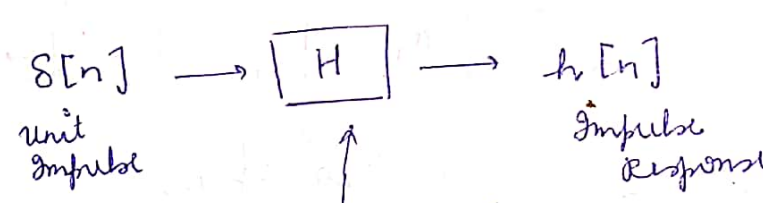
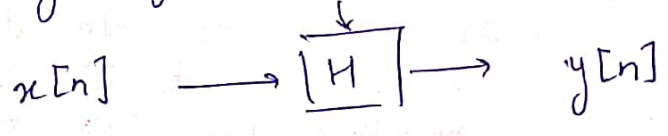


Impulse Response
 For this, I knew what the response of a linear sys. was
 do



I want to know what'll happen if I put an arbitrary signal



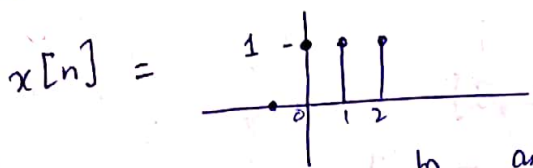
We showed that this O/P was convolution of I/P with impulse resp.

$$y[n] = x[n] * h[n]$$

$$= \sum_{k=-\infty}^{\infty} x[k] h[n-k]$$

what does this convolution sum doing?

Consider this LTI system
 The HOD. $y[n] = x[n] - 2x[n-1] + 3x[n-2]$
 what is the response to



* $[1, 1, 1]$
 ↑
 origin (n=0)

there're many ways to answer this

① Direct

$y[-1] = 0$

also -2, -3, ..

$y[0] = 1$

$y[1] = 1 - 2(1) = -1$

$[1, -1, 2, 1, 3]$
 ↑
 n=0

$y[2] = 1 - 2 + 3 = 2$

$y[3] = -2 + 3 = 1$

$y[4] = 3$

list of no. One thing that'll make this a bit easier
 is to represent as *
 never do this thing in practice (large 9/P)

② using convolution sum

$y[n] = \sum_{k=-\infty}^{\infty} x[k] h[n-k]$

what is the impulse response?
 go back to formula & see what the 9.R is

$\delta[n] = [1]$

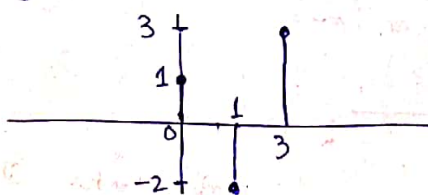
$h[0] = 1$

$h[1] = 0 - 2 = -2$

$[1, -2, 3]$
 ↑

$h[2] = 0 - 0 + 3 = 3$

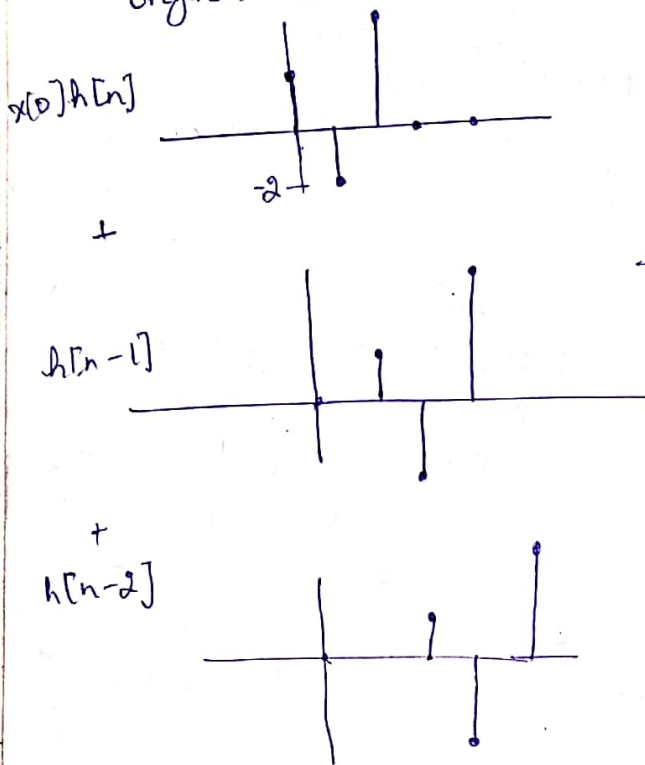
$h[3] = 0$



mark the heights!

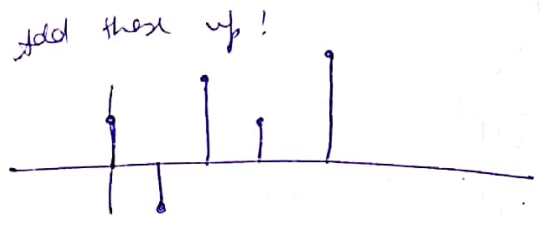
$$y[n] = x[0]h[n] + x[1]h[n-1] + x[2]h[n-2]$$

That's like saying I'm doing is I'm multiplying these delayed versions of the original signal by the coeff. of the signal



$$\begin{bmatrix} 1 & -2 & 3 \\ 0 & 1 & -2 & 3 \\ 0 & 0 & 1 & -2 & 3 \end{bmatrix}$$

$$= [1, -1, 2, 1, 3]$$

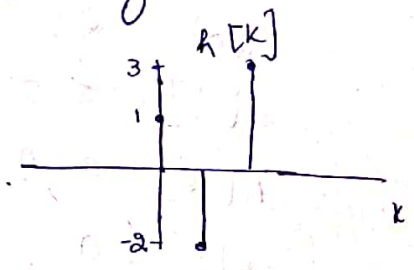
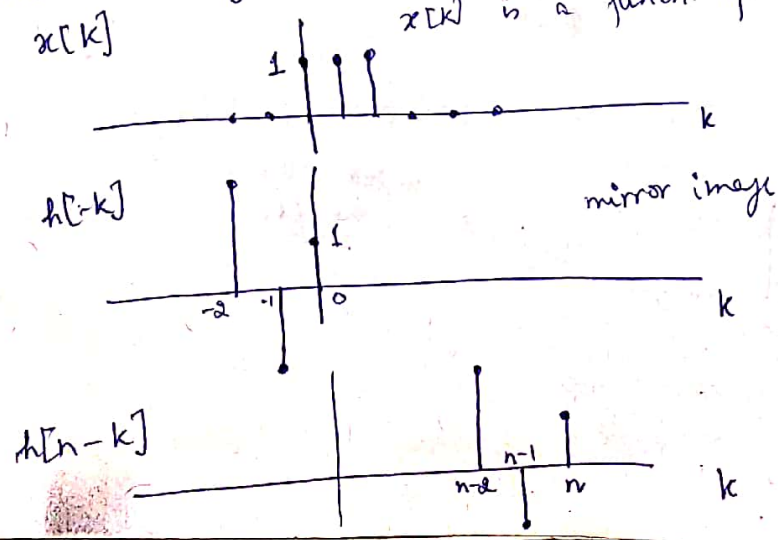


It's still tedious; add up zillion copies of shifted response everywhere we've new signal that leads to more graphical method of doing convolution

③ Flip & slide one signal

$$\sum_{k=-\infty}^{\infty} x[k]h[n-k]$$

Another way of looking at this is following: $x[k]$ is a funcn. of k

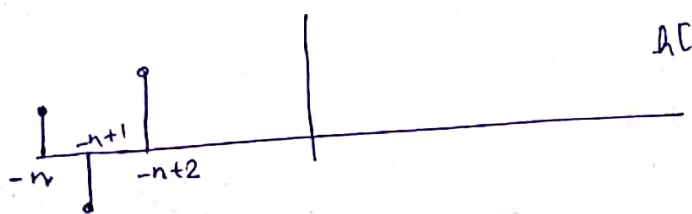
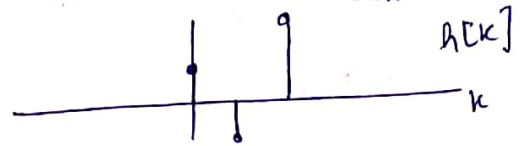


looks like the signal slid to the right by n units take what used to be at $k=0$ & move it over to n

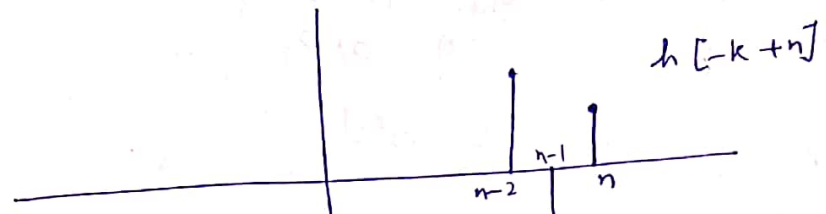
mul. corr. entries log. & sum them.

$$h[k] \rightarrow h[n-k] = h[-k+n]$$

↑
const.

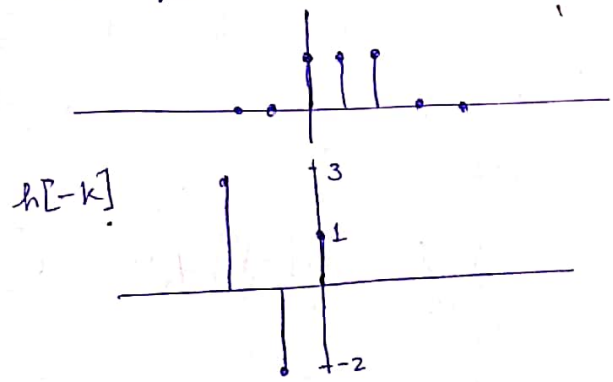


n is a +ve no.



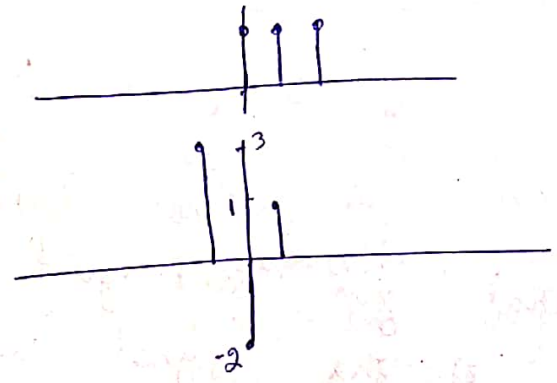
if it is to be done in proper order!

$$y[0] = \sum_{k=-\infty}^{\infty} x[k] h[-k]$$



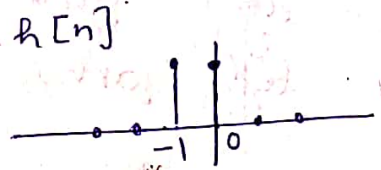
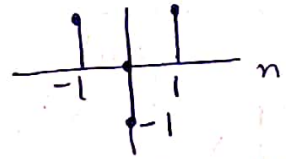
$$y[0] = 1$$

$$y[1] = \sum x[k] h[1-k]$$

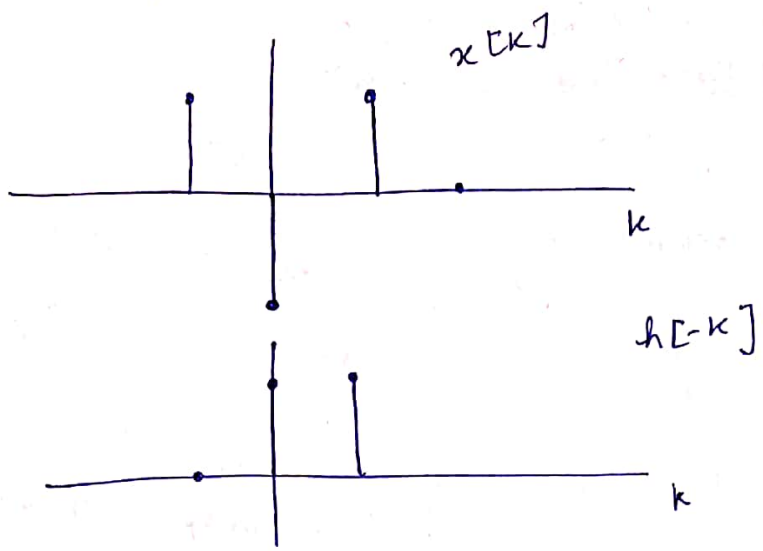


$$y[1] = -2 + 1 = -1$$

eg. 2 $x[n] =$



1st thing I draw them as a function of k



this is for $n=0$ position of o/p

I look at the corr. products & add

$$y[0] = 0 - 1 + 1 = 0$$

when I move the signal to the right see the overlaps

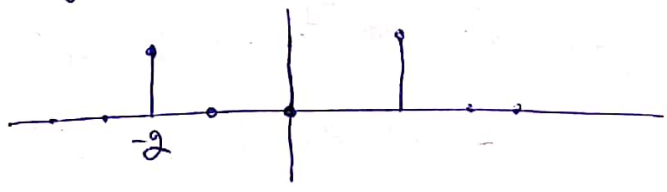
$$y[1] = 1$$

$$y[2] = 0$$

Also move the signal to the left $x[-1] \neq 0$

$$y[-1] = 0$$

$$y[-2] = 1$$

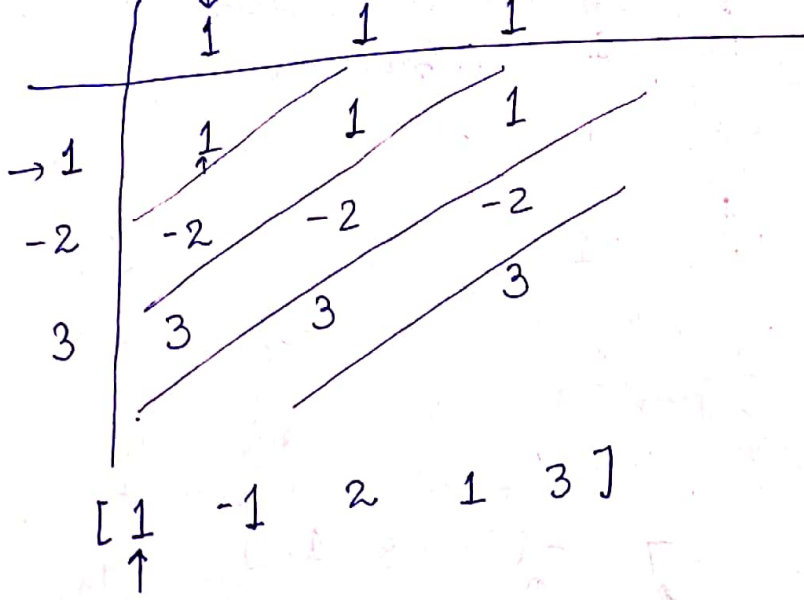


\therefore it is important to know that you can't just stop for the right part of the signal but also for the left part if $x[k] \neq 0$ for $k < 0$

Faster way of doing convolution!

④ convolution Array (wozny)

h

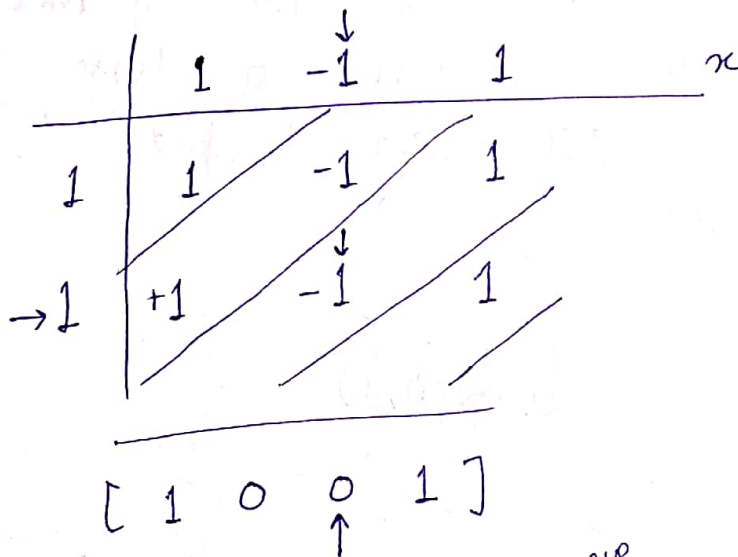


look at the diagonals of the table & add up the corr. values

- How do I know what is the zero element of the result

mark the zero response of the I/P and zero resp. of I/m. R. & where the 2 cross is the zero element of the O/P

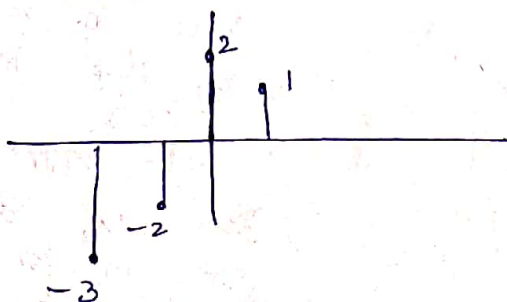
eg ②



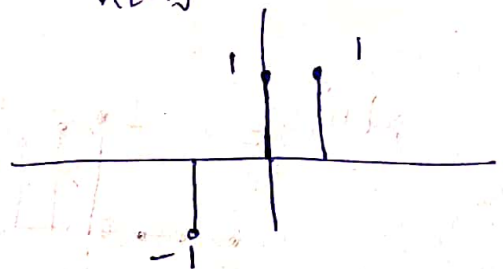
(3rd sum \rightarrow see arrow)

when you've a finitely I/P signal & impulse resp. then this is reasonable way to use this \rightarrow but cannot for any I/P signals particularly those that're ∞ long.

eg ③ $x[n]$



$A[n]$



	-3	-2	2	1	x
-1	3	2	-2	-1	
→ 1	-3	-2	+2	+1	
1	-3	-2	2	1	

$$[3 \quad -1 \quad -7 \quad -1 \quad 3 \quad 1]$$

↑
 (4th element added)

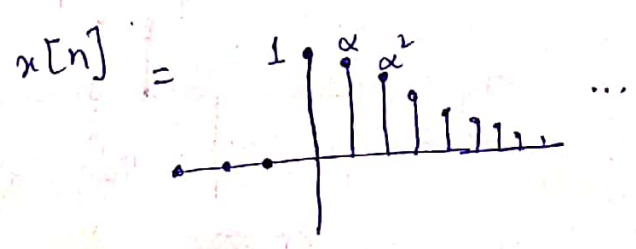
It's true that a lot of times we deal with what are called F.I.R systems (ones for which $h[n] \neq 0$ for finite no. of values) & often we're looking at S/Ps that're ∞ large. So, in that case we still need to find out how to do convolution

so,

eg. $x[n] = \alpha^n u[n] \quad \alpha \in (0, 1)$

$$h[n] = u[n]$$

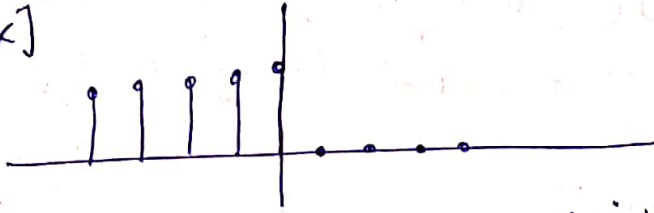
It's good to visualize what the signals look like



since its multiplied by step func. (< 0) nothing happens
 This exp. decay goes on for a long time for any value of n little non-zero x

- It doesn't matter which (order) signal we flip & shift
 In this case it'll be easier to choose h

$h[-k]$



- We can see for $n < 0$ (shifting $h[-k]$ to left) we will be having nothing ($y[n] = 0$) (no overlap)
- Right at zero

$$y[0] = 1 \times 1 = 1$$

$$y[1] = \text{shift one to right} = 1 \times 1 + \alpha$$

$$y[2] = 1 + \alpha + \alpha^2$$

$$y[n] = \begin{cases} 0 & n < 0 \\ \sum_{k=0}^n \alpha^k & n \geq 0 \end{cases} \quad (\text{for } n=2 \quad 1 + \alpha + \alpha^2)$$

$$= \left(\sum_{k=0}^n \alpha^k \right) \cdot u[n] \rightarrow \text{for } n < 0 = 0$$

\downarrow
g.p

Remember

$$\sum_{k=0}^{\infty} \alpha^k = \frac{1}{1-\alpha}$$

$$0 < \alpha < 1$$

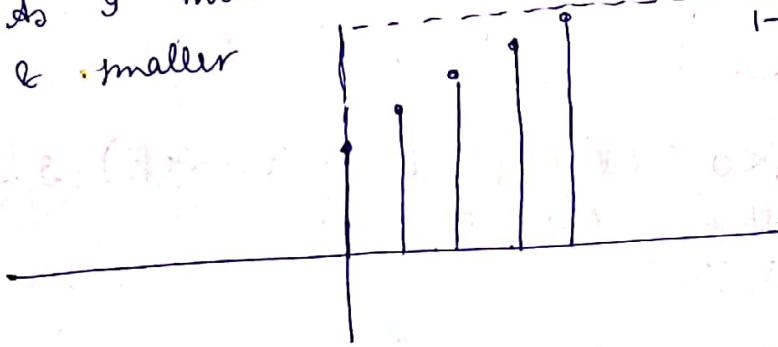
$$\begin{aligned} k' &= 0 \\ n+1 - (n+1) & \\ k' &= k + (n+1) \end{aligned}$$

$$(1 + \alpha + \alpha^2 + \dots) (1 - \alpha) = 1 + \alpha + \alpha^2 + \dots - \alpha - \alpha^2 - \dots = 1$$

$$\begin{aligned} \sum_{k=0}^n \alpha^k &= \sum_{k=0}^{\infty} \alpha^k - \sum_{k=n+1}^{\infty} \alpha^k \\ &= \frac{1}{1-\alpha} - \frac{\alpha^{n+1}}{1-\alpha} = \frac{1 - \alpha^{n+1}}{1-\alpha} \end{aligned}$$

$$y[n] = \left(\frac{1 - \alpha^{n+1}}{1 - \alpha} \right) u[n]$$

As α move ahead, the O/P is getting smaller & smaller (never reaches asymptote) $(\alpha < 1)$



This should be the same as raw convolution sum

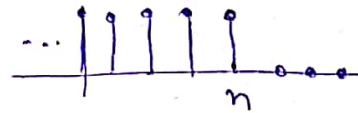
$$y[n] = \sum_{k=-\infty}^{\infty} x[k] h[n-k]$$

$$= \sum_{k=-\infty}^{\infty} \alpha^k u[k] \cdot u[n-k]$$

It's good to think like what do these step funcs. & what does their product look like

funcn of k
 $u[k] = 0$ k is negative

$$\sum_{k=0}^{\infty} \alpha^k u[n-k]$$



$$= \left(\sum_{k=0}^n \alpha^k \right) u[n]$$

($n > 0$)
 for the sum to make sense $k > 0$

$$u[n-k] = 1 \quad \text{for } n-k > 0 \quad \text{or } \boxed{k \leq n}$$

sum doesn't dip below lower index (i.e. $k=0$)

Properties of LTI systems:-

- ① An LTI system is entirely determined by its impulse response. This is not true for non-LTI systems

eg $y[n] = n x[n]$ (in particular not time invar.)

$$\delta[n] \rightarrow n \delta[n] = 0$$

Impulse response is zero

For LTI then I'd never get non-zero O/P for any I/P signal but for non-LTI it's not so

(power of LTI sys.)

don't have to figure what happens to other signals

(But remem. I can do this things also if I've other responses like step)

② Commutative

$$x[n] * h[n] = h[n] * x[n]$$

$$\sum_{k=-\infty}^{\infty} x[k] h[n-k]$$

I keep x as it is
flip h & run h
across the signal

simple change of var.

$$\begin{aligned} m = n - k & \quad \text{if } k = -\infty \text{ to } \infty \\ k = n - m & \quad m = -\infty \text{ to } \infty \end{aligned}$$

$$\sum_{m=-\infty}^{\infty} x[n-m] h[m]$$

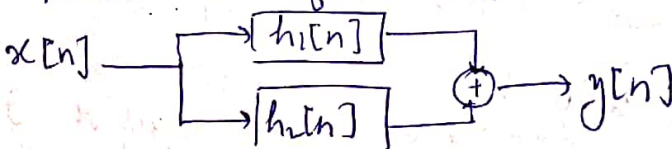
interchange the order of the signals

③ Distributive Property
 $x[n] * (h_1[n] + h_2[n])$

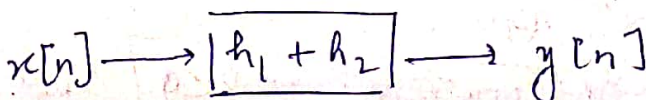
$$= x[n] * h_1[n] + x[n] * h_2[n]$$

go through 2 systems in 11

falls out from LTI prop. of sys.
what this means is if I have a signal that goes through

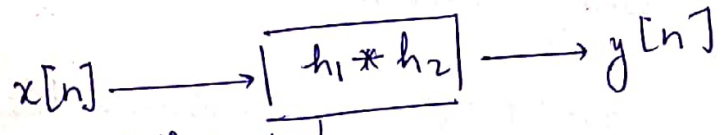
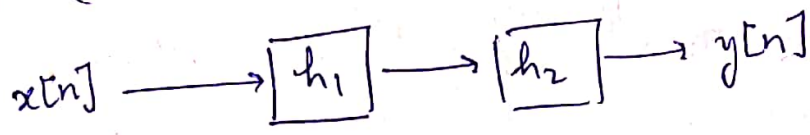


an eq. sys. is this

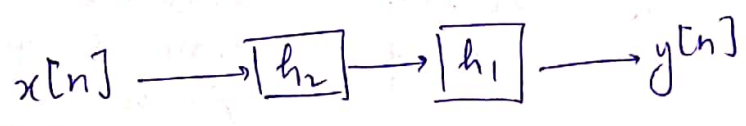


or if I add up 2 signals combined I.P

④ Associative Property
 $x[n] * (h_1[n] * h_2[n])$
 $= (x[n] * h_1[n]) * h_2[n]$



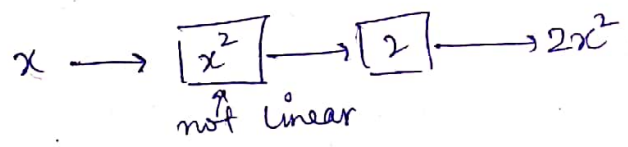
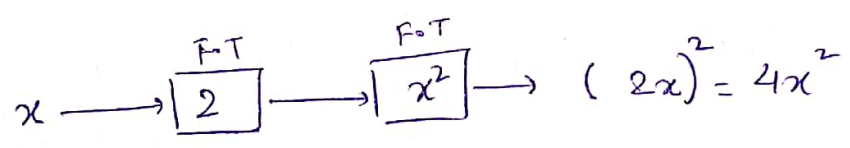
From commutative prop.
 $x[n] \rightarrow [h_2 * h_1] \rightarrow y[n]$



It doesn't matter which order I put the systems in

Conclusions:-

- ① Combine all my LTI syst. into 1 block
- ② put the sys. in any order I want and get the same answer
- (not true for non-linear)



⑤ Causality

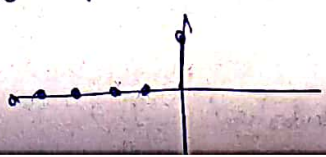
when impulse resp. of the sys. is zero for values less than 0.

$y[n]$ doesn't depend on $x[n+k]$ for $k > 0$
 (not looking into future!)

$$y[n] = \sum_{k=-\infty}^{\infty} h[k] x[n-k]$$

→ if k is negative I'm looking at future value $z[n]$

For causal system $k < 0$ $h[k]$ must be zero to make those zero



$$h[k] = 0 \quad k < 0$$

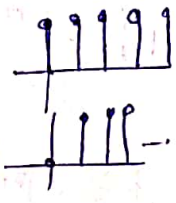
other way of looking at it is that the impulse response cannot be doing stuff before it occurs → system sees the impulse

6. Step Response

$$s[n] = u[n] - u[n-1]$$

$$h[n] = s[n] - s[n-1] \quad (\text{by LTI})$$

↑
step resp. $H(u[n])$



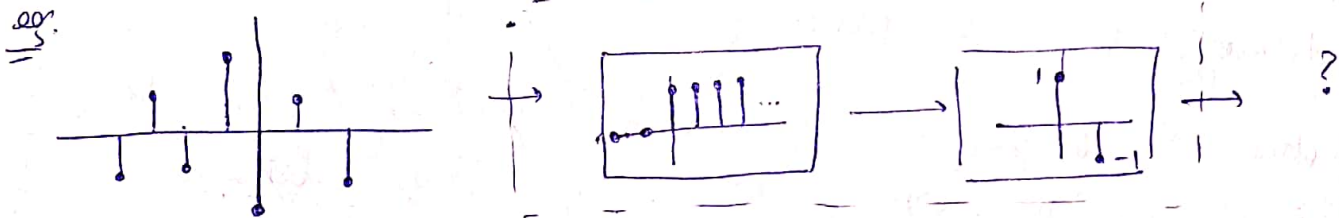
$$u[n] = \sum_{k=-\infty}^n \delta[k]$$

$$s[n] = \sum_{k=-\infty}^n h[k]$$

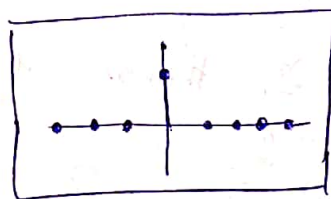
i.e. if I know the step resp. I can get I.R. & vice versa

→ Step Response also charac. the LTI sys.
∴ I can compute the I.R. of the system if I know the step resp.

• STEP response also charac. make life easier!



To do this one has to convolve 1st with 2nd & then the result with 3rd
But convolution of the 2 systems is easy (flip & overlap)



(identity function)

Basic motive is that can I simplify the system to be a bit easier

• many electrical & mechanical systems are described by differential eqns. (Mass Spring, RLC)

• Discrete versions of these are called Difference Equations

$$\sum_{k=0}^N a_k y[n-k] = \sum_{k=0}^M b_k x[n-k]$$

called linear const. Coeff. Difference Eqn

Soln will be of the form

$$y[n] = y_h[n] + y_p[n]$$

↓
homog. soln.
↓
particular soln.

Simpler case

$$y[n] = \sum b_k x[n-k]$$

→ F.I.R

Finite Imp. Sys.

there is nothing to solve here, I use convolution to reach to OIP (now doing these type of problems)

• But if pre. previous values of y , then it'll be a part of recursive eqn. & then go thru machinery of homog. soln. & particular.

• convolution & responses of DTI systems can be greatly simplified with tools of Freq. domain analysis (Fourier & Z-transform)

• In C.T, Laplace Transform was the best way to solve differential eqns.

• Same here once Z trans. life is easy.

Sec 4.1

Fourier Series

(Fourier Analysis → decomposition of signals into sines & cosines (why sines & cosines? It turns out that they naturally occur in a lot of our systems))

• mechanical systems → swinging pend. & spinning wheels

• communication → cellphones, radio etc built on carrier waves (high freq. sin)

• Power signals

(nice thing is that LTI systems respond in a particular way to sinusoidal I/P)

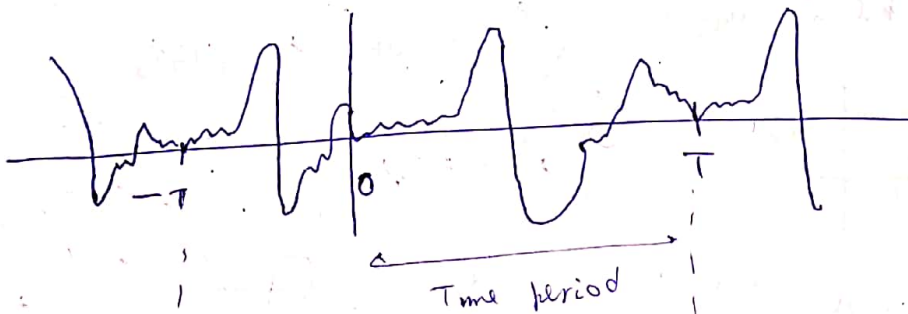
Principle.

• Every periodic c.t. signal can be written as a sum of sinusoids.

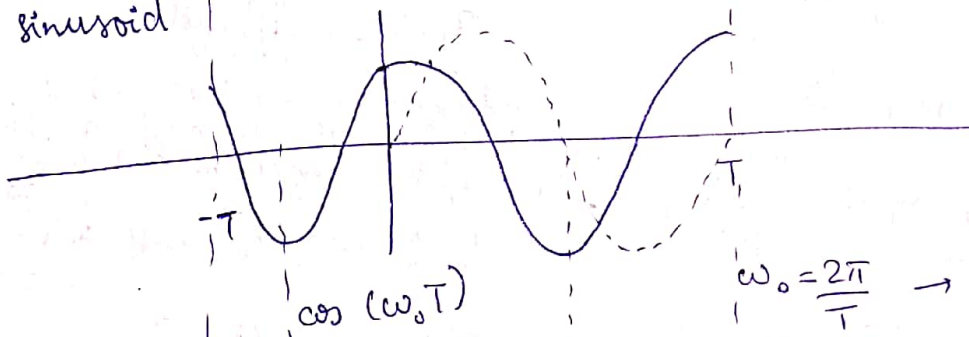
(aperiodic $\rightarrow T \rightarrow \infty$)

• Assume we have a periodic signal $x(t)$

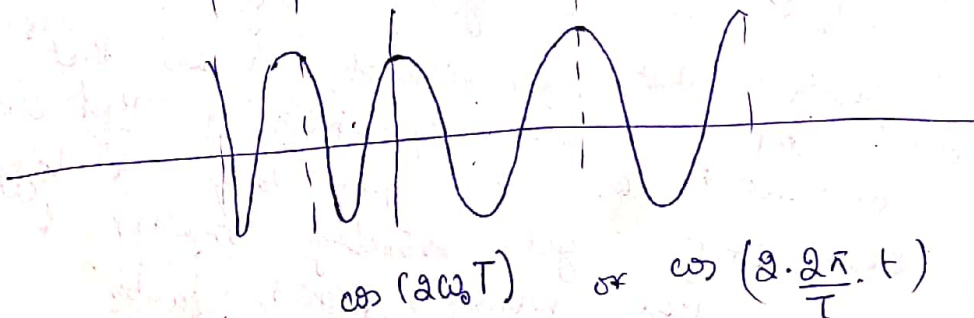
$$x(t+T) = x(t)$$



1st question is? What other signals have the same period T ?
most natural signal that has a period is a sinusoid



$$\omega_0 = \frac{2\pi}{T} \rightarrow \text{Fundamental freq}$$



That's not the only freq. that has period T . I can also think about freq. that wiggle faster

also, $\cos k\omega_0 t$, $\sin k\omega_0 t$, k an integer has this period

$$\text{also, } e^{jk\omega_0 t} = \cos k\omega_0 t + j \sin k\omega_0 t$$

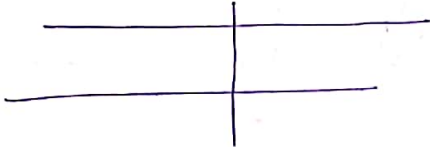
is also periodic (Real & Im.) with period T

by, $\sum_{k=-\infty}^{\infty} a_k e^{jk\omega_0 t}$ is also periodic with period T .
 combination of such signals with period T

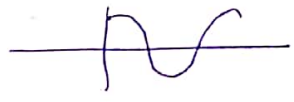

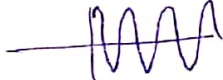
↑ a_k Coeff. (any complex #)
 (because it won't change its period) (only amplitude & phase)

Philosophy of what we're gonna do...
 we're gonna take our signal $x(t)$ & make it look like a bunch of such sinusoids with increasing freq.
 (don't know these a_k)

for $k=0$
 a_0 "DC"



+ $a_1 \cdot (\cos \omega_0 t + j \sin \omega_0 t)$
 + $a_{-1} (\cos(-\omega_0 t) + j \sin(-\omega_0 t))$
 ↳ $a_{-1} (\cos \omega_0 t - j \sin \omega_0 t)$

$a_{\pm 1}$ 
 $a_{\pm 2}$  higher freq.
 $a_{\pm 3}$ 

our goal is to represent
 $x(t) = \sum_{k=-\infty}^{\infty} a_k e^{jk\omega_0 t}$ (A)

↑ k Interpretation is that as g get the coeff. of things that wiggle more & more (Synthesize $x(t)$)?
 Fourier Series

How to compute $\{a_k\}$ for a given $x(t)$?

It stands for reason that for most signals that we care about the amount of wiggling that we do take into account as k gets bigger & bigger at some pts these coeff. get probably smaller & smaller because a real world signal prob. doesn't wiggle like crazy fast.
 We just need to here certain sums of coeffs that we need to add up.

↳

We're gonna assume initially that this property holds

multiply b/s by $e^{-jn\omega_0 t}$

$$x(t) \cdot e^{-jn\omega_0 t} = \sum_{k=-\infty}^{\infty} a_k e^{jk\omega_0 t} \cdot e^{-jn\omega_0 t}$$

(n is a fixed integer)

$$= \sum_{k=-\infty}^{\infty} a_k e^{j(k-n)\omega_0 t}$$

(collect these 2 terms together)

integrate b/s from 0 to T

$$\int_0^T x(t) e^{-jn\omega_0 t} dt = \int_0^T \sum_{k=-\infty}^{\infty} a_k e^{j(k-n)\omega_0 t} dt$$

$$= \sum_{k=-\infty}^{\infty} a_k \int_0^T e^{j(k-n)\omega_0 t} dt$$

(switch the order of sum & integral) well behaved

now,

$$\int_0^T e^{j(k-n)\omega_0 t} dt = \int_0^T \cos(k-n)\omega_0 t dt + j \int_0^T \sin(k-n)\omega_0 t dt$$

now, k and n are integers

• say $k=n$

$$= \int_0^T 1 dt = T$$

Diff. to visualize comp - integral just to make a bit easier

∴ for $k=n$, the integral is T

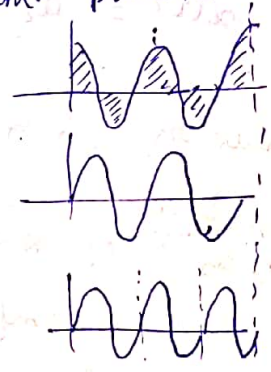
• say $k \neq n$

$$\int_0^T \cos(\text{integer})\omega_0 t dt + j \sin(\text{integer})T$$

oscillates an eq. no. of times inside $[0, T] \Rightarrow$ integral is 0

thus, $\int_0^T x(t) e^{-jn\omega_0 t} dt = a_n T$

Rem. picture



∴ $a_n = \frac{1}{T} \int_0^T x(t) e^{-jn\omega_0 t} dt$ Analysis eqn.

$a_0 = \frac{1}{T} \int_0^T x(t) dt \rightarrow$ DC or Avg.

How do I get a sp. coeff.

$\{a_k\}$ are the Fourier series coefficient of $x(t)$ (spectral coeff.)

(in gen. $x(t)$ can be a complex valued signal) & a_k are complex

→ what if $x(t)$ is real?
 $\{a_k\}$ are ~~not~~ complex, but there're patterns

$$x(t) = \sum_{k=-\infty}^{\infty} a_k e^{jk\omega_0 t}$$

complex conjugate,

$$x^*(t) = \sum_{k=-\infty}^{\infty} a_k^* e^{-jk\omega_0 t}$$

If z is real $z = z^*$

$$= \sum_{k=-\infty}^{\infty} a_{-k}^* e^{jk\omega_0 t}$$

$$= \sum_{k=-\infty}^{\infty} a_{-k}^* e^{jk\omega_0 t}$$

reorder numbers
 change the use of variables
 for comparison

$$\boxed{a_k = a_{-k}^*} \quad \therefore x(t) = x^*(t)$$

So, $a_1 = 1 + 2j$

$a_{-1} = 1 - 2j$

\therefore keep track of +ve a_k & I can immed-check what -ve a_k are

for $x(t)$ real, we can also write

$$x(t) = a_0 + \sum_{k=1}^{\infty} 2 A_k \cos(\theta_k + k\omega_0 t)$$

\uparrow real
 \uparrow amplitude \uparrow phase

(the intuition is that I should be able to make that signal out of a bunch of real cosines or sinusoids) eg. to get a sine wave we need to shift the signal to get 0 at $t=0$
 These are computable from the $\{a_k\}$

Alternatively (for no phase shift)

$$x(t) = a_0 + 2 \sum_{k=1}^{\infty} B_k \cos k\omega_0 t - C_k \sin k\omega_0 t$$

\uparrow real \uparrow real

a bunch of non shifted sines & cosines (related to a_k) (only for real)

Fast Fourier Transform (complex exponentials)

$$x(t) = 5 + 2 \cos(\omega_0 t)$$

$$a_k = \frac{1}{T} \int_0^T x(t) e^{-jk\omega_0 t} dt$$

note when u've something i.e strictly visibly figure a sum of cosines, one can immediately figure out what the Fourier series is since

$$x(t) = 5 + 2 \left(\frac{1}{2} (e^{j\omega_0 t} + e^{-j\omega_0 t}) \right)$$

$$= 5 + e^{j\omega_0 t} + e^{-j\omega_0 t}$$

$$\downarrow \begin{matrix} a_0 e^{j0\omega_0 t} \\ a_1 e^{j1\omega_0 t} \\ a_{-1} e^{-j1\omega_0 t} \end{matrix} + \dots$$

$$a_0 = 5$$

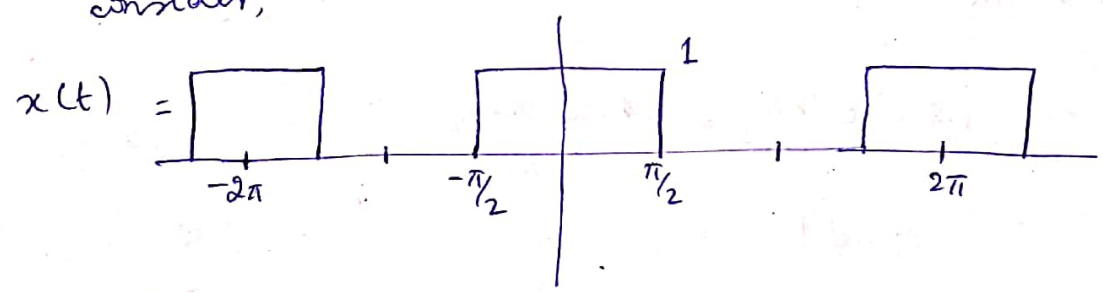
$$a_1 = 1$$

$$a_{-1} = -1$$

(signal is real; they complex conjugate prop.)

All other $a_k = 0$

lets consider,



Fundamental

$$T = 2\pi, \quad \omega_0 = \frac{2\pi}{T} = 1$$

$$DC \text{ term } a_0 = \frac{1}{T} \int_0^T x(t) e^{-j0\omega_0 t} dt = \frac{1}{T} \int_0^T x(t) dt$$

= Average value of the signal over one period

$$a_0 = \frac{1}{2\pi} \int_{-\pi}^{\pi} x(t) dt = \frac{1}{2\pi} \int_{-\pi/2}^{\pi/2} 1 dt = \frac{\pi}{2\pi} = \frac{1}{2}$$

$$= \frac{1}{2\pi} \int_0^{2\pi} x(t) dt$$

(same stuff inside) simpler single piece

$$a_k = \frac{1}{T} \int_{-\pi/2}^{\pi/2} x(t) e^{-jk\omega t} dt$$

$$= \frac{1}{2\pi} \int_{-\pi}^{\pi} x(t) e^{-jkt} dt$$

$$= \frac{1}{2\pi} \int_{-\pi/2}^{\pi/2} e^{-jkt} dt$$

$$= \frac{1}{2\pi} \cdot \frac{-1}{jk} \cdot e^{-jkt} \Big|_{-\pi/2}^{\pi/2}$$

$$= \frac{-1}{2\pi jk} (e^{-jk\pi/2} - e^{jk\pi/2})$$

$$= \frac{1}{\pi k} \left(\frac{e^{jk\pi/2} - e^{-jk\pi/2}}{2j} \right) = \frac{\sin k\pi/2}{\pi k}$$

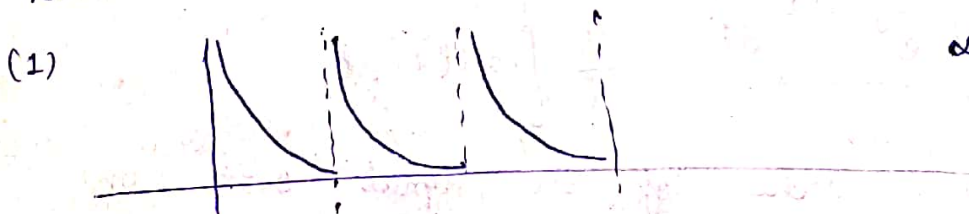
$$a_1 = \frac{\sin \pi/2}{\pi} = \frac{1}{\pi} ; a_{-1} = a_1^* = \frac{1}{\pi} , a_2 = \frac{\sin \pi}{2\pi} = 0$$

$$\text{sinc funcn.} = \frac{\sin x}{x}$$

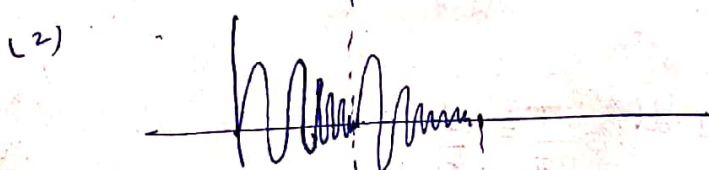
$$\frac{1}{2} \text{sinc } k\pi/2 = \frac{1}{2} \frac{\sin k\pi/2}{k\pi}$$

Fourier Series Applet → falstad.com

→ notes & Properties of the Fourier series :-
 • when doesn't it work?



∞ area under the curve



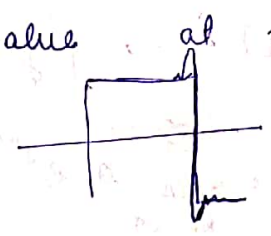
∞ wiggling



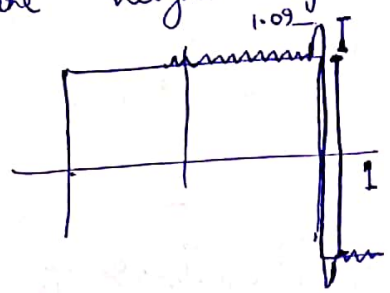
∞ discontinuity

What happens at a discontinuity?

- (a) F.S converges at every continuous pt.
- (b) F.S converges to the avg. value at every discontinuity (say square wave)



GIBBS Phenomenon never get the overshoot below $\approx 9\%$ of the height of discontinuity



Properties:-

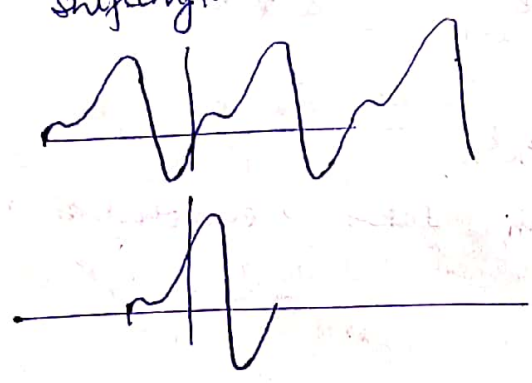
We've $x(t)$ periodic with period T
 F.S $\{a_k\}$ $x(t) \leftrightarrow a_k$

① Linearity

$x(t) \leftrightarrow \{a_k\}$
 $y(t) \leftrightarrow \{b_k\}$ have same period T

$$\alpha x(t) + \beta y(t) \leftrightarrow \{\alpha a_k + \beta b_k\}$$

② Time Shifting:-

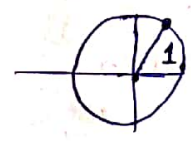


Let $x(t) \leftrightarrow \{a_k\}$

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

$$y(t) \leftrightarrow a_k e^{-jkw_0 t_0}$$

ie a phase shift of F.S coeff. & the magnitude doesn't change
 $|b_k| = |a_k|$



mag. is always 1 (or same)

③ Differentiation

$$x(t) \leftrightarrow \{a_k\}$$

$$x'(t) \leftrightarrow \{jk\omega_0 a_k\}$$

$$x(t) = \sum_{k=-\infty}^{\infty} a_k e^{jk\omega_0 t}$$

$$x'(t) = \sum_{k=-\infty}^{\infty} (a_k \cdot jk\omega_0) e^{jk\omega_0 t}$$

Triangular wave $\xrightarrow{\text{diff}}$ square wave

④ Parseval's theorem

$$\frac{1}{T} \int_0^T |x(t)|^2 dt = \sum_{k=-\infty}^{\infty} |a_k|^2$$

power signal \rightarrow Energy ∞
 • exist over ∞ time
 • periodic

energy signal \rightarrow power is 0
 • time limited
 • non-periodic

avg. power of the signal = power of F.S coeff
 (same amount of power in time as in freq. domain)
 or no inform. lost by converting

⑤ Convolution

$$x(t) \leftrightarrow \{a_k\}$$

$$y(t) \leftrightarrow \{b_k\}$$

$$x(t) \cdot y(t) \leftrightarrow \sum_{l=-\infty}^{\infty} a_l \cdot b_{k-l} = a * b$$

or

$$\int_0^T x(\tau) \cdot y(t-\tau) d\tau \leftrightarrow T a_k b_k$$

multiplication in one domain takes convolution in another domain

(mail me for any questions: aaqi072@gmail.com)

Department of Electrical Engineering
National Institute of Technology Srinagar

Tutorial I

Course Title: Digital Signal processing

Semester: Sixth (6th)

Date: 23.04.2020

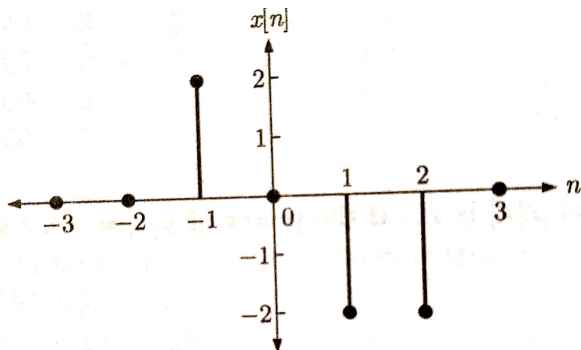
Course Code: ELE-605

I. Discrete-Time Fourier Transform

Q.1) Find the even and odd parts of the following signals:

1. $x[n] = (6, 4\uparrow, 2, 2)$
2. $x[n] = (-4, 5, 1, -2\uparrow, -3, 0, 2)$
3. $x[n] = a^{|n|}$
4. $x[n] = na^n u[n]$

Q.2) Consider a signal $x[n]$ as shown in the figure below



1. If $x[n]$ is transformed into $y[n] = \frac{2}{3}x[-n-2] - 2$, $y[n]$ is
2. What is $y[n] = x[-n/3]$

Q.3) Determine whether or not each of the following sequences is periodic. If your answer is yes, determine the period.

1. $x[n] = A \cos\left(\frac{3\pi}{7}n - \frac{\pi}{8}\right)$
2. $x[n] = e^{j\left(\frac{n}{8} - \pi\right)}$

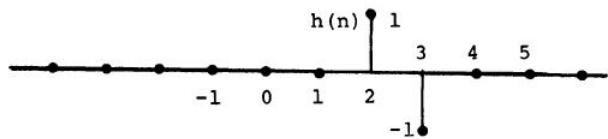
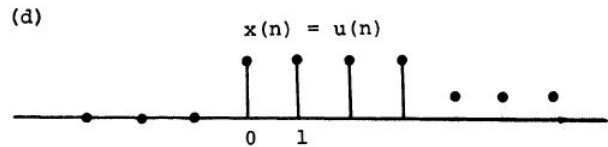
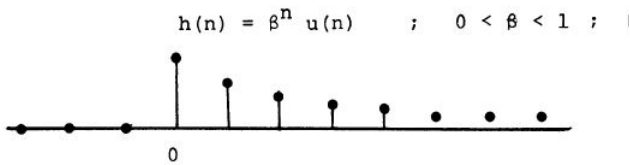
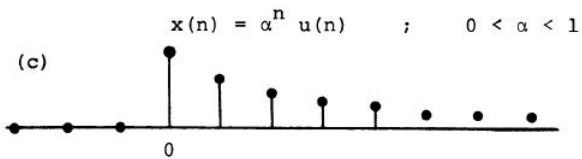
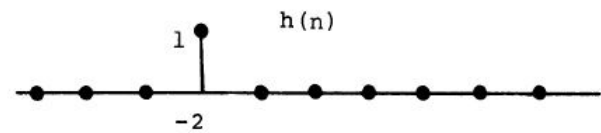
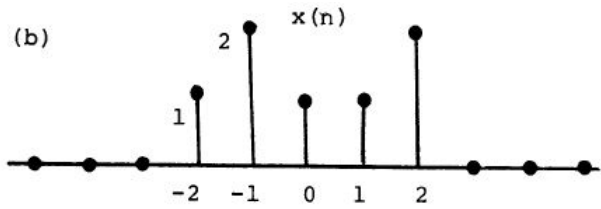
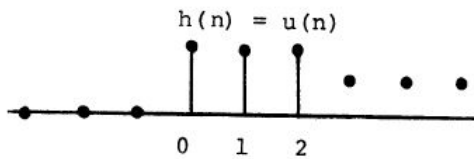
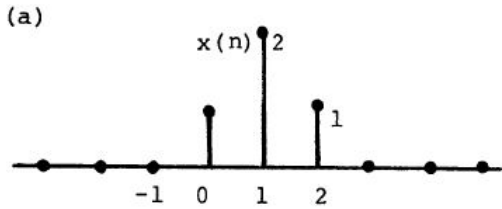
Q.4) For each of the following systems, $y(n)$ denotes the output and $x(n)$ the input. Determine for each whether the specified input-output relationship is linear, shift-invariant and causal.

1. $y[n] = 2x[n] + 3$
2. $y[n] = x[n] \sin\left(\frac{2\pi}{7}n + 6\right)$

3. $y[n] = (x[n])^2$

4. $y[n] = \sum_{m=-\infty}^n x[m]$

Q.5) For each of the following pairs of sequences, $x(n)$ represents the input to an LTI system with unit-sample response $h(n)$. Determine each output $y(n)$. Sketch your results.



Q.6) The system shown below contains two LTI subsystems with unit sample responses $h_1(n)$ and $h_2(n)$, in cascade. Consider $x[n]$ as a unit step.

