t_0}$$

Magnitude of transfer fn
independent of frequency.

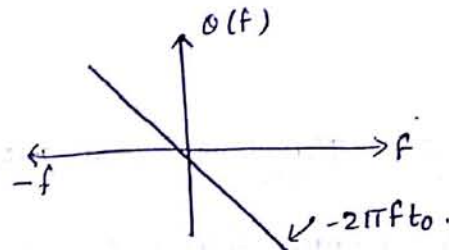
→ Transfer function has constant amplitude at all frequencies. The phase shift is

$$\begin{aligned} \theta(f) &= -2\pi f t_0 \\ &= (-2\pi t_0) f \end{aligned}$$

→ phase shift is linearly proportional to frequency.



(a) Amplitude Spectrum



(b) phase spectrum passing through origin

→ By considering simple example

Let there be signal in time domain as

$$x(t) = \cos(2\pi f t)$$

Let the o/p sgl be same in amplitude but shifted in time by t_0 sec.

$$y(t) = \cos[2\pi f (t - t_0)]$$

$$\therefore y(t) = \cos(2\pi f t - 2\pi f t_0) = \cos(2\pi f t - \theta(f))$$

∴ phase shift of $y(t)$ is

$$\theta(f) = -2\pi f t_0$$

which is proportional to frequency 'f'.

Two types :

- (i) Amplitude distortion
- (ii) phase distortion.

AMPLITUDE DISTORTION

→ This distortion occurs when $|H(\omega)|$ is not constant over frequency band of interest and the frequency components present in $1/p$ sgl are transmitted with different gain and attenuation.

PHASE DISTORTION:

→ This distortion occurs when phase of $H(\omega)$ is not linearly changing with time and different frequency components in $1/p$ are subjected to different time delays during transmission.

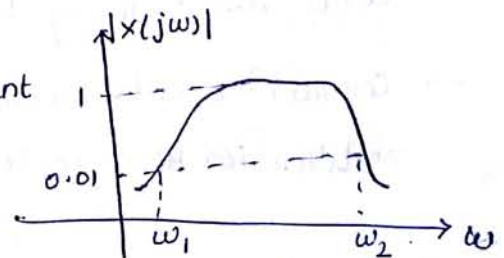
SIGNAL BANDWIDTH: The band of frequencies that contains most of signal energy is called B.W of signal denoted by f_m .

→ It is the range of significant signal frequencies which are present in the signal.

→ observe in the waveform $x(t)$ has significant frequencies from ω_1 to ω_2 .

→ The B.W of this signal is $\omega_2 - \omega_1$.

→ All the physically obtained signals have limited bandwidth.

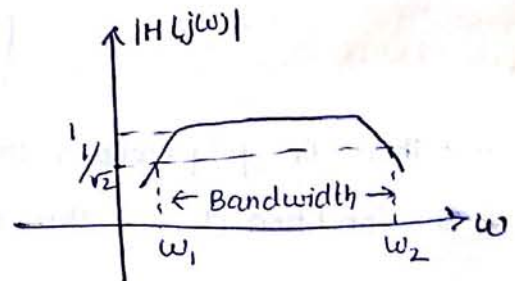


SYSTEM BANDWIDTH

→ The B.W of a system is defined as

○ range of frequencies over which $|H(\omega)|$ remains within $1/\sqrt{2}$ times of its mid-band value. For distortionless transmission the

system must have infinite B.W but physical system are limited to finite B.W.



→ So a system with finite B.W can provide distortionless transmission for a band limited signal if $|H(\omega)|$ remains constant over B.W of the signal.

→ The range of frequencies for which magnitude $|H(j\omega)|$ of the systems remains within $1/\sqrt{2}$ of its maximum value

CASUAL SYSTEMS:

→ A system is said to be causal if $h(t) = 0$ if $t < 0$
 $h(t-t_0) = 0 ; t < t_0$

i.e if i/p is zero for $t < t_0$, then o/p is also zero for $t < t_0$.

→ Any system which does not obey the above rule is non-causal system.
→ If two i/p to a causal system are equal upto some time 't₀' then corresponding o/p must be equal upto that time instant.

POLY-WIENER CRITERION

→ This gives the condition for causality in frequency domain (or) in other words the frequency domain equivalent of causal system i.e $H(\omega)$.

→ Consider a system with transfer function $H(\omega)$, the necessary and sufficient condition for $H(\omega)$ to be transfer function of causal fn is

$$\int_{-\infty}^{\infty} \frac{|\ln|H(j\omega)||}{1+\omega^2} d\omega < \infty \rightarrow \textcircled{1}$$

provided $|H(j\omega)|$ is square integral.

$$\int_{-\infty}^{\infty} |H(j\omega)|^2 d\omega < \infty \rightarrow \textcircled{2}$$

→ This is poly-wiener criteria. If condition $\textcircled{2}$ is not satisfied then the condition $\textcircled{1}$ is neither necessary nor sufficient.

PHYSICAL REALIZABILITY:

→ A system is said to be physically realizable if it obeys the causal condition.

i.e $h(t) = 0$ for $t < 0$.

Ex: $H(\omega) = \frac{1}{1+j\omega}$

$$h(t) = e^{-t} u(t)$$

$$= 0 \quad \text{for } t < 0$$

So the above system for transfer fn is realizable in freq. domain.

$$\int_{-\infty}^{\infty} \frac{|\ln|H(\omega)||}{|H(\omega)|^2} d\omega < \infty$$

→ The frequency domain statements can be interpreted as $|H(\omega)|$ if a physically realizable system may be zero for some discrete frequency but it can never be zero for a finite band of frequencies.

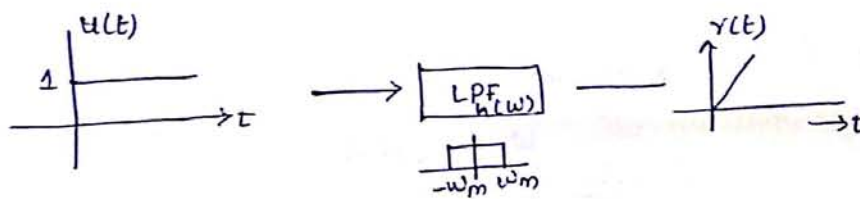
→ $H(\omega)$ for a realisable system cannot decay faster than a function of exponential order.

Ex: A system with T.F $e^{-\omega}$ is realisable whereas $e^{-\omega^2}$ is not as it decays faster.

RELATIONSHIP BETWEEN RISE TIME AND BANDWIDTH:

→ If a unit step $f(t) = u(t)$ is applied to an ideal LPF, the o/p will show a gradual rise instead of a sharp rise in the i/p.

→ The rise time (t_r) is the time required by the response to reach its final value from initial value.



Transfer function of ideal low pass filter is

$$H(\omega) = |H(\omega)| e^{j\theta(\omega)}$$

$$= G(\omega) e^{-j\omega t_0}$$

↓
Rectangular pulse
with magnitude k .

for $-B < f \leq B$ i.e. $-\omega_m \leq \omega \leq \omega_m$ where $\omega_m = 2\pi B$.

and $\theta(\omega) = -2\pi f t_0 = -\omega t_0$.

→ Fourier transform of unit step $f(t) = u(t)$

$$FT\{u(t)\} \Rightarrow u(\omega) = \pi\delta(\omega) + \frac{1}{j\omega}$$

→ Fourier transform of response $R(\omega)$, input and $H(\omega)$ related as

$$R(\omega) = \left[\pi\delta(\omega) + \frac{1}{j\omega} \right] H(\omega) = \pi\delta(\omega) \cdot H(\omega) + \frac{1}{j\omega} H(\omega)$$

$\delta(\omega)$ exists only for $\omega=0$ and $H(\omega)|_{\omega=0}=1$

$$R(\omega) = \pi \delta(\omega) + \frac{1}{j\omega} H(\omega)$$

By taking IFT for above eqn

$$\begin{aligned} r(t) &= \text{IFT}[R(\omega)] = \text{IFT}\left\{\pi \delta(\omega) + \frac{1}{j\omega} H(\omega)\right\} \\ &= \text{IFT}\left\{\pi \delta(\omega) + \frac{1}{j\omega} G(\omega) e^{-j\omega t_0}\right\} \quad (\because H(\omega) = G(\omega) e^{-j\omega t_0}) \end{aligned}$$

Inverse fourier transform of $\pi \delta(\omega)$ is $\frac{1}{2}$.

$$\left. \begin{aligned} &\therefore 1 \rightarrow 2\pi \delta(\omega) \\ &\frac{1}{2} \leftarrow \pi \delta(\omega) \end{aligned} \right\}$$

$$\begin{aligned} r(t) &= \frac{1}{2} + \text{IFT}\left\{\frac{1}{j\omega} G(\omega) e^{-j\omega t_0}\right\} \\ &= \frac{1}{2} + \frac{1}{2\pi} \int_{-\infty}^{\infty} \frac{1}{j\omega} G(\omega) e^{-j\omega t_0} \cdot e^{j\omega t} d\omega \end{aligned}$$

we know $G(\omega) = 1$ for $-\omega_m \leq \omega \leq \omega_m$

$$\begin{aligned} &= \frac{1}{2} + \frac{1}{2\pi} \int_{-\omega_m}^{\omega_m} \frac{e^{j\omega(t-t_0)}}{j\omega} d\omega \\ &= \frac{1}{2} + \frac{1}{2\pi} \int_{-\omega_m}^{\omega_m} \frac{\cos \omega(t-t_0) + j \sin \omega(t-t_0)}{j\omega} d\omega \\ &= \frac{1}{2} + \frac{1}{2\pi} \int_{-\omega_m}^{\omega_m} \frac{\cos \omega(t-t_0)}{j\omega} d\omega + \frac{1}{2\pi} \int_{-\omega_m}^{\omega_m} \frac{\sin \omega(t-t_0)}{\omega} d\omega \end{aligned}$$

\downarrow
 = zero for its odd term

$\underbrace{\hspace{2cm}}_{\downarrow}$
 even

$\text{Si}(x)$ is an odd fn
 $\text{Si}(-x) = -\text{Si}(x)$

(ii) $\text{Si}(0) = 0$

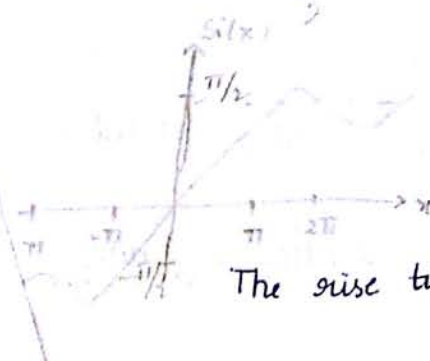
(iii) $\text{Si}(\omega) = \frac{\pi}{2}$

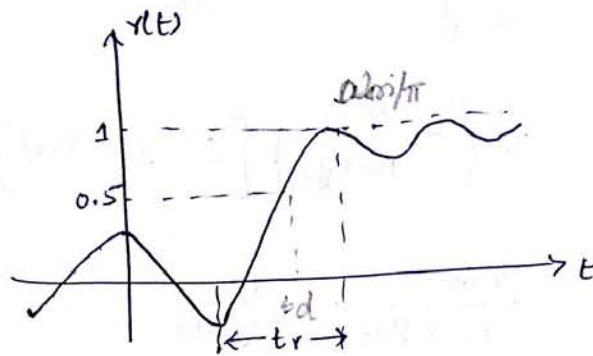
$\text{Si}(-\omega) = -\frac{\pi}{2}$

$$\begin{aligned} r(t) &= \frac{1}{2} + \frac{1}{2\pi} \times 2 \int_0^{\omega_m} \frac{\sin \omega(t-t_0)}{\omega} d\omega = \frac{1}{2} + \frac{1}{\pi} \int_0^{\omega_m} \frac{\sin \omega(t-t_0)}{\omega} d\omega \\ &= \frac{1}{2} + \frac{1}{\pi} \left[\text{Si} \omega(t-t_0) \right]_0^{\omega_m} \end{aligned}$$

$= \frac{1}{2} + \frac{1}{\pi} \text{Si} \omega_m(t-t_0) \rightarrow$ sine integral

The rise time is given as $t_r = \frac{2\pi}{\omega_m} = \frac{1}{B}$
 $\rightarrow \frac{d|g(t)|}{dt} \Big|_{t=t_0} = \frac{1}{\pi} \cos[\omega_m(t-t_0)] \Big|_{t=t_0} = \frac{1}{\pi}$
 \downarrow
 cut off frequency of LPF $\frac{1}{t_r} = \frac{\omega_m}{\pi} \Rightarrow t_r = \frac{\omega_m}{\pi}$





$$\omega_c \rightarrow \infty$$

$$y(t) = 1$$

$$\omega_c \rightarrow -\infty$$

$$y(t) = 0$$

Note: { Elements of block diagram }

① Adder: $x_1(t)$ and $x_2(t)$ are inputs to a block with a plus sign (+), resulting in $y(t) = x_1(t) + x_2(t)$.

which performs the addition of two signal sequences to form sum

② constant multiplier:

$x(t)$ is input to a block with a multiplier 'a', resulting in $y(t) = a x(t)$.

It represents applying a scale factor on i/p $x(t)$.

③ Signal multiplier:

$x_1(t)$ and $x_2(t)$ are inputs to a block with a multiplier 'x', resulting in $y(t) = x_1(t) \cdot x_2(t)$.

The multiplication of two signal to form product sequence.

PROBLEMS:

① The impulse response of continuous time system is given as

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

Determine the frequency response & plot the magnitude phase plots.

Sol

Take FT

$$H(\omega) = \int_{-\infty}^{\infty} h(t) \cdot e^{-j\omega t} dt$$

$$= \int_{-\infty}^{\infty} \frac{1}{RC} \cdot e^{-t/RC} \cdot u(t) \cdot e^{-j\omega t} dt$$

$$= \frac{1}{RC} \int_{0}^{\infty} e^{-t/RC} \cdot e^{-j\omega t} dt \quad (\because u(t) = 1 \text{ for } t \geq 0 \text{ and } 0 \text{ otherwise})$$

$$= \frac{1}{RC} \int_0^{\infty} e^{-t(j\omega + \frac{1}{RC})} dt$$

$$= \frac{1}{RC} \left(-\frac{1}{j\omega + \frac{1}{RC}} \right) \left[e^{-t(j\omega + \frac{1}{RC})} \right]_0^{\infty}$$

$$H(\omega) = \frac{1/RC}{j\omega + 1/RC} = \frac{1}{1 + j\omega RC}$$

Magnitude & phase

$$H(\omega) = \frac{1}{1 + j\omega RC} \times \frac{1 - j\omega RC}{1 - j\omega RC} = \frac{1 - j\omega RC}{1 + (\omega RC)^2}$$

$$= \frac{1}{1 + (\omega RC)^2} + j \frac{-\omega RC}{1 + (\omega RC)^2}$$

$$|H(\omega)| = \left\{ \frac{1}{[1 + (\omega RC)^2]^2} + \frac{(\omega RC)^2}{[1 + (\omega RC)^2]^2} \right\}^{1/2}$$

$$= \frac{1}{\sqrt{1 + (\omega RC)^2}}$$

$$\angle H(\omega) = \tan^{-1} \left\{ \frac{(-\omega RC) / [1 + (\omega RC)^2]}{1 / [1 + (\omega RC)^2]} \right\} = -\tan^{-1}(\omega RC)$$

If $RC = 1$, $|H(\omega)| = \frac{1}{\sqrt{1 + \omega^2}}$; $\angle H(\omega) = -\tan^{-1}(\omega)$.



(11)

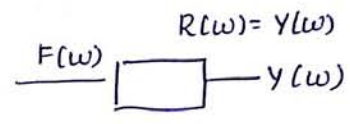
Q For the system shown find the T.T & impulse response of the system.

$$f(t) = \begin{cases} e^{-at} & t > 0 \\ 0 & \text{elsewhere} \end{cases} ; y(\omega) = \frac{1}{a+j\omega}$$

Sol

$$H(\omega) = \frac{R(\omega)}{F(\omega)}$$

$$F(\omega) = e^{-a\omega}$$



$$F(\omega) = \frac{1}{a+j\omega} ; y(\omega) = \frac{1}{a+j\omega}$$

$$H(\omega) = \frac{1/a+j\omega}{1/a+j\omega} = \frac{a+j\omega}{a+j\omega}$$

$$F^{-1} \left[\frac{a+j\omega}{a+j\omega} \right] \Rightarrow \frac{a+\alpha-\alpha+j\omega}{\alpha+j\omega} = \frac{a-\alpha}{\alpha+j\omega} + \frac{\alpha+j\omega}{\alpha+j\omega}$$

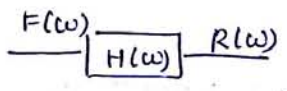
$$= \frac{a-\alpha}{\alpha+j\omega} + 1$$

$$h(t) = (a-\alpha) e^{-\alpha t} u(t) + \delta(t)$$

Q The linear system impulse response is $[e^{-2t} + e^{-3t}] u(t)$ find the excitation to produce an o/p of $t \cdot e^{-2t} u(t)$?

Sol

$$h(t) = [e^{-2t} + e^{-3t}] u(t)$$



$$r(t) = t \cdot e^{-2t} u(t)$$

$$H(\omega) = \frac{R(\omega)}{F(\omega)}$$

$$F(\omega) = \frac{R(\omega)}{H(\omega)}$$

$$r(t) = t \cdot e^{-2t} u(t) \xleftrightarrow{FT} \frac{1}{(2+j\omega)^2} \quad \left(\because t \cdot e^{-at} u(t) \leftrightarrow \frac{1}{(a+j\omega)^2} \right)$$

$$R(\omega) = \frac{1}{(2+j\omega)^2}$$

$$h(t) = e^{-2t} u(t) + e^{-3t} u(t)$$

$$H(\omega) = \frac{1}{2+j\omega} + \frac{1}{3+j\omega}$$

$$R(\omega) = \frac{\frac{1}{(2+j\omega)^2}}{\frac{3+j\omega+2+j\omega}{(2+j\omega)(3+j\omega)}} = \frac{1}{2+j\omega} \times \frac{3+j\omega}{5+2j\omega}$$

$$\frac{3+j\omega}{(2+j\omega)(5+2j\omega)} = \frac{A}{2+j\omega} + \frac{B}{5+2j\omega}$$

$$3+j\omega = A(5+2j\omega) + B(2+j\omega)$$

$$\text{put } 3+j\omega = 5A + 2B + j\omega(2A+B)$$

$$\text{put } j\omega = 0 \quad ; \quad \text{put } j\omega(-2)$$

$$(3 = 5A + 2B) \times 1$$

$$(1 = 2A + B) \times 2$$

$$A = 1, B = -1$$

$$R(\omega) = \frac{1}{2+j\omega} - \frac{1}{5+2j\omega} = \frac{1}{2+j\omega} - \frac{1}{2[5/2+j\omega]}$$

$$r(t) = e^{-2t} u(t) - \frac{1}{2} e^{-5/2t} u(t)$$

DIFFERENTIAL EQUATION:

→ To obtain frequency response & impulse response.

$$\sum_{k=0}^N a_k \frac{d^k}{dt^k} y(t) = \sum_{k=0}^M b_k \frac{d^k}{dt^k} x(t)$$

differentiation property of FT is

$$\frac{d}{dt} x(t) \xleftrightarrow{FT} j\omega x(\omega)$$

$$\sum_{k=0}^N a_k (j\omega)^k Y(\omega) = \sum_{k=0}^M b_k (j\omega)^k X(j\omega)$$

system transfer fn. \swarrow

$$H(\omega) = \frac{Y(\omega)}{X(\omega)} = \frac{\sum_{k=0}^M b_k (j\omega)^k}{\sum_{k=0}^N a_k (j\omega)^k}$$

PROBLEMS:

- ① The differential equation of system is given as $\frac{d^2 y(t)}{dt^2} + 5 \frac{dy(t)}{dt} + 6y(t) = -\frac{dx(t)}{dt}$

Determine the frequency response & impulse response.

Sol

$$\frac{d^2 y(t)}{dt^2} + 5 \frac{dy(t)}{dt} + 6y(t) = -\frac{dx(t)}{dt}$$

Taking F.T

$$(j\omega)^2 Y(\omega) + 5(j\omega)Y(\omega) + 6Y(\omega) = -j\omega X(\omega)$$

$$Y(\omega) [(j\omega)^2 + 5j\omega + 6] = -j\omega X(\omega)$$

$$H(\omega) = \frac{Y(\omega)}{X(\omega)} = \frac{-j\omega}{(j\omega)^2 + 5j\omega + 6}$$

$$H(\omega) = \frac{-j\omega}{(j\omega+2)(j\omega+3)} = \frac{A}{j\omega+2} + \frac{B}{j\omega+3}$$

$$= \frac{2}{j\omega+2} - \frac{3}{j\omega+3}$$

$$h(t) = [2 \cdot e^{-2t} - 3e^{-3t}] u(t)$$

impulse response of the system. \swarrow

$$\left\{ \because e^{-at} u(t) \xleftrightarrow{FT} \frac{1}{a+j\omega} \right\}$$

② The input voltage to the RC circuit is given by $x(t) = t e^{-t/RC} u(t)$ and impulse response of this circuit is given by $h(t) = \frac{1}{RC} e^{-t/RC} u(t)$. Find output $y(t)$

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

In frequency domain

$$Y(\omega) = X(\omega) H(\omega)$$

$$\text{and } H(\omega) = F[h(t)]$$

$$H(\omega) = F\left\{\frac{1}{RC} e^{-t/RC} u(t)\right\}$$

$$= \frac{1}{RC} \cdot \frac{1}{\frac{1}{RC} + j\omega} = \frac{1}{1 + j\omega RC}$$

$$X(\omega) = F[t \cdot e^{-t/RC} u(t)]$$

$$= \int_{-\infty}^{\infty} t \cdot e^{-t/RC} e^{-j\omega t} dt = \frac{1}{\left(\frac{1}{RC} + j\omega\right)^2} = \frac{(RC)^2}{(1 + j\omega RC)^2}$$

$$\left[\because t e^{-at} u(t) \xrightarrow{FT} \left(\frac{1}{a + j\omega}\right)^2 \right]$$

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

$$= \frac{(RC)^2}{(1 + j\omega RC)^2} \cdot \frac{1}{(1 + j\omega RC)} = \frac{(RC)^2}{(1 + j\omega RC)^3}$$

$$Y(t) = F^{-1}\{Y(\omega)\} = F^{-1}\left\{\frac{(RC)^2}{(1 + j\omega RC)^3}\right\} = F^{-1}\left\{\frac{(RC)^2}{(RC)^3 \left(\frac{1}{RC} + j\omega\right)^3}\right\}$$

$$\boxed{y(t) = \frac{1}{RC} \cdot \frac{t^2 \cdot e^{-t/RC}}{2} u(t)}$$

$$1) h(t) = e^{-5|t|}$$

For stability $\int_{-\infty}^{\infty} |h(t)| dt < \infty$

$$\therefore \int_{-\infty}^{\infty} |h(t)| dt = \int_{-\infty}^{\infty} |e^{-5|t|}| dt = \int_{-\infty}^{\infty} e^{-5|t|} dt$$

$$= \int_{-\infty}^0 e^{5t} dt + \int_0^{\infty} e^{-5t} dt = \left[\frac{e^{5t}}{5} \right]_{-\infty}^0 + \left[\frac{e^{-5t}}{-5} \right]_0^{\infty}$$

$$= \frac{2}{5} = \text{constant} / \text{so system is stable.}$$

$$2) h(t) = e^{4t} u(t)$$

$$= \int_{-\infty}^{\infty} |e^{4t} u(t)| dt = \int_{-\infty}^{\infty} e^{4t} u(t) dt$$

$$= \int_0^{\infty} e^{4t} dt = \frac{e^{\infty}}{4} - \frac{e^0}{4} = \infty - \frac{1}{4} = \infty \text{ (Unstable)}$$

$$3) h(t) = e^{-4t} u(t) \text{ (stable)}$$

$$4) h(t) = t \cos t u(t) \text{ (unstable)}$$

$$\int_0^{\infty} t \cos t dt$$

$$5) h(t) = e^{-t} \sin t u(t) \text{ (stable)}$$

$$= \int_0^{\infty} e^{-t} \sin t dt$$

→ The system produces the o/p of $y(t) = e^{-t} u(t)$ for an input of $x(t) = e^{-2t} u(t)$. Determine the impulse response and frequency response of the system.

Sol $y(t) = e^{-t} u(t)$
 $x(t) = e^{-2t} u(t)$

Consider standard Fourier transform pair $e^{-at} u(t) \xleftrightarrow{FT} \frac{1}{a+j\omega}$

$$Y(\omega) = \frac{1}{1+j\omega}; \quad X(\omega) = \frac{1}{2+j\omega}$$

From equation

$$H(\omega) = \frac{Y(\omega)}{X(\omega)} = \frac{1/1+j\omega}{1/2+j\omega} = \frac{2+j\omega}{1+j\omega}$$

Multiply and divide the numerator & denominator by $1-j\omega$

$$H(\omega) = \frac{2+j\omega}{1+j\omega} \times \frac{1-j\omega}{1-j\omega} = \frac{(2+j\omega)(1-j\omega)}{(1-j\omega)(1+j\omega)} = \frac{2-2j\omega+j\omega+\omega^2}{1+\omega^2}$$

$$= \frac{2+\omega^2-j\omega}{1+\omega^2} \Rightarrow \frac{2+\omega^2}{1+\omega^2} + j \frac{-\omega}{1+\omega^2}$$

$$\text{Magnitude } |H(\omega)| = \sqrt{\left[\frac{2+\omega^2}{1+\omega^2} \right]^2 + \left[\frac{-\omega}{1+\omega^2} \right]^2}$$

$$|H(\omega)| = \sqrt{\frac{4+\omega^2}{1+\omega^2}}; \quad \angle H(\omega) = \tan^{-1} \left(\frac{-\omega}{\frac{2+\omega^2}{1+\omega^2}} \right)$$

$$= -\tan^{-1} \left(\frac{\omega}{2+\omega^2} \right)$$

$$\therefore H(\omega) = \frac{2+j\omega}{1+j\omega} \Rightarrow \frac{1+j\omega+1}{1+j\omega} \Rightarrow 1 + \frac{1}{1+j\omega}$$

Inverse Fourier transform

$$h(t) = \text{IFT} \{ H(\omega) \} = \delta(t) + e^{-t} u(t)$$

↓
impulse response.

(1) The transfer function of LPF is given by

$$H(\omega) = \begin{cases} (1+k \cos \omega T) e^{-j\omega T} & ; |\omega| < 2\pi B \\ 0 & ; |\omega| > 2\pi B \end{cases}$$

Determine the output $y(t)$ when a pulse $x(t)$ bandlimited in B is applied at the input.

sol

$$Y(\omega) = X(\omega) H(\omega)$$

$$= X(\omega) [1 + k \cos \omega T] e^{-j\omega T}$$

$$= X(\omega) e^{-j\omega T} + k X(\omega) \cos \omega T e^{-j\omega T}$$

we know

$$x(t-\tau) + x(t+\tau) \longleftrightarrow 2X(\omega) \cos \omega T$$

$$x(t-\tau) \longleftrightarrow X(\omega) e^{-j\omega \tau}$$

$$y(t) = F^{-1}[Y(\omega)]$$

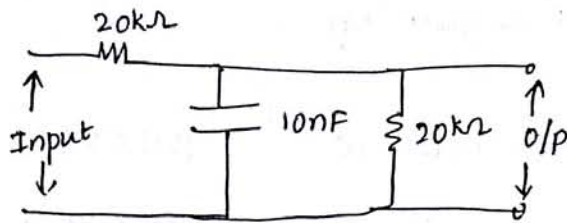
$$= F^{-1}[X(\omega) e^{-j\omega T} + k X(\omega) e^{-j\omega T} \cos \omega T]$$

$$= x(t-T) + \frac{k}{2} [x(t-T-T) + x(t-T+T)]$$

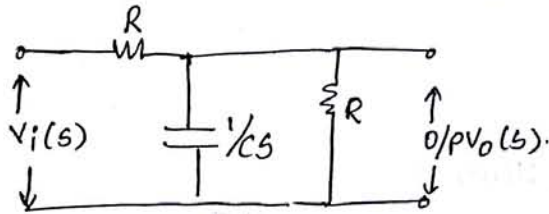
$$y(t) = x(t) + \frac{k}{2} [x(t-T) + x(t+T)]$$

delayed by T .

2) Determine the maximum bandwidth of signals that can be transmitted through low pass RC filter as shown in figure, if over this bandwidth, the gain variation is to be 10% and the phase variation is to be within 7% of ideal characteristics.



RC network transformed into s-domain representation.



$$H(s) = \frac{V_o(s)}{V_i(s)}$$

$$= \frac{[R \parallel (1/cs)]}{([R \parallel (1/cs)] + R)}$$

$$= \frac{(R/cs) / [R + 1/cs]}{[R/cs] / [R + 1/cs] + R}$$

$$= \frac{R / (1 + sCR)}{[R / (1 + sCR)] + R} = \frac{R}{R + R(1 + sCR)} = \frac{R}{R(1 + 1 + sCR)}$$

$$H(s) = \frac{1}{2 + sCR}$$

But given $R = 20k\Omega$

$C = 10nF$.

$$H(s) = \frac{1}{2 + s(10 \times 10^{-9} \times 20 \times 10^3)} = \frac{1}{2 + (25/10^4)s} = \frac{10^4}{2 \times 10^4 + 25} = \frac{1}{2 + s(2 \times 10^{-4})}$$

$$\therefore H(s) = \frac{5000}{s + 10000}$$

put $s = j\omega$

$$H(\omega) = \frac{5000}{j\omega + 10000}$$

$$|H(\omega)| = \frac{5000}{\sqrt{\omega^2 + 10000}}$$

$$\phi(\omega) = -\tan^{-1}\left(\frac{\omega}{10000}\right)$$

$$\text{At } \omega = 0, |H(\omega)|_{\omega=0} = \frac{5000}{10000} = 0.5$$

But there is 10% variation in gain over bandwidth B.

$$|H(\omega)| = 0.5 - 0.5 \times 10\% = 0.45$$

$$|H(\omega)| = \frac{5000}{\sqrt{B^2 + 10^8}}$$

$$B^2 + 10^8 = \left(\frac{5000}{0.45}\right)^2 \Rightarrow B^2 = 23.46 \times 10^6$$

$$B = 4.84 \text{ KHZ}$$

$$\text{But } B = 2\pi f$$

$$f = \frac{B}{2\pi} = \frac{4.84 \times 10^3}{2\pi} = 770.8 \text{ HZ}$$

phase at frequency, $f = 770.8 \text{ HZ}$

$$\phi(\omega) = -\tan^{-1}\left(\frac{4.84}{10}\right) = -25.83\%$$

(2) There are several possible ways of estimating an essential bandwidth of non-bandlimited signal. For a low pass signal, for example, the essential b.w may