



BMS INSTITUTE OF TECHNOLOGY AND MANAGEMENT

YELAHANKA – BANGALORE - 64

DEPARTMENT OF ELECTRONICS & TELECOMMUNICATION ENGINEERING

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ENGINEERING

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DISCRETE FOURIER TRANSFORMS (DFT)

Frequency domain sampling & reconstruction of discrete time signals. DFT as a linear transformation, its relationship with other transforms. 6HN

Introduction to DSP

* A signal is a function i.e. used to describe an observed physical variable of a physical process.

* It is an abstract mathematical description of an observation.
eg. Image signal, ECG (electrocardiogram) ^{about condition of heart}, speech signal, EEG (electroencephalogram) ^{analysis of brain activity}, its mathematical description consists of infinite no. of sinusoids of diff. freq's & can be written

$$s(t) = \sum_{i=-\infty}^{\infty} A_i(t) \sin[\underbrace{\omega_i(t)}_{\text{freq}} + \underbrace{\theta_i(t)}_{\text{phase angle}}]$$

* System: A system may also be defined as a physical device that performs an operation on a signal.

eg. speech signals are generated by forcing air thro' the vocal cords.
input system consists of vocal cords & vocal tract (vocal cavity) 1

eg. filter: used to reduce noise & interference
It performs some ~~the~~ operation(s) on the signal which has the effect of reducing (filtering) noise & interference from the desired information bearing signal

Signal Processing: deals with manipulation or modification of signal so that it results in more desirable or interpretable form.
(or)

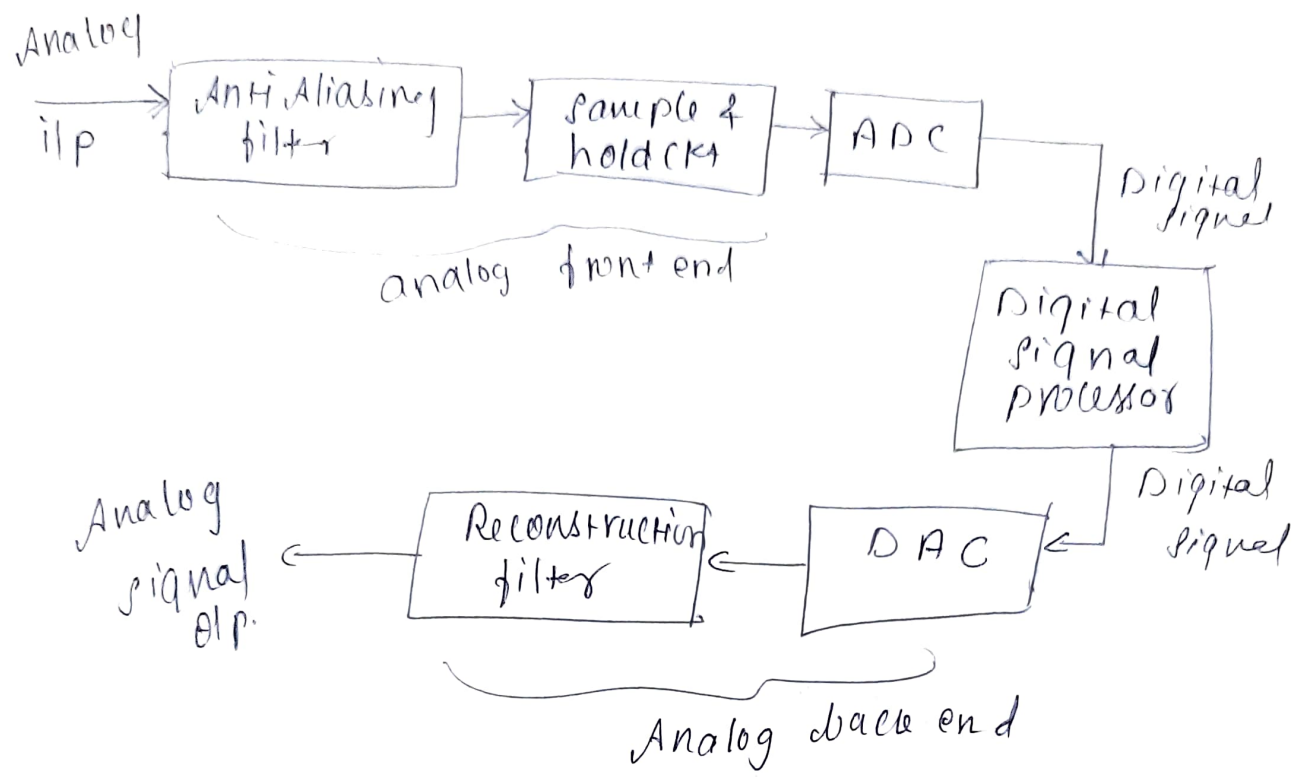
The action of changing one or more features (parameters) of a signal according to a predetermined requirements

- * Parameters \rightarrow freq, amplitude, phase etc
- * The signal which undergoes such a process is known as input signal
- * The entity which performs this processing is system
- * The processed signal is o/p signal

eg of signal processing: Amplification, filtering, modn, etc
Analog
All these uses transistor, FETs, op-amp.

* A platform for signal processing can be analog system, where we can design an analog system using design technique, test & redesign until an acceptable performance is met

* There is another approach called Digital signal processing where the infoⁿ is represented in digital format.



* The anti aliasing filter, an analog Lpf is used to band limit the i/p analog signal to the required freq ~~component~~ range & prevent freq components beyond this range from appearing as aliases in the sampled spectrum of the i/p signal

* The sample & hold ckt → discretizes the analog i/p signal w.r.t time
∴ the analog i/p signal is picked up at discrete intervals in time

* The ADC maps various binary value to each sample appearing at the i/p chosen from a set of finite values. we get a digital signal discretized both w.r.t time & amplitude 3

* The digital signal so obtained is processed by employing DSP techniques & the o/p is another seqn of binary no's which is converted to analog form using

DAC

* The LPE at o/p removes the undesired high freq noise & gives out the desired analog signal

Advantages of Digital over Analog signal processing

1) digital programmable system allows flexibility in reconfiguring the digital signal processing ops simply by changing the pgms

Reconfiguration of an analog system usually implies a redesign of the HW followed by testing & verification to see that it operates properly

2) digital system provides much better control of accuracy requirements

3) digital signals are easily stored on magnetic media (tape or disk) w/o loss of signal

4) DSP method also allows for the implementation of more sophisticated signal processing algorithms

- (5) It is difficult to perform precise mathematical operations on signals in analog form but the same operations can be implemented on a digital computer using slw's
- (6) digital implementation of signal processing system is cheaper than its analog.
- (7) Adaptable to low freq signal processing where analog processors would require very large passive elements like inductors & capacitors

Limitation

- * system complexity pres. ∴ need of conversion of real-time analog signals to digital & processed digital-signal back to analog
- * Limited BW due to constraint on sampling rate.

Applications

- * consumer electronic applns - TV, music synthesizer,

1. The Discrete Fourier Transform

Its properties & Applications

- * Frequency analysis of discrete-time signals is usually & most conveniently performed on a digital signal processor [i.e. may be a general purpose digital computer or specially designed digital HW].
- * To perform frequency analysis on a discrete-time signal $\{x(n)\}$, we convert the time-domain sequence to an equivalent freq-domain representation. Such a representation is given by the Fourier transform $X(\omega)$ of the sequence $\{x(n)\}$.
- * However, $X(\omega)$ is a ~~constant~~ continuous fun of freq & ∞ it is not a computationally convenient representation of the sequence $\{x(n)\}$.
- * Here we consider the representation of a sequence $\{x(n)\}$ by samples of its spectrum $X(\omega)$. Such a freq-domain representation leads to the discrete Fourier transform [DFT], which is a powerful computational tool for performing discrete-time signals freq analysis of

(ref. Proakis)

1.1 Frequency Domain Sampling: The Discrete Fourier Transform

Before we introduce DFT, we consider the sampling of the FT of an aperiodic discrete-time seqⁿ. Thus we establish the relationship bet' the sampled FT & DFT.

Frequency Domain Sampling & Reconstruction of Discrete-Time Signals

Let us consider an aperiodic discrete-time signal $x(n)$ with Fourier transform

$$X(\omega) = \sum_{n=-\infty}^{\infty} x(n) e^{-j\omega n} \rightarrow \textcircled{1}$$

$x(n) \rightarrow$ discrete-time signals
 $\omega \rightarrow$ freq & is continuous fun from 0 to 2π .

* This means that $x(n)$ is discrete & its spectrum $X(\omega)$ is continuous. Such a continuous fun^s cannot be evaluated on a digital processor. Since only ~~digit~~ discrete signals can be evaluated.

* So to overcome this problem of digital processing, the spectrum $X(\omega)$ is sampled uniformly (periodic) in freq at a spacing of ω radians bet' successive samples with period of 2π .

* Since $X(\omega)$ is periodic with period of 2π , the samples are taken from 0 to 2π & the spacing bet' successive samples will be $\Delta\omega = \frac{2\pi}{N}$. For convenience we take N equidistant samples in the interval $0 \leq \omega < 2\pi$ with spacing samples in the interval

∴ substituting $\omega = \frac{2\pi}{N}k$ in eq (1), we get (2) (4)

$$X\left(\frac{2\pi}{N}k\right) = \sum_{n=-\infty}^{\infty} x(n) e^{-j\frac{2\pi}{N}kn}$$

where $k = 0, 1, 2, \dots, N-1$ → (2)

n = index of samples.

& $x(\omega)$ is calculated only at discrete values.

* for eq let $N=8$. samples are taken over a period of 2π & $x(\omega)$ will be calculated at

$$\omega = 0, \frac{\pi}{4}, \frac{\pi}{2}, \frac{3\pi}{4}, \pi, \frac{5\pi}{4}, \frac{3\pi}{4}, \frac{7\pi}{4}$$

& the samples are addressed as

$$k = 0, 1, 2, \dots, 7$$

by substituting these values in eq (2), the sampled spectrum obtained is shown in fig (1.1) below.

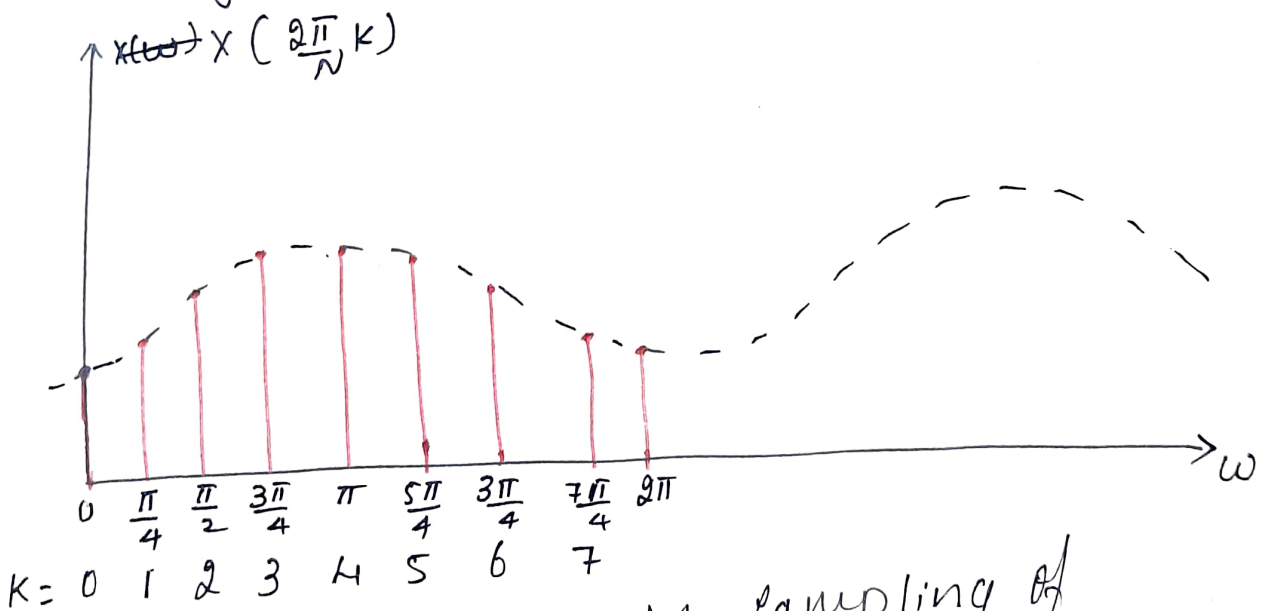


fig (1.1) freq - domain sampling of the fourier-transforms

from eq (2), 'n' varies from $-\infty$ to $+\infty$, let us divide this summation into individual summations containing only 'N' samples of $x(n)$

i.e.,

$$X\left(\frac{2\pi}{N}k\right) = \dots + \sum_{n=-N}^{-1} x(n) e^{-j2\pi kn/N} + \sum_{n=0}^{N-1} x(n) e^{-j2\pi kn/N} + \sum_{n=N}^{2N-1} x(n) e^{-j2\pi kn/N} + \dots$$

The above individual summations can be represented as

$$X\left(\frac{2\pi}{N}k\right) = \sum_{l=-\infty}^{\infty} \sum_{n=lN}^{lN+N-1} x(n) e^{-j2\pi kn/N}$$

Let us change $n \rightarrow n - lN$ of the inner summation hence the limits will be at $n=0$ to $N-1$

$$X\left(\frac{2\pi}{N}k\right) = \sum_{l=-\infty}^{\infty} \sum_{n=0}^{N-1} x(n-lN) e^{-j2\pi k(n-lN)/N}$$

$$= \sum_{l=-\infty}^{\infty} \sum_{n=0}^{N-1} x(n-lN) e^{-j2\pi kn/N} e^{j2\pi kl}$$

here $e^{j2\pi kl} = 1$ always

$$X\left(\frac{2\pi}{N}k\right) = \sum_{l=-\infty}^{\infty} \sum_{n=0}^{N-1} x(n-lN) e^{-j2\pi kn/N}$$

Let us interchange the order of the summation. we obtain,

$$\begin{aligned} n &= n - lN \\ &= n - lN - lN \\ &= n - 2lN - lN \end{aligned}$$

'n = lN

n = n - lN

= lN - lN

n = 0

(ii) lN + N - 1

n = n - lN

= lN + N - 1

+ -lN

N - 1

$$\begin{aligned} e^{+0} &= \cos 0 + j \sin 0 \\ e^{+2\pi k} &= \cos 2\pi k + j \sin 2\pi k \\ &= 1 + 0 \end{aligned}$$

(3) (5)

$$X\left(\frac{2\pi}{N}k\right) = \sum_{n=0}^{N-1} \left[\sum_{l=-\infty}^{\infty} x(n-lN) \right] e^{-j\frac{2\pi}{N}kn} \quad \text{---> (3)}$$

$K = 0 \dots N-1$

$$= \sum_{n=0}^{N-1} x_p(n) e^{-j\frac{2\pi}{N}kn} \quad \text{---> (4)}$$

here $k = 0, 1, \dots, N-1$

$$x_p(n) = \sum_{l=-\infty}^{\infty} x(n-lN) \quad \text{---> (4.1)}$$

$$\dots + x(n+2N) + x(n+N) + x(n) + x(n-N) + x(n-2N) + \dots$$

This means that $x_p(n)$ is a periodic repetition of $x(n)$ with the period of 'N' samples.

* Let us consider some arbitrary non-periodic signal $x(n)$. It contains 'L' samples & is shown in fig (1.20a)

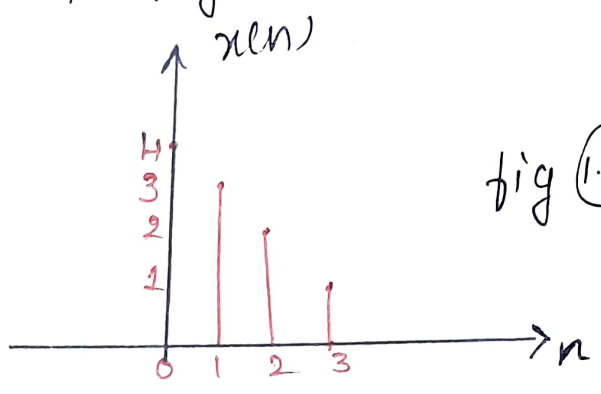


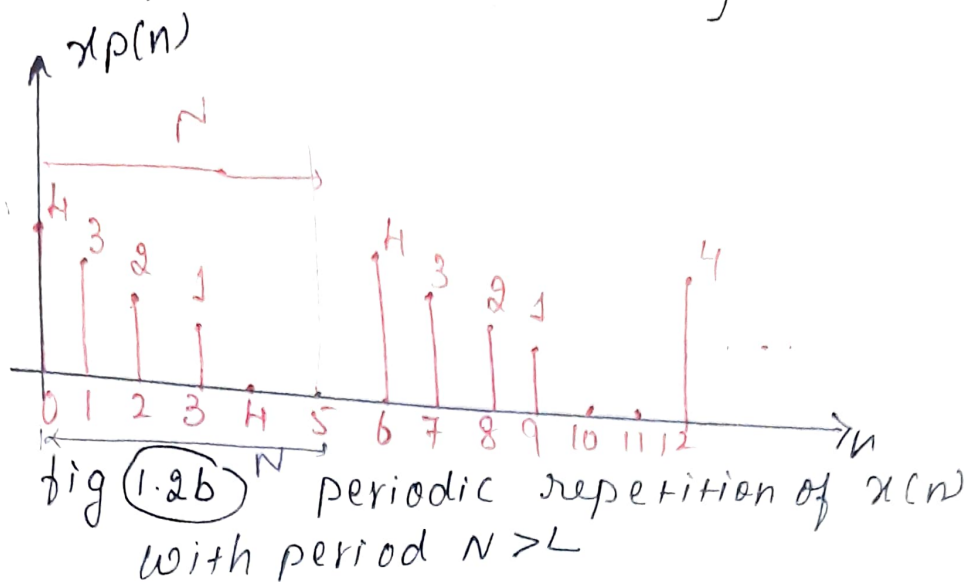
fig (1.20a) An original signal $x(n)$ of length 'L'

case-1

(i) $N > L$, NO Aliasing.

Now let us prepare the signal $x_p(n)$ which is obtained by periodic repetition of $x(n)$. Let the period 'N' be greater than 'L'. For eg. let $N=6$ & the $x_p(n)$ is shown in fig (1.20b). here at $n=4$ & 5 , the sample values are zero.

because the signal repeats at $n=6, 12, \dots$ etc & hence for $N > L$, there is no Aliasing



case (ii): $N < L$, Aliasing
 Now let us consider $N < L$, for eg $N=3$
 the w/f is shown in fig (1.2c).

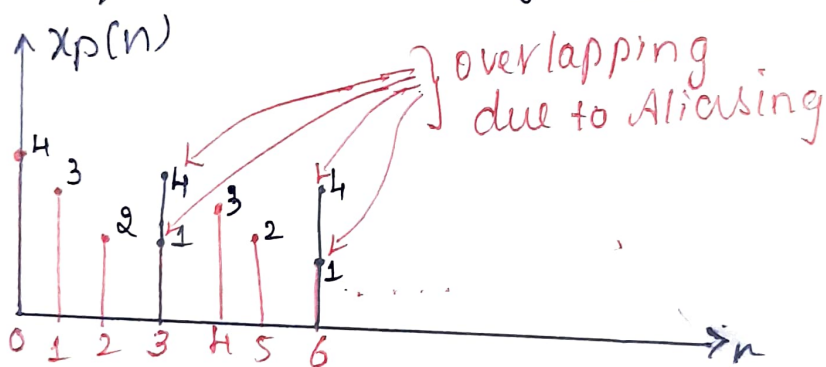


fig (1.2c) periodic repetition of $x(n)$ with period $N < L$.

here from the w/f we observe that since $N < L$, the 2 samples are overlapping at $n=3, 6, \dots$ etc. This is Aliasing hence it isn't possible to obtain $x(n)$ from $x_p(n)$.

∴ To avoid Aliasing in time-domain, the no of samples in freq spectrum must be greater than no of samples in time-domain
 i.e. $N \geq L$.

Reconstruction:

WKT $x_p(n)$ is periodic with period 'N'
It can be expressed by discrete Fourier series as

$$x_p(n) = \sum_{k=0}^{N-1} C_k e^{j\frac{2\pi}{N}kn} \rightarrow (5)$$

where $n = 0, 1, \dots, N-1$

C_k = Fourier coefficient & can be expressed as

$$C_k = \frac{1}{N} \sum_{n=0}^{N-1} x_p(n) e^{-j\frac{2\pi}{N}kn} \rightarrow (6)$$

$k = 0, 1, \dots, N-1$

WKT from eq (4) $X\left(\frac{2\pi}{N}k\right)$ is

$$X\left(\frac{2\pi}{N}k\right) = \sum_{n=0}^{N-1} x_p(n) e^{-j\frac{2\pi}{N}kn} \rightarrow (4)$$

\therefore Comparing eq (4) & (6) we get

$$C(k) = \frac{1}{N} X\left(\frac{2\pi}{N}k\right) \rightarrow (7)$$

Substituting (7) in (5)

$$x_p(n) = \sum_{k=0}^{N-1} \frac{1}{N} X\left(\frac{2\pi}{N}k\right) e^{j\frac{2\pi}{N}kn} \rightarrow (8)$$

$n = 0, 1, \dots, N-1$

This eqn (8) provides the reconstruction of periodic signal $x_p(n)$ from the samples of spectrum $X(\omega)$

- * However it doesn't imply that we can recover $x(n)$ or $X(\omega)$ from the same
- * to accomplish this we need to consider the relationship betⁿ $x_p(n)$ + $x(n)$
- * since $x_p(n)$ is periodic extension of $x(n)$ as in eq (4.1) i.e,
$$x_p(n) = \sum_{d=-\infty}^{\infty} x(n - dN)$$

* It is clear that $x(n)$ can be recovered from $x_p(n)$ if there is no aliasing in time domain, as illustrated in the fig 1.2.

* ~~from the fig we observe when~~ we consider a finite-duration seqⁿ $x(n)$ which is non zero in the interval $0 \leq n \leq L-1$

* we observe that when $N \geq L$
 $x(n) = x_p(n)$, $0 \leq n \leq N-1$

so that $x(n)$ can be recovered from $x_p(n)$ without ambiguity

* if $N < L$ it is not possible to recover $x(n)$ from its periodic extension due to time domain aliasing.

* Thus we conclude that the spectrum of an aperiodic discrete-time signal with finite duration L can be exactly recovered from its samples at frequencies

$$\omega_k = \frac{2\pi k}{N} \quad \text{if } N \geq L.$$

$$x(n) = \begin{cases} x_p(n), & 0 \leq n \leq N-1 \\ 0, & \text{elsewhere} \end{cases} \rightarrow \textcircled{9}$$

& finally $X(\omega)$ can be computed eq (1)

$$\text{as } X(\omega) = \sum_{n=-\infty}^{\infty} x(n) e^{-j\omega n}$$

for $X(\omega)$

1.2 Discrete Fourier Transform [DFT]

If $x(n)$ has a finite duration of length $L \leq N$, then $x_p(n)$ is periodic repetition of $x(n)$

$$\text{where } x_p(n) = \begin{cases} x(n), & 0 \leq n \leq L-1 \\ 0, & L \leq n \leq N-1 \end{cases}$$

[defined for one period of N]

* consider eq (1) i.e.

$$X\left(\frac{2\pi}{N}k\right) = \sum_{n=0}^{N-1} x_p(n) e^{-j\frac{2\pi}{N}kn}$$

from fig 1.2b we have shown that if no of samples in $x(n)$ are less than N then there is no aliasing. If we calculate the above eq for $x(n)$, then we can write

$$X\left(\frac{2\pi}{N}k\right) = \sum_{n=0}^{L-1} x(n) e^{-j\frac{2\pi}{N}kn} \rightarrow \textcircled{9}$$

WKT $N > L$, to avoid aliasing in time domain, hence the upper limit of the summation can be made as $N-1$.

∴ eq ⑨ becomes

$$\text{DFT: } X(k) = X\left(\frac{2\pi}{N}k\right) = \sum_{n=0}^{N-1} x(n) e^{-j\frac{2\pi}{N}kn} \rightarrow \textcircled{10}$$

$$k = 0, 1, \dots, N-1$$

Note: $X\left(\frac{2\pi}{N}k\right)$ is written as $X(k)$. ∴ the values of $X\left(\frac{2\pi}{N}k\right)$ are addressed by 'k' only. eq ⑩ is called as Discrete Fourier Transform [DFT]

Now consider eq (8)

$$x_p(n) = \sum_{k=0}^{N-1} \frac{1}{N} x\left(\frac{2\pi k}{N}\right) e^{j\frac{2\pi k}{N}n}$$

if we evaluate the above eqⁿ for $n=0, 1, \dots, N-1$
then $x_p(n) = x(n)$

$$\text{IDFT: } x(n) = \frac{1}{N} \sum_{k=0}^{N-1} X(k) e^{j\frac{2\pi k}{N}n}$$

$$n = 0, 1, \dots, N-1$$

→ (11)

The above eqⁿ gives the original seq $x(n)$ from its DFT. Hence it is called as Inverse Discrete Fourier Transforms [IDFT].

DFT as a linear transformation: -

let us define $W_N = e^{-j\frac{2\pi}{N}}$ → (12)

= Twiddle factor.

hence DFT & IDFT eqⁿ can be written as

$$X(k) = \sum_{n=0}^{N-1} x(n) W_N^{kn}, \quad k=0, 1, \dots, N-1 \rightarrow (13)$$

$$\& \quad x(n) = \frac{1}{N} \sum_{k=0}^{N-1} X(k) W_N^{-kn}, \quad n=0, 1, \dots, N-1 \rightarrow (14)$$

So eq (13) & (14) can be denoted symbolically as follows:

$$x(n) = \frac{1}{N} \sum_{k=0}^{N-1} X(k) W_N^{-kn} \xleftarrow{\text{DFT}} X(k) = \sum_{n=0}^{N-1} x(n) W_N^{kn}$$

Let us represent seq $x(n)$ as vector x_N of 'N' samples i.e.

$$x_N = \begin{matrix} n=0 \\ n=1 \\ \vdots \\ n=N-1 \end{matrix} \begin{bmatrix} x(0) \\ x(1) \\ \vdots \\ x(N-1) \end{bmatrix} \quad \rightarrow \quad (15)$$

$N \times 1$

$X(k)$ can be represented as a vector X_N of 'N' samples i.e.

$$X_N = \begin{matrix} k=0 \\ k=1 \\ \vdots \\ k=N-1 \end{matrix} \begin{bmatrix} X(0) \\ X(1) \\ \vdots \\ X(N-1) \end{bmatrix} \quad \rightarrow \quad (16)$$

$N \times 1$

The values W_N^{kn} can be represented as a matrix $[W_N]$ of size $N \times N$ as follows:

$$[W_N] = \begin{matrix} W_N^{kn} \\ W_N^{kn} \\ W_N^{kn} \\ \vdots \\ W_N^{kn} \end{matrix} \begin{matrix} / k=0, n=0 \\ / k=1, n=0 \\ / k=2, n=0 \\ \\ / k=N-1, n=0 \end{matrix} = \begin{matrix} W_N^0 \\ W_N^0 \\ W_N^0 \\ \\ W_N^0 \end{matrix}$$

$$[W_N] = \begin{matrix} & \underline{n=0} & & \underline{n=1} & \\ \begin{matrix} k=0 \\ k=1 \\ k=2 \\ \vdots \\ k=N-1 \end{matrix} & \left[\begin{matrix} W_N^{kn} / k=0, n=0 \\ W_N^{kn} / k=1, n=0 \\ W_N^{kn} / k=2, n=0 \\ \vdots \\ W_N^{kn} / k=N-1, n=0 \end{matrix} \right] & = & W_N^0 & \left[\begin{matrix} W_N^{kn} / k=0, n=1 \\ W_N^{kn} / k=1, n=1 \\ W_N^{kn} / k=2, n=1 \\ \vdots \\ W_N^{kn} / k=N-1, n=1 \end{matrix} \right] & = & W_N^0 \end{matrix}$$

$$\left. \begin{matrix} W_N^{kn} / k=0, n=2 = W_N^0 \dots \dots W_N^{kn} / k=0, n=N-1 = W_N^0 \\ W_N^{kn} / k=1, n=2 = W_N^2 \dots \dots W_N^{kn} / k=1, n=N-1 = W_N^{N-1} \\ W_N^{kn} / k=2, n=2 = W_N^4 \dots \dots W_N^{kn} / k=2, n=N-1 = W_N^{2(N-1)} \\ \vdots \\ W_N^{kn} / k=N-1, n=2 = W_N^{2(N-1)} \dots \dots W_N^{kn} / k=N-1, n=N-1 = W_N^{(N-1)(N-1)} \end{matrix} \right\} N \times N$$

$$[W_N] = \begin{bmatrix} W_N^0 & W_N^0 & W_N^0 & \dots & W_N^0 \\ W_N^0 & W_N^1 & W_N^2 & \dots & W_N^{N-1} \\ W_N^0 & W_N^2 & W_N^4 & \dots & W_N^{2(N-1)} \\ \vdots & \vdots & \vdots & \dots & \vdots \\ W_N^0 & W_N^{N-1} & W_N^{2(N-1)} & \dots & W_N^{(N-1)(N-1)} \end{bmatrix} \rightarrow 17$$

here the individual elements are written as W_N^{kn} with 'k' rows & 'n' columns

Then N -point DFT of eq (13) can be represented in the matrix form as

$$X_N = [W_N] x_N \rightarrow (18)$$

Similarly, IDFT of eq (14) can be represented in the matrix form as

$$x_N = \frac{1}{N} [W_N^*] X_N \rightarrow (19)$$

where $W_N^* \rightarrow$ complex conjugate of W_N .

~~Comparing eqns (18) & (19)~~

~~Let~~
 * We observe that W_N is a symmetric matrix, if we assume that inverse of W_N exists then eq (18) can be written as

$$x_N = W_N^{-1} X_N \rightarrow (20)$$

* comparing eq (19) & (20),

$$W_N^{-1} = \frac{1}{N} W_N^*$$

$$\frac{1}{W_N} = \frac{1}{N} W_N^*$$

$$(or) \boxed{W_N W_N^* = N \cdot I_N} \rightarrow (21)$$

where I_N is an $N \times N$ identity matrix

Let us see the values of W_N for the foll values:

(i) $N=8$

Let $W_N = e^{-j\frac{2\pi}{N}}$

with $N=8$ $W_N = e^{-j\frac{2\pi}{8}} = e^{-j\frac{\pi}{4}}$

then $W_N^k = e^{-j\frac{2\pi}{N}k}$

$W_8^k = e^{-j\frac{2\pi}{8}k}$

$= e^{-j\frac{\pi}{4}k}$ where $k: 0, 1, 2, \dots, N-1$

(i) $k=0$ $W_8^0 = e^0 = 1$

(ii) $k=1$ $W_8^1 = e^{-j\frac{\pi}{4}} = \cos\frac{\pi}{4} - j\sin\frac{\pi}{4} = \frac{1}{\sqrt{2}} - j\frac{1}{\sqrt{2}}$

(iii) $k=2$ $W_8^2 = e^{-j\frac{2\pi}{4}} = \cos\frac{\pi}{2} - j\sin\frac{\pi}{2} = 0 - j1 = -j$

(iv) $k=3$ $W_8^3 = e^{-j\frac{3\pi}{4}} = \cos\frac{3\pi}{4} - j\sin\frac{3\pi}{4} = -\frac{1}{\sqrt{2}} - j\frac{1}{\sqrt{2}}$

(v) $k=4$ $W_8^4 = e^{-j\frac{4\pi}{4}} = \cos\pi - j\sin\pi = -1$

(vi) $k=5$ $W_8^5 = e^{-j\frac{5\pi}{4}} = \cos\frac{5\pi}{4} - j\sin\frac{5\pi}{4} = -\frac{1}{\sqrt{2}} + j\frac{1}{\sqrt{2}}$

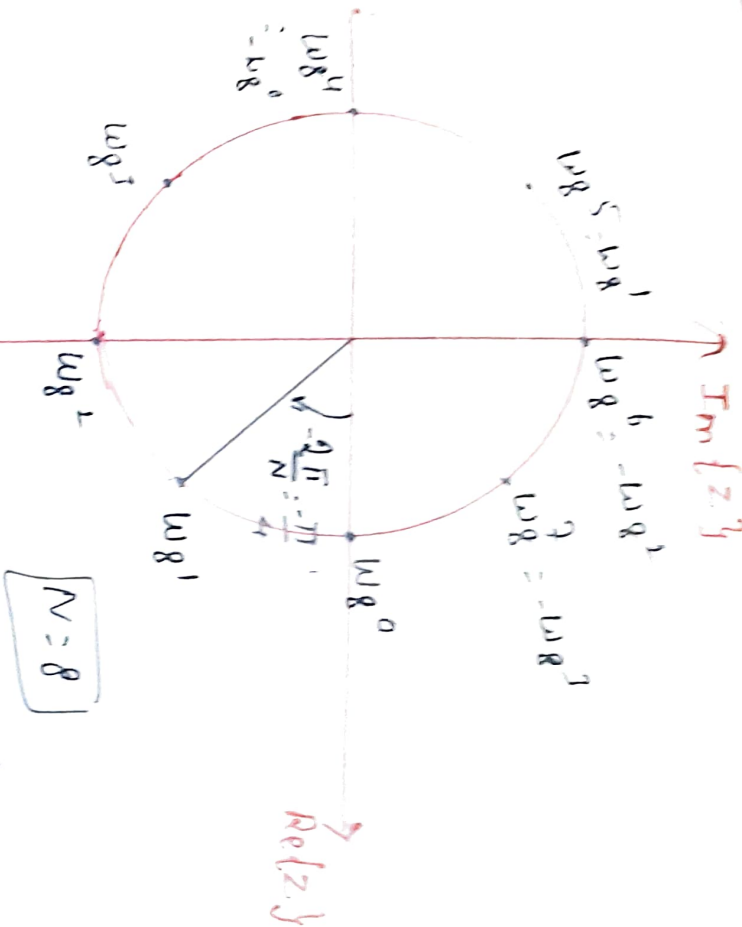
$W_8^5 = -W_8^1$

(vii) $k=6$ $W_8^6 = e^{-j\frac{6\pi}{4}} = e^{-j\frac{3\pi}{2}} = \cos\frac{3\pi}{2} + j\sin\frac{3\pi}{2}$

$W_8^6 = -W_8^2$

(viii) $k=7$ $W_8^7 = e^{-j\frac{7\pi}{4}} = \cos\frac{7\pi}{4} - j\sin\frac{7\pi}{4} = \frac{1}{\sqrt{2}} + j\frac{1}{\sqrt{2}}$

$W_8^7 = W_8^3$



the fig above shows the phasors in complex plane
against unit circle

(11)

$$\frac{N=4}{N^k} = W_4^k \quad k=0,1,2,3$$

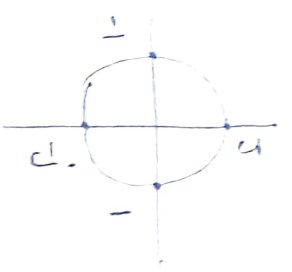
$$W_N^k = e^{-j\frac{2\pi}{N}k} \quad W_4^k = e^{-j\frac{2\pi}{4}k} = e^{-j\frac{\pi}{2}k}$$

$$k=0 \quad W_4^0 = e^0 = 1$$

$$k=1 \quad W_4^1 = e^{-j\frac{\pi}{2}} = \cos \frac{\pi}{2} - j \sin \frac{\pi}{2} = -j$$

$$k=2 \quad W_4^2 = e^{-j\pi} = \cos \pi - j \sin \pi = -1$$

$$k=3 \quad W_4^3 = e^{-j\frac{3\pi}{2}} = \cos \frac{3\pi}{2} - j \sin \frac{3\pi}{2} = +j$$



DFT of some standard signals.

(1) compute the N-point DFT for the foll signal

(a) $x(n) = \delta(n)$ [unit sample]

(Ans)

Soln:- The unit sample $\delta(n)$ is given as

$$x(n) = \begin{cases} 1, & n=0 \\ 0, & n \neq 0 \end{cases}$$

DFT is given by

$$X(k) \triangleq \sum_{n=0}^{N-1} x(n) e^{-j \frac{2\pi}{N} kn}$$

substituting for $x(n)$

$$X(k) = x(0) e^0 = 1$$

compute the N-point DFT of foll signals
 $x(n) = \delta(n)$
 $x(n) = \delta(n-n_0)$
 $x(n) =$

(b) $x(n) = \delta(n-n_0)$

$$X(k) \triangleq \sum_{n=0}^{N-1} x(n) e^{-j \frac{2\pi}{N} kn}$$

$$= \sum_{n=0}^{N-1} \delta(n-n_0) e^{-j \frac{2\pi}{N} kn}$$

$$= \delta(n_0) e^{-j \frac{2\pi}{N} kn_0}$$

$\delta(n-n_0) = 1; n=n_0$
 $= 0, n \neq n_0$

$$= e^{-j \frac{2\pi}{N} kn_0}$$

~~(c) $x(n) = a^n$~~

~~(c) a~~

③ $x(n) = a^n$ for $0 \leq n \leq N-1$

$$X(k) \triangleq \sum_{n=0}^{N-1} x(n) e^{-j\frac{2\pi}{N}kn}$$

$$= \sum_{n=0}^{N-1} a^n e^{-j\frac{2\pi}{N}kn}$$

$$= \sum_{n=0}^{N-1} [a e^{-j\frac{2\pi}{N}k}]^n$$

$$\left[\sum_{n=0}^{N-1} \alpha^n = \frac{1-\alpha^N}{1-\alpha}, \alpha \neq 1 \right]$$

$\omega_k T$

$$= N, \alpha = 1$$

$$\sum_{n=N_1}^{N_2} a^n = \frac{a^{N_1} - a^{N_2+1}}{1-a}$$

$$X(k) = \frac{1 - [a e^{-j\frac{2\pi}{N}k}]^N}{1 - a e^{-j\frac{2\pi}{N}k}} = \frac{1 - a^N e^{-j2\pi k}}{1 - a e^{-j\frac{2\pi}{N}k}}$$

$$e^{-j2\pi k} = \cos 2\pi k - j \sin 2\pi k$$

= 1 always

$$X(k) = \frac{1 - a^N}{1 - a e^{-j\frac{2\pi}{N}k}}$$

→

$$\textcircled{d} \quad x(n) = e^{j\frac{2\pi}{N}k_0n}, \quad 0 \leq n \leq N-1$$

$$X(k) \triangleq \sum_{n=0}^{N-1} x(n) e^{-j\frac{2\pi}{N}kn}$$

$$= \sum_{n=0}^{N-1} e^{j\frac{2\pi}{N}k_0n} e^{-j\frac{2\pi}{N}kn}$$

$$= \sum_{n=0}^{N-1} \left(e^{j\frac{2\pi}{N}(k-k_0)n} \right)^n$$

$$= \frac{1 - \left(e^{j\frac{2\pi}{N}(k-k_0)} \right)^N}{1 - e^{j\frac{2\pi}{N}(k-k_0)}}$$

$$1 - e^{j\frac{2\pi}{N}(k-k_0)}$$

$$= \frac{1 - e^{j\frac{2\pi}{N}(k-k_0)}}{1 - e^{j\frac{2\pi}{N}(k-k_0)}}$$

$$1 - e^{j\frac{2\pi}{N}(k-k_0)}$$

When $k \neq k_0$

$$= \frac{1 - 1}{1 - e^{j\frac{2\pi}{N}(k-k_0)}} = 0$$

$$1 - e^{j\frac{2\pi}{N}k} e^{-j\frac{2\pi}{N}k_0}$$

When $k = k_0$

$$X(k) = \sum_{n=0}^{N-1} \left(e^{j\frac{2\pi}{N}(k_0-k_0)n} \right)^n$$

$$= \sum_{n=0}^{N-1} 1 = N$$

$$X(k) = \begin{cases} 0, & k \neq m \\ N, & k = m \end{cases} = N \cdot \delta(k-m)$$

Q2) compute 8-point DFT for the foll seqn.:

$$x(n) = \{1, 1, 1, 1, 0, 0, 0, 0\}$$

Soln:

$$W_N^k \triangleq e^{-j\frac{2\pi}{N}kn}$$

since $N=8$. $W_8^k = e^{-j\frac{\pi}{4}k}$ $k=0, 1, 2, \dots, 7$

$$W_8^0 = 1, \quad W_8^1 = \frac{1}{\sqrt{2}} - j\frac{1}{\sqrt{2}}, \quad W_8^2 = -j, \quad W_8^3 = -\frac{1}{\sqrt{2}} - j\frac{1}{\sqrt{2}}$$

$$W_8^4 = -1, \quad W_8^5 = -\frac{1}{\sqrt{2}} + j\frac{1}{\sqrt{2}}, \quad W_8^6 = j, \quad W_8^7 = \frac{1}{\sqrt{2}} + j\frac{1}{\sqrt{2}}$$

$$DFT\{x(n)\} = X(k) \triangleq \sum_{n=0}^{N-1} x(n) W_N^{kn}$$

$$k=0, 1, 2, \dots, N-1$$

$$\therefore X(k) = \sum_{n=0}^7 x(n) W_8^{kn}, \quad 0 \leq k \leq 7$$

$$X(k) = 1 \cdot W_8^0 + 1 \cdot W_8^k + 1 \cdot W_8^{2k} + 1 \cdot W_8^{3k}$$

$$= 1 + W_8^k + W_8^{2k} + W_8^{3k}$$

$$X(0) = 1 + 1 + 1 + 1 = \boxed{4}$$

$$X(1) = 1 + W_8^1 + W_8^2 + W_8^3 = 1 + \frac{1}{\sqrt{2}} - j\frac{1}{\sqrt{2}} - j - \frac{1}{\sqrt{2}} - j\frac{1}{\sqrt{2}}$$

$$= 1 - j\frac{1}{\sqrt{2}} - j - j\frac{1}{\sqrt{2}}$$

$$\boxed{X(1) = 1 - j2.41}$$

$$X(2) = 1 + W_8^2 + W_8^4 + W_8^6$$

$$= 1 - j - 1 + j = \boxed{0}$$

$$X(3) = 1 + W_8^3 + W_8^6 + W_8^9$$

$$= 1 + W_8^3 + W_8^6 + W_8^1$$

$$= 1 - \frac{1}{\sqrt{2}} - j\frac{1}{\sqrt{2}} + j + \frac{1}{\sqrt{2}} - j\frac{1}{\sqrt{2}} = \boxed{1 - j0.41}$$

X(4) = 1 + w8^4 + w8^0 + w8^4 = 1 - 1 + 1 - 1 = 0

X(5) = 1 + w8^5 + w8^2 + w8^7 = 1 - 1/√2 + j1/√2 - j + 1/√2 + j1/√2

X(6) = 1 + w8^6 + w8^4 + w8^2 = 1 + j0.41

X(7) = 1 + w8^7 + w8^6 + w8^5 = 1 + j2.41

X(k) = { 4, 1 - j2.41, 0, 1 - j0.41, 0, 1 + j0.41, 0, 1 + j2.41 }

3 find the 4-point DFT of the seq^n given below x(n) = { 1, 0, 1, 0 } using x(k) found above find its inverse using the defining eq^n.

Soln:

N = 4, wN = e^-j2π/N

w4 = e^-jπ/2

w4^0 = 1, w4^1 = -j, w4^2 = -1, w4^3 = j

DFT{x(n)} = X(k) = Σ_{n=0}^{N-1} x(n) wN^{kn}

= Σ_{n=0}^3 x(n) w4^{kn} 0 ≤ k ≤ 3

= x(0)w4^0 + x(1)w4^{1k} + x(2)w4^{2k} + x(3)w4^{3k} = 1 + w4^{2k}

X(0) = 1 + 1 = 2

X(1) = 1 + w4^2 = 1 - 1 = 0

X(2) = 1 + w4^4 = 1 + w4^0 = 1 + 1 = 2

X(3) = 1 + w4^6 = 1 + w4^2 = 1 - 1 = 0

X(k) = { 2, 0, 2, 0 }

$$\text{Note: } \omega_N^{-kn} = [\omega_N^{kn}]^*$$

$$\therefore \omega_4^{-0} = (\omega_4^0)^* = 1, \quad \omega_4^{-1} = [\omega_4^1]^* = -j$$

$$\omega_4^{-2} = [\omega_4^2]^* = -1, \quad \omega_4^{-3} = [\omega_4^3]^* = j$$

$$x(n) = \text{IDFT}\{X(k)\} = \frac{1}{N} \sum_{k=0}^{N-1} X(k) \omega_N^{-kn}$$

$$x(n) = \frac{1}{4} \sum_{k=0}^3 X(k) \omega_4^{-kn} \quad n = 0, 1, 2, 3$$

$$= \frac{1}{4} \left[2 + 0 \omega_4^{-n} + 2 \times \omega_4^{-2n} + 0 \times \omega_4^{-3n} \right]$$

$$= \frac{1}{4} \left[2 + 2 \omega_4^{-2n} \right] = \frac{1}{2} \left[1 + \omega_4^{-2n} \right]$$

$$x(0) = \frac{1}{2} \left[1 + \omega_4^0 \right] = 1$$

$$x(1) = \frac{1}{2} \left[1 + \omega_4^{-2} \right] = 0$$

$$x(2) = \frac{1}{2} \left[1 + \omega_4^0 \right] = 1$$

$$x(3) = \frac{1}{2} \left[1 + \omega_4^{-2} \right] = 0$$

$$x(n) = \{1, 0, 1, 0\}$$

Q4 Compute 4-point DFT of the seqn using matrix method. $x(n) = \{0, 1, 2, 3\}$

Soln:

here $N=4$ evaluate w_4^k where $k=0, 1, 2, 3$

$$w_4^0 = 1, w_4^1 = -j, w_4^2 = -1 \text{ \& } w_4^3 = j$$

$$x_N = x_4 = \begin{bmatrix} 0 \\ 1 \\ 2 \\ 3 \end{bmatrix}$$

$$X_N = ?$$

$$[W_N]_{N \times N} = [W_4] = \begin{bmatrix} w_4^0 & w_4^0 & w_4^0 & w_4^0 \\ w_4^0 & w_4^1 & w_4^2 & w_4^3 \\ w_4^0 & w_4^2 & w_4^4 & w_4^6 \\ w_4^0 & w_4^3 & w_4^6 & w_4^9 \end{bmatrix}_{4 \times 4}$$

$$w_4^0 = w_4^4 = w_4^8$$

$$w_4^1 = w_4^5 = w_4^9$$

$$w_4^2 = w_4^6 = w_4^{10}$$

$$w_4^3 = w_4^7 = w_4^{11}$$

$$= \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & -j & -1 & j \\ 1 & -1 & 1 & 1 \\ 1 & j & -1 & -j \end{bmatrix}$$

NOW WKT

$$\text{DFT } X_N \triangleq [W_N] x_N$$

$$X_4 = [W_4] x_4$$

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

$$X_4 = \begin{bmatrix} 0+1+2+3 \\ 0-j-2+3j \\ 0-1+2-j \\ 0+j-2-3j \end{bmatrix} = \begin{bmatrix} 6 \\ -2+2j \\ -2 \\ -2-2j \end{bmatrix}$$

② Calculate 8-point DFT of the seqⁿ

$$x(n) = \{1, 1, 1, 1\}$$

$$X(K) = \begin{bmatrix} 4 \\ 1-j(1+\sqrt{2}) \\ 0 \\ 1+j(1-\sqrt{2}) \\ 0 \\ 1-j(1-\sqrt{2}) \\ 0 \\ 1+j(1+\sqrt{2}) \end{bmatrix}$$

⑤ find 4-point DFT of the seqⁿ

$$x(n) = \cos\left(\frac{n\pi}{4}\right)$$

The 1st four samples of $x(n)$ can be obtained by putting $n=0, 1, 2, 3$

$$x(0) = \cos(0) = 1, \quad x(2) = \cos\left(\frac{2 \cdot \pi}{4}\right) = 0$$

$$x(1) = \cos(\pi/4) = 0.707, \quad x(3) = \cos\left(\frac{3 \cdot \pi}{4}\right) = -0.707$$

The DFT of this seqⁿ is

$$X(K) = \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & -j & -1 & j \\ 1 & -1 & 1 & -1 \\ 1 & j & -1 & -j \end{bmatrix} \begin{bmatrix} 1 \\ 0.707 \\ 0 \\ -0.707 \end{bmatrix}$$

$$= \{1, 1-j1.414, 1, 1+j1.414\}$$

Relationship of DFT to other transforms:-

(i) Relationship to the Fourier transform:
of a non-periodic sequence

The FT of a non periodic seqⁿ $x(n)$ having length 'N' is given by

$$X(e^{j\omega}) = \sum_{n=0}^{N-1} x(n) e^{-j\omega n} \rightarrow (1)$$

where $X(e^{j\omega})$ is a continuous funⁿ of ω .

The discrete FT of $x(n)$ is given by

$$X(k) = \sum_{n=0}^{N-1} x(n) e^{-j\frac{2\pi}{N}kn} \rightarrow (2)$$

$k = 0, 1, 2, \dots, N-1$

comparing eq (1) & (2) we find that DFT of $x(n)$ is sampled version of the FT of the seqⁿ & is given by

$X(k) = X(e^{j\omega}) \Big|_{\omega = \frac{2\pi}{N}k}$

 $\rightarrow (3)$

$k = 0, 1, 2, \dots, N-1$

(ii) Relationship to the Z-transform:-

Let us consider a seqⁿ $x(n)$ of the finite duration 'N' with Z-transform.

$$X(z) = \sum_{n=-\infty}^{\infty} x(n) z^{-n} \rightarrow (5)$$

with ROC that includes the unit circle if $X(z)$ is sampled at 'N' equally spaced points on the unit circle

$Z_k = e^{-j\frac{2\pi}{N}k}$;

$$x(k) = X(z) \Big|_{z = e^{-j\frac{2\pi}{N}k}}$$

$$= \sum_{n: +\infty}^{+\infty} x(n) e^{-j\frac{2\pi}{N}kn} \rightarrow (2)$$

eq (2) is identical to the FT $X(e^{j\omega})$ evaluated at N equally spaced frequencies

$$\omega_k = \frac{2\pi}{N}k, \quad 0 \leq k \leq N-1$$

let us consider a seqⁿ $x(n)$ of finite durⁿ 'N' with Z-transform.

$$X(z) = \sum_{n=0}^{N-1} x(n) z^{-n} \rightarrow (3)$$

we have

$$x(n) = \frac{1}{N} \sum_{k=0}^{N-1} x(k) e^{j\frac{2\pi}{N}kn} \rightarrow (4)$$

$$, 0 \leq n \leq N-1$$

substituting eq (4) in eq (3)

$$X(z) = \sum_{n=0}^{N-1} \left[\frac{1}{N} \sum_{k=0}^{N-1} x(k) e^{j\frac{2\pi}{N}kn} \right] z^{-n}$$

$$= \frac{1}{N} \sum_{k=0}^{N-1} x(k) \sum_{n=0}^{N-1} \left[e^{j\frac{2\pi}{N}k} z^{-1} \right]^n$$

$$= \frac{1}{N} \sum_{k=0}^{N-1} x(k) \begin{cases} 1 - a^N \\ 1 - a, a \neq 1 \\ = N, a = 1 \end{cases}$$

Not used

$$\begin{aligned}
 X(z) &= \frac{1}{N} \sum_{k=0}^{N-1} X(k) \left[\frac{1 - e^{j\frac{2\pi}{N}k \cdot N} z^{-N}}{1 - e^{j\frac{2\pi}{N}k} z^{-1}} \right] \\
 &= \frac{1}{N} \sum_{k=0}^{N-1} X(k) \left[\frac{1 - z^{-N}}{1 - e^{j\frac{2\pi}{N}k} z^{-1}} \right] \quad [e^{j2\pi k} = 1] \\
 &= \frac{1 - z^{-N}}{N} \sum_{k=0}^{N-1} \frac{X(k)}{1 - e^{j\frac{2\pi}{N}k} z^{-1}} \quad \rightarrow (5)
 \end{aligned}$$

if eq (5) is evaluated on a unit circle at 'N' equally spaced points [i.e, $z_k = e^{j\frac{2\pi}{N}k}$; $0 \leq k \leq N-1$]

we get the FT of the finite duration seq $x(n)$ in terms of its DFT

$$X(e^{j\omega}) = \frac{1 - e^{-j\omega N}}{N} \sum_{k=0}^{N-1} \frac{X(k)}{1 - e^{-j(\omega - \frac{2\pi}{N}k)}}$$

(13) Relation to the Fourier series coefficients of an aperiodic sequence $\rightarrow (6)$

A periodic seqⁿ $x_p(n)$ with fundamental period N can be represented in a F.S as

$$x_p(n) = \sum_{k=0}^{N-1} C_k e^{j\frac{2\pi}{N}nk} \quad \rightarrow (1) \quad -\infty < n < \infty$$

where the Fourier series coefficients are given by the

$$C_k = \frac{1}{N} \sum_{n=0}^{N-1} x_p(n) e^{-j\frac{2\pi}{N}kn} \quad \rightarrow (2)$$

comparing eq (1) & (2) with

$$X(k) = \sum_{n=0}^{N-1} x(n) e^{-j\frac{2\pi}{N}kn} \quad \& \quad x(n) = \frac{1}{N} \sum_{k=0}^{N-1} X(k) e^{j\frac{2\pi}{N}kn}$$

$$\boxed{X(k) = N \cdot C_k}$$

Find the IDFT of $x(k) = \{1, 0, 1, 0\}$

$$x(n) = \frac{1}{N} \sum_{k=0}^{N-1} x(k) e^{j\frac{2\pi}{N}kn} ; 0 \leq n \leq N-1$$

Assume $N=4$

$$x(n) = \frac{1}{4} \sum_{k=0}^3 x(k) e^{j\frac{2\pi}{4}kn} \quad 0 \leq n \leq 3$$

$$= e^{j\frac{\pi}{2}kn}$$

$$= \frac{1}{4} \left[x(0) + x(1) e^{j\frac{\pi}{2}n} + x(2) e^{j\pi n} + x(3) e^{j\frac{3\pi}{2}n} \right]$$

$$= \frac{1}{4} [x(0) + x(2) e^{j\pi n}]$$

for $n=0$

$$x(0) = \frac{1}{4} [1 + 0 + 1 + 0] = \frac{2}{4} = 0.5$$

$$x(1) = \frac{1}{4} [1 + 1 e^{j\pi}]$$

$$e^{j\pi} = \cos \pi + j \sin \pi$$

$$= (-1) + 0$$

$$= -1$$

$$= \frac{1}{4} [1 + 1(-1)] = \frac{1}{4} [1-1] = 0$$

$$x(2) = \frac{1}{4} [1 + x(2) e^{j2\pi}]$$

$$e^{j2\pi} = \cos 2\pi + j \sin 2\pi$$

$$= \frac{1}{4} [1 + 1 e^{j2\pi}] = \frac{1}{4} [1+1] = \frac{2}{4} = 0.5$$

$$= 1 + 0$$

$$x(3) = \frac{1}{4} [1 + 1 e^{j3\pi}] = \frac{1}{4} [1-1] = 0$$

$$e^{j3\pi} = \cos 3\pi + j \sin 3\pi$$

$$= -1 - 0$$

$$= -1$$

(89) Find the N-point DFT of the seqⁿ

$$x(n) = \begin{cases} 1 & ; n = \text{even} \\ 0 & ; n = \text{odd} \end{cases} ; 0 \leq n \leq N-1$$

? $N = \text{odd}$

$$X(K) = \sum_{n=0}^{N-1} x(n) e^{-j\frac{2\pi}{N}Kn} ; 0 \leq K \leq N-1$$

$$= 1 + x(2) e^{-j\frac{2\pi}{N}K \cdot 2} + x(4) e^{-j\frac{2\pi}{N}K \cdot 4} + \dots + x(N-1) e^{-j\frac{2\pi}{N}K(N-1)}$$

$$= \sum_{n=0}^{\frac{N-1}{2}} \left(e^{-j\frac{2\pi}{N}K \cdot 2} \right)^m$$

WKT $\sum_{n=N_1}^{N_2} a^n = \frac{a^{N_1} - a^{N_2+1}}{1-a}$

$$N_1 = 0 \quad N_2 = e^{-j\frac{2\pi}{N}K \cdot 2} \quad \therefore$$

$$N_2 + 1 = e^{-j\frac{4\pi}{N}K}$$

Wkt $N_1 = 0 \quad N_2 = \frac{N-1}{2} \quad \therefore N_2 + 1 = \frac{N-1}{2} + 1 = \frac{N-1+2}{2} = \frac{N+1}{2}$

$$X(K) = \frac{1 - \left(e^{-j\frac{2\pi}{N}K \cdot 2} \right)^{\frac{N+1}{2}}}{1 - e^{-j\frac{4\pi}{N}K}}$$

$$= \frac{1 - e^{-j\frac{4\pi}{N}K}}{1 - e^{-j\frac{4\pi}{N}K}}$$

$$= \frac{1 - e^{-j\frac{2\pi}{N}k}}{1 - e^{-j\frac{4\pi}{N}k}}$$

$$e^{-j\frac{2\pi}{N}k} \cdot e^{-j\frac{2\pi}{N}k} = e^{-j\frac{4\pi}{N}k}$$

$$a^2 - b^2 = (a+b)(a-b)$$

$$= \frac{1 - e^{-j\frac{2\pi}{N}k}}{(1 - e^{-j\frac{2\pi}{N}k})(1 + e^{-j\frac{2\pi}{N}k})}$$

$$(1 - e^{-j\frac{2\pi}{N}k})(1 + e^{-j\frac{2\pi}{N}k})$$

9) Find the 4-point DFT of the sequence $x(n) = [1, 0, 0, 1]$ using matrix method & verify the answer by taking the 4-point IDFT of the result.

$$(a) X_N = W_N x_N$$

$$N = 4$$

$$\begin{bmatrix} x(0) \\ x(1) \\ x(2) \\ x(3) \end{bmatrix} = \begin{bmatrix} W_4^0 & W_4^0 & W_4^0 & W_4^0 \\ W_4^0 & W_4^1 & W_4^2 & W_4^3 \\ W_4^0 & W_4^2 & W_4^4 & W_4^6 \\ W_4^0 & W_4^3 & W_4^6 & W_4^9 \end{bmatrix} \begin{bmatrix} X(0) \\ X(1) \\ X(2) \\ X(3) \end{bmatrix}$$

$$= \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & -j & -1 & j \\ 1 & -1 & 1 & -1 \\ 1 & j & -1 & -j \end{bmatrix} \begin{bmatrix} 1 \\ 0 \\ 0 \\ 1 \end{bmatrix} = \begin{bmatrix} 2 \\ 1+j \\ 0 \\ 1-j \end{bmatrix}$$

$$X_N = \frac{1}{N} W_N^k X_N$$

$$\underline{N=4}$$

$$\begin{bmatrix} X(0) \\ X(1) \\ X(2) \\ X(3) \end{bmatrix} = \frac{1}{4} \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & j & -1 & -j \\ 1 & -1 & 1 & -1 \\ 1 & -j & -1 & j \end{bmatrix} \begin{bmatrix} \text{[scribbled out]} \\ \text{[scribbled out]} \\ \text{[scribbled out]} \\ \text{[scribbled out]} \end{bmatrix}$$

$$= \begin{bmatrix} 2 \\ 1+j \\ 0 \\ 1-j \end{bmatrix}$$

$$= \frac{1}{4} \begin{bmatrix} 4 \\ 0 \\ 0 \\ 4 \end{bmatrix} = \begin{bmatrix} 1 \\ 0 \\ 0 \\ 1 \end{bmatrix}$$

Properties of DFT

(1) Periodicity :

If $x(n) \xleftrightarrow[N]{\text{DFT}} X(K)$

then

$$X(K+N) = X(K) \quad \forall K$$
$$X(n+N) = X(n) \quad \forall n$$

The sequences $x(n)$ & $X(K)$ are implicit periodic with a period equal to N

Proof:

WKT $x(n) \triangleq \frac{1}{N} \sum_{k=0}^{N-1} X(K) \omega_N^{-kn}$

Replacing n by $(n+N)$ we get

$$X(n+N) = \frac{1}{N} \sum_{k=0}^{N-1} X(K) \omega_N^{-k(n+N)}$$

$$= \frac{1}{N} \sum_{k=0}^{N-1} X(K) \omega_N^{-kn} \cdot \omega_N^{-kN}$$

WKT $\omega_N^{-kN} = e^{j \frac{2\pi}{N} kN}$
 $= e^{j 2\pi k} = 1$ always

$$X(n+N) = \frac{1}{N} \sum_{k=0}^{N-1} X(K) \omega_N^{-kn}$$

$$X(n+N) = X(n) \rightarrow \textcircled{1}$$

eq (1) \Rightarrow $x(n)$ is implicit periodic with a period $= N$

$$X(k) \triangleq \sum_{n=0}^{N-1} x(n) W_N^{kn}$$

Replace k by $k+N$, we get

$$X(k+N) = \sum_{n=0}^{N-1} x(n) W_N^{(k+N) \cdot n}$$

$$= \sum_{n=0}^{N-1} x(n) W_N^{kn} \cdot W_N^{Nn}$$

$$W_N^{Nn} = e^{-j \frac{2\pi}{N} Nn} = e^{-j2\pi n} = 1 \text{ always}$$

$$X(k+N) = \sum_{n=0}^{N-1} x(n) W_N^{kn}$$

$$X(k+N) = X(k) \rightarrow \textcircled{2}$$

eq (2) \Rightarrow $x(k)$ is implicit periodic with a period $\cdot N$.

(1) find a 4-point DFT of the sequence $x(n) = 2^n$, $0 \leq n \leq 3$. Also find $X(-1)$ & $X(5)$

Soln: given $x(n) = \{1, 2, 4, 8\}$

WKT $W_4^0 = 1$, $W_4^1 = -j$, $W_4^2 = -1$, $W_4^3 = +j$

$$X(k) \triangleq \sum_{n=0}^3 x(n) W_4^{kn}, \quad 0 \leq k \leq 3$$

$$= 1 + 2W_4^k + 4W_4^{2k} + 8W_4^{3k}$$

$$\therefore X(0) = 1 + 2 + 4 + 8 = \boxed{15}$$

$$X(1) = 1 + 2W_4^1 + 4W_4^2 + 8W_4^3 = 1 - 2j - 4 + 8j = \boxed{-3 + 6j}$$

$$x(2) = 1 + 2\omega_4^2 + 4\omega_4^4 + 8\omega_4^6$$

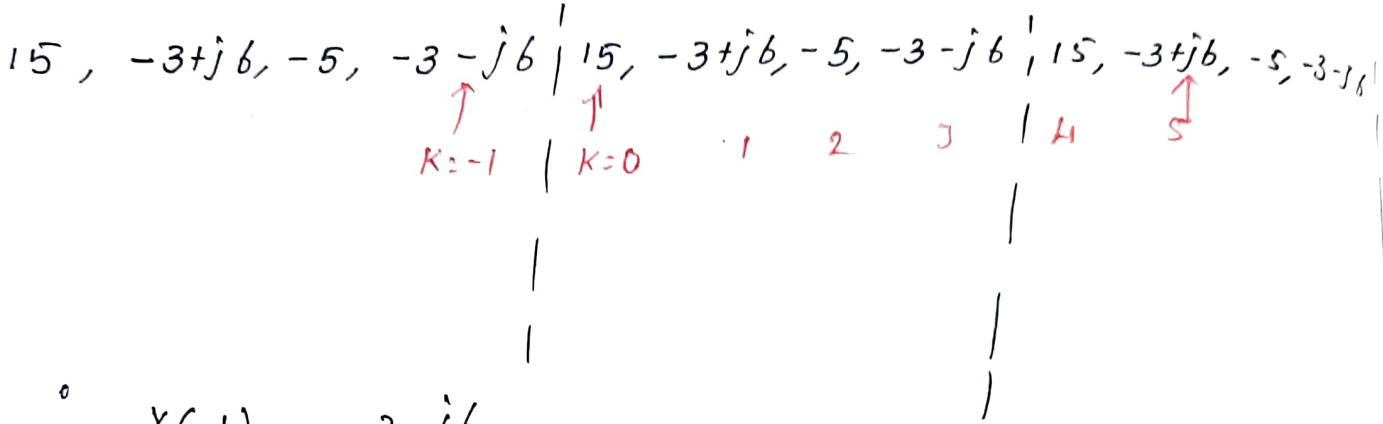
$$= 4\omega_4^0 - 8\omega_4^2$$

$$= 1 - 2 + 4 - 8 = \boxed{-5}$$

$$x(3) = 1 + 2\omega_4^3 + 4\omega_4^6 = 4\omega_4^2 + 8\omega_4^1 = 8\omega_4^1$$

$$= 1 + 2j - 4 - 8j = \boxed{-3 - 6j}$$

Periodic extension of $x(k)$ is



$$x(-1) = -3 - j6$$

$$-x(5) = -3 + j6$$

② Linearity:

The Linearity property of DFT states that:

$$x_1(n) \xleftrightarrow[N]{DFT} X_1(k)$$

$$\& x_2(n) \xleftrightarrow[N]{DFT} X_2(k), \text{ then}$$

$$a_1 x_1(n) + a_2 x_2(n) \xleftrightarrow[N]{DFT} a_1 X_1(k) + a_2 X_2(k)$$

where a_1 & a_2 are constants.

Proof:

By definition of DFT,

$$X(k) \triangleq \sum_{n=0}^{N-1} x(n) \omega_N^{kn} \rightarrow (1)$$

Let $x(n) = a_1 x_1(n) + a_2 x_2(n)$

then eq (1) becomes

$$\begin{aligned}
 X(K) &= \sum_{n=0}^{N-1} [a_1 x_1(n) + a_2 x_2(n)] \omega_N^{Kn} \\
 &= \sum_{n=0}^{N-1} a_1 x_1(n) \omega_N^{Kn} + \sum_{n=0}^{N-1} a_2 x_2(n) \omega_N^{Kn} \\
 &= a_1 \sum_{n=0}^{N-1} x_1(n) \omega_N^{Kn} + a_2 \sum_{n=0}^{N-1} x_2(n) \omega_N^{Kn} \\
 &= a_1 X_1(K) + a_2 X_2(K)
 \end{aligned}$$

② compute 4-point DFT of the seqn $x(n)$ given below using linearity property.

$$x(n) = \cos\left(\frac{\pi}{4}n\right) + j \sin\left(\frac{\pi}{4}n\right), \quad 0 \leq n \leq 3$$

Soln:

Let $x(n) = x_1(n) + jx_2(n)$

where $x_1(n) = \cos\left(\frac{\pi}{4}n\right)$ & $x_2(n) = \sin\left(\frac{\pi}{4}n\right)$

n	$\cos\left(\frac{\pi}{4}n\right)$	$\sin\left(\frac{\pi}{4}n\right)$
0	1	0
1	$1/\sqrt{2}$	$1/\sqrt{2}$
2	0	1
3	$-1/\sqrt{2}$	$1/\sqrt{2}$

to find $X_1(K)$ & $X_2(K)$

$$\omega_4^0 = 1, \quad \omega_4^1 = -j, \quad \omega_4^2 = -1, \quad \omega_4^3 = +j$$

$$X_1(K) = \sum_{n=0}^3 x_1(n) \omega_4^{Kn}$$

$$= 1 + \frac{1}{\sqrt{2}} \omega_4^K - \frac{1}{\sqrt{2}} \omega_4^{3K}$$

$$X_1(0) = 1 + \frac{1}{\sqrt{2}} - \frac{1}{\sqrt{2}} = \boxed{1}, \quad X_1(1) = 1 + \frac{1}{\sqrt{2}} \omega_4^1 - \frac{1}{\sqrt{2}} \omega_4^3$$

$$= 1 - \frac{1}{2}j - \frac{1}{\sqrt{2}}j$$

$$= \boxed{1 - 1.414j}$$

X_2 (or)
using matrix method

$$X_N = [W_N] \cdot x_N$$

$X_1(0)$
 $X_2(0)$

$$= \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & -j & -1 & -j \\ 1 & -1 & 1 & -1 \\ 1 & j & -1 & -j \end{bmatrix} \begin{bmatrix} 1 \\ 0.707 \\ 0 \\ -0.707 \end{bmatrix}$$

$$= \begin{bmatrix} 1 \\ 1 - j1.414 \\ 1 \\ 1 + j1.414 \end{bmatrix}$$

iffy

$$X_2(k) = \sum_{n=0}^3 x_2(n) \omega_4^{kn}, \quad 0 \leq k \leq 3$$

$$X_{2N} = [W_N] \cdot x_N$$

$$= \begin{bmatrix} 1 & 1 & 1 & 1 \\ 1 & -j & -1 & -j \\ 1 & -1 & 1 & -1 \\ 1 & j & -1 & -j \end{bmatrix} \begin{bmatrix} 0 \\ 0.707 \\ 1 \\ 0.707 \end{bmatrix}$$

$$= \begin{bmatrix} 2.414 \\ -j \\ 0.414 \\ -1 \end{bmatrix}$$

$$X(k) = X_1(k) + j X_2(k) \quad k = 0, 1, 2, 3$$

$$= (1 + j2.414) + (1 - j1.414 - j)$$

$$+ (1 - j0.414) + (1 + j1.414 - j1)$$

Circular time shift / (or) circular translation (4) (6)

WKT, N -point DFT of a finite duration sequence $x(n)$ of length $L \leq N$ is equivalent to N -point DFT of a periodic seqⁿ $x_p(n)$ of period N , which is obtained by periodically extending $x(n)$.

$$x_p(n) = \sum_{l=-\infty}^{+\infty} x(n-lN) \rightarrow (1)$$

i.e.,

$$x(n) \xleftrightarrow[N]{\text{DFT}} X(k) \rightarrow (2)$$

$$x_p(n) \xleftrightarrow[N]{\text{DFT}} X(k) \rightarrow (3)$$

eq (2) & (3) are related by eq (1).

& can be rewritten as

$$x(n) = \begin{cases} x_p(n), & 0 \leq n \leq N-1 \\ 0, & \text{E.W} \end{cases} \rightarrow (4)$$

* Let $x_p(n)$ be shifted by 'k' units to right then the new seqⁿ $x'_p(n)$ is

$$x'_p(n) = x_p(n-k) \rightarrow (5)$$

$$= \sum_{l=-\infty}^{\infty} x(n-k-lN) \rightarrow (6)$$

Then the corresponding seqⁿ $x'(n)$ can be obtained from eq (4)

$$x'(n) = \begin{cases} x'_p(n) & 0 \leq n \leq N-1 \\ 0, & \text{e.w} \end{cases} \rightarrow (7)$$

The seqⁿ $x'(n)$ is related to $x(n)$ by the circular shift. This concept is illustrated in fig below. 42

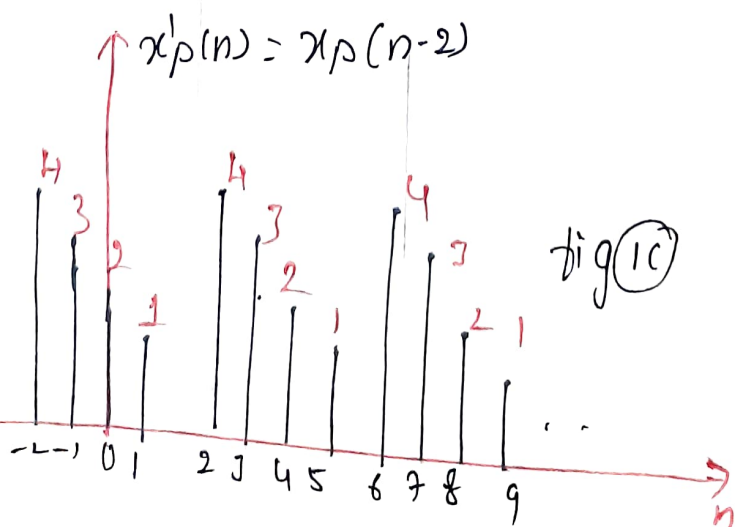
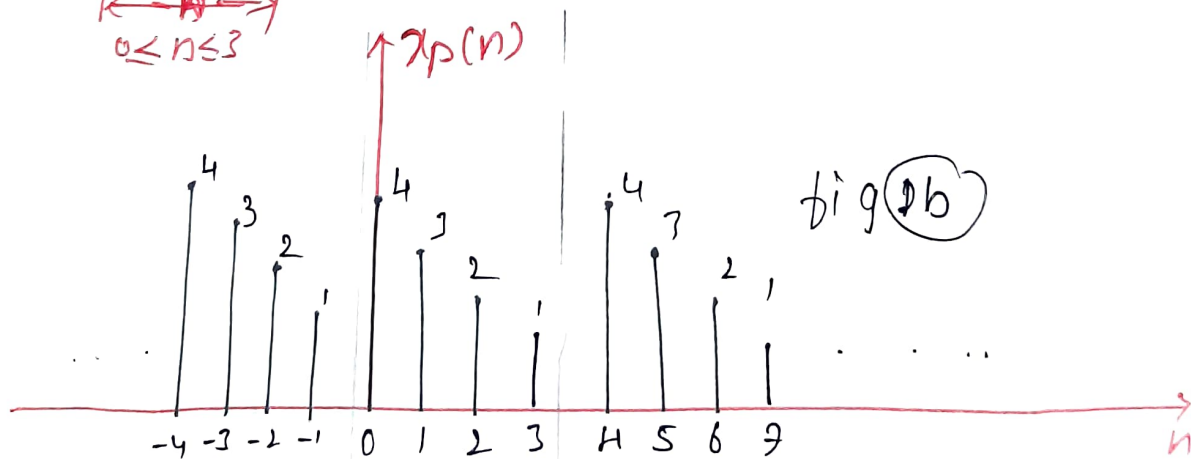
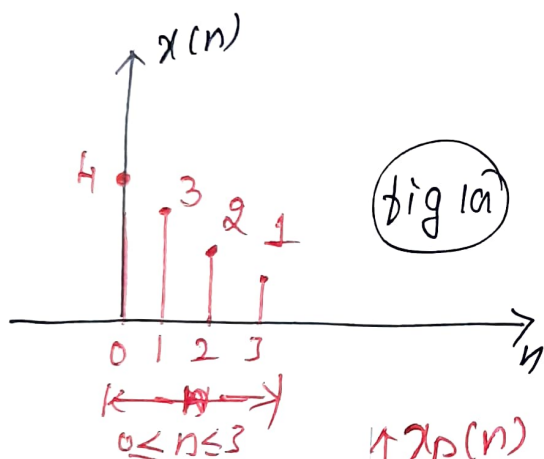
let

$x(n) \rightarrow$ contains 4-samples or $N=4$ & is shown in fig (1a).

$x_p(n) \rightarrow$ periodic repetition of $x(n)$ is shown in fig (1b)

$x'_p(n) \rightarrow$ which is obtained by shifting $x_p(n)$ to right by 2 samples is shown in fig (1c)

$x'_p(n) = x_p(n-2)$, $x(n) \rightarrow$ shown in fig (1a) as per eq (7)



The seqn $x'(n)$ is related to $x(n)$ by a circular shift & is represented as

$$x'(n) = x(n-k, \text{ modulo } N) \rightarrow (8)$$
$$= x((n-k)_N)$$

Let us evaluate $x'(n)$ as per eq (8) with $N=4$ $k=2$

$$x'(n) = x((n-2)_4) = x((n-k)_N)$$

$$\therefore x'(0) = x((0-2)_4) = x((-2)_4) = x(2) = 2$$

" $[x(N-2) = x(4-2) = x(2)]$

$$x'(1) = x((1-2)_4) = x((-1)_4) = x(3) = 1$$

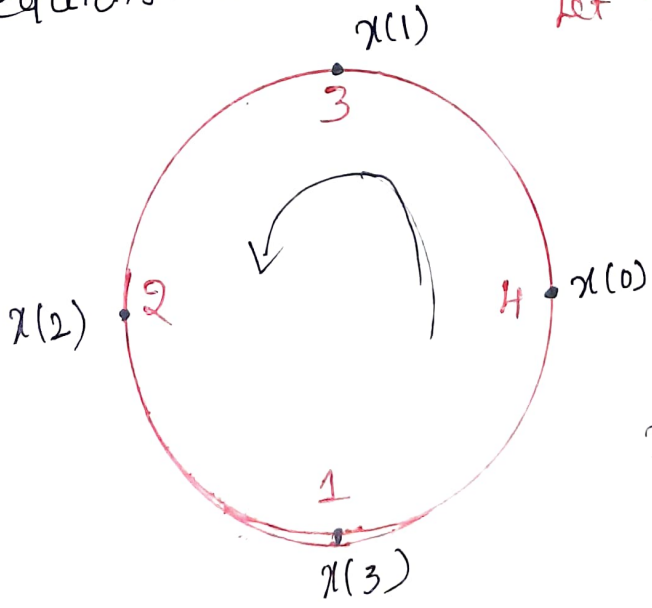
" $[x(N-1) = x(4-1) = x(3)]$

$$x'(2) = x((2-2)_4) = x(0) = 4$$

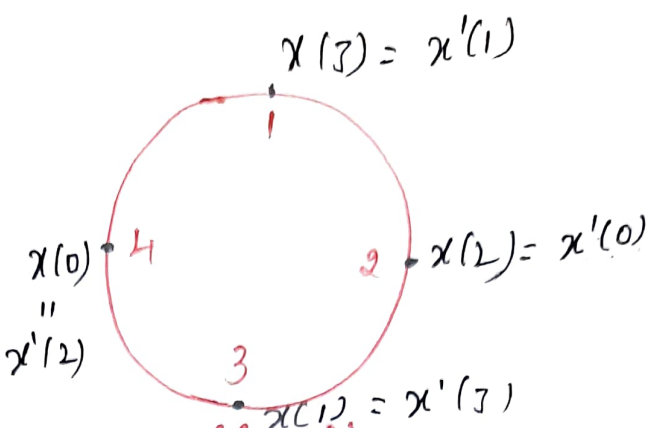
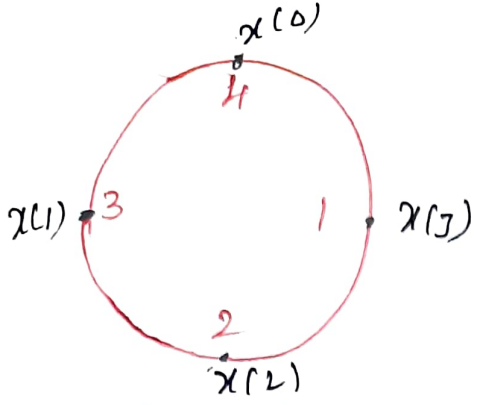
$$x'(3) = x((3-2)_4) = x(1) = 3$$

Another simple way is to write the samples of seqn on the circumference of a circle at equidistant in anticlockwise dirn.

Let $x(n) = \{4, 3, 2, 1\}$



$x(n)$ is written in anticlockwise dirn.
 $x((n)_4)$



(b) seq $x(n)$ shifted by one sample in anticlockwise dir

(c) seq $x(n)$ shifted by 2 samples in anticlockwise dir

$$x'(n) = \{2, 1, 4, 3\}$$

NOTE:-

$x((n-k))_N$, if k is +ve, the circle is to be rotated in anticlockwise dir by ' k ' steps [~~the~~ right circular shift]

if ' k ' is -ve, the circle is to be rotated in clockwise dir by ' k ' steps [left shift]

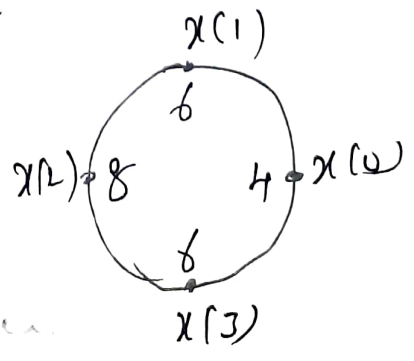
circularly even sequence

A seqⁿ is said to be circularly even, if it is symmetric about the point zero on the circle

$$x(N-n) = x(n) \quad 1 \leq n \leq N-1 \rightarrow \textcircled{1}$$

consider the seq $x(n) = \{4, 6, 8, 6\}$

$$\begin{aligned} x(4-1) &= x(3) = x(1) \\ x(4-2) &= x(2) = x(2) \\ x(4-3) &= x(1) = x(3) \end{aligned}$$



no need to explain this

circularly odd sequence

A seqⁿ is said to be circularly odd, if it is antisymmetrical about point $x(0)$ on the o/e i.e.,

$$x(N-n) = -x(n), \quad 1 \leq n \leq N-1 \rightarrow (2)$$

consider the seqⁿ $x(n) = \{+4, -6, 8, 6\}$

$$\left. \begin{aligned} x(4-1) = x(3) &= -x(1) \\ x(4-2) = x(2) &= -x(2) \\ x(4-3) = x(1) &= -x(3) \end{aligned} \right\} \text{no need to explain this}$$

circularly folded seqⁿ

A circularly folded seqⁿ is represented as $x((-n))_N$. It is obtained by plotting $x(n)$ in clockwise dirⁿ along the o/e & is represented as $x((-n))_N = x(N-n)$, $0 \leq n \leq N-1$

① A length-6 seqⁿ is given by

$$x(n) = \{1, 3, 2, 5, -2, 7\}$$

obtain $y(n) = x((n-4))_6$

Soln: $x(n) = \left\{ \begin{array}{cccccc} 1 & 3 & 2 & 5 & -2 & 7 \\ 0 & 1 & 2 & 3 & 4 & 5 \end{array} \right\}$

$$y(n) = x((n-4))_6$$

$$y(0) = x((0-4))_6 = x((-4))_6 = x(6-4) = x(2) = 2$$

$$y(1) = x((1-4))_6 = x((-3))_6 = x(6-3) = x(3) = 5$$

$$y(2) = x((2-4))_6 = x((-2))_6 = x(6-2) = x(4) = -2$$

$$y(n) = \{2, 5, -2, 7, 1, 3\}$$

② $x(n) = \{0, 3, 4, -1, 4, 2, 8, 9, 2, 3\}$

find $x'(n) = x((n-7))_{10}$

$y(n) = x((n+3))_{10}$

$\{-1, 4, 2, 8, 9, 2, 3, 0, 3, 4\}$

③ $x(n) = \{6, 5, 4, 3\}$. sketch $x'(n) = x((n-2))_4$

$x(n) = \{6, 5, 4, 3\}$

$x'(n-1) = \{3, 6, 5, 4\}$

$x(n-2) = \{4, 3, 6, 5\}$

$x_1(n) = x((n-2))_4 = \{4, 3, 6, 5\}$

③ Circular-time shift property.

(UK) if $x(n) \xleftrightarrow[N]{\text{DFT}} X(k)$

then $x((n-l))_N \xleftrightarrow[N]{\text{DFT}} X(k) e^{-j\frac{2\pi}{N}kl}$
 (or) $X(k) W_N^{kl}$

Proof:

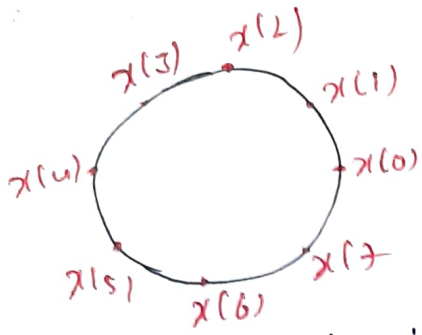
from the definition of DFT,

$$\text{DFT}\{x((n-l))_N\} = \sum_{n=0}^{N-1} x((n-l))_N e^{-j\frac{2\pi}{N}kn}$$

we can split the summations into 2 parts

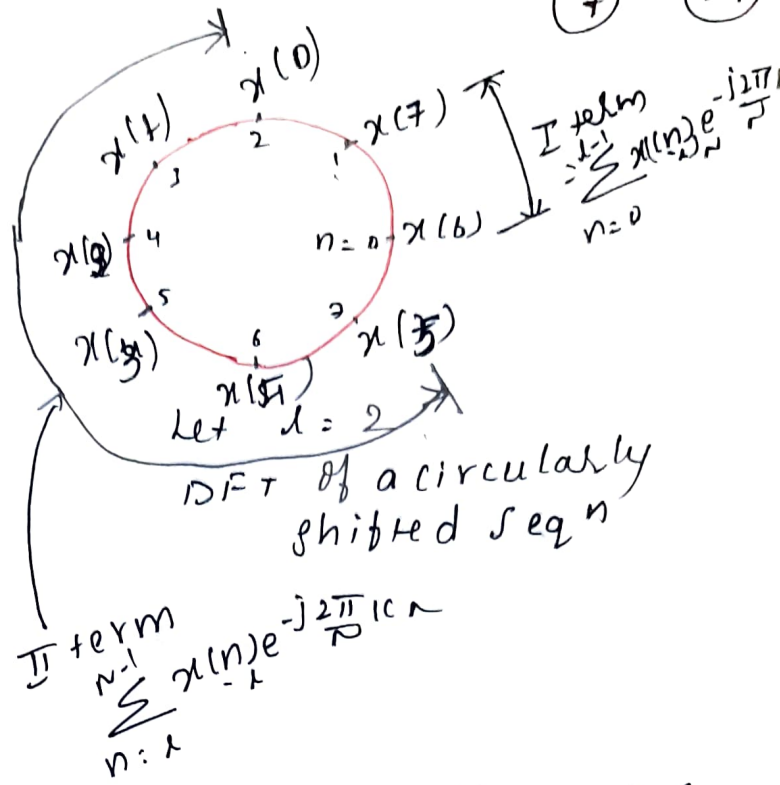
$$= \underbrace{\sum_{n=0}^{l-1} x((n-l))_N e^{-j\frac{2\pi}{N}kn}}_{\text{I}} + \underbrace{\sum_{n=l}^{N-1} x(n-l) e^{-j\frac{2\pi}{N}kn}}_{\text{II}} \rightarrow \text{①}$$

(for $n=0 \dots l-1$, the I seqⁿ is circular shifted, remaining values of $l \dots N-1$ - linear shifting) Ref fig below 47



A seqⁿ having 8 samples plotted on a circle

$x_0, x_1, x_2, x_3, x_4, x_5, x_6, x_7$
 $\{x_7, x_0, x_1, x_2, x_3, x_4, x_5, x_6\}$



but $x((n-l))_N = x(N-l+n)$ since there is circular shift.

consider I term

$$\sum_{n=0}^{l-1} x((n-l))_N e^{-j\frac{2\pi}{N}kn}$$

$$= \sum_{n=0}^{l-1} x(N-l+n) e^{-j\frac{2\pi}{N}kn}$$

put $m = N-l+n$ in the above eqⁿ.

$$= \sum_{m=N-l}^{N-1} x(m) e^{-j\frac{2\pi}{N}k(m+l-N)}$$

$$= \sum_{m=N-l}^{N-1} x(m) e^{-j\frac{2\pi}{N}k(m+l)} e^{j\frac{2\pi}{N}kN}$$

$$\therefore \sum_{m=N-l}^{N-1} x(m) e^{-j\frac{2\pi}{N}k(m+l)} \longrightarrow \textcircled{2}$$

$$\text{II } \underline{\text{Term}} \rightarrow \sum_{n=l}^{N-1} x(n-l) e^{-j\frac{2\pi}{N}kn}$$

$$\text{Put } m = n - l$$

$$= \sum_{m=0}^{N-l-1} x(m) e^{-j\frac{2\pi}{N}k(m+l)} \rightarrow \textcircled{3}$$

$$\therefore \text{DFT} \{x((n-l))_N\} = \text{II term} + \text{I term}$$

$$= \sum_{m=0}^{N-l-1} x(m) e^{-j\frac{2\pi}{N}k(m+l)}$$

$$+ \sum_{m=N-l}^{N-1} x(m) e^{-j\frac{2\pi}{N}k(m+l)}$$

The above summations can be combined into single one. Note that even though we have assumed 2 diff' values for m it's just an index. first summation in above eqn is performed from 0 to $N-l-1$ & 2nd summation is from $N-l$ to $N-1$. \therefore The overall summation can be replaced by 0 to $N-1$.

$$\text{DFT} \{x((n-l))_N\} = \sum_{m=0}^{N-1} x(m) e^{-j\frac{2\pi}{N}km} e^{-j\frac{2\pi}{N}kl}$$

$$= X(k) e^{-j\frac{2\pi}{N}kl}$$

$$= X(k) \omega_N^{kl}$$

the
Wenu-proof

① find 4-point DFT of the seqn

$$x(n) = \delta(n) + 2\delta(n-1) + 3\delta(n-2) + 4\delta(n-3)$$

also find the 4-point DFT $Y(K)$ if

$$y(n) = x((n-2))_4$$

Soln:

$$\text{DFT} \{x(n)\} = X(K) \triangleq \sum_{n=0}^{N-1} x(n) \omega_N^{Kn}$$

$$X(K) = \sum_{n=0}^3 [\delta(n) + 2\delta(n-1) + 3\delta(n-2) + 4\delta(n-3)] \omega_4^{Kn}$$

$$0 \leq K \leq 3$$

Applying shifting property, we get.

$$X(K) = \omega_4^{Kn} \Big|_{n=0} + 2\omega_4^{Kn} \Big|_{n=1} + 3\omega_4^{Kn} \Big|_{n=2} + 4\omega_4^{Kn} \Big|_{n=3}$$

$$= 1 + 2\omega_4^K + 3\omega_4^{2K} + 4\omega_4^{3K}$$

$$\omega_4^0 = 1, \omega_4^1 = -j, \omega_4^2 = -1, \omega_4^3 = +j$$

$$X(K) = \{10, -2+j2, -2, -2-j2\}$$

using time-shifting property

$$\text{DFT} \{x((n-1))_N\} \xleftrightarrow[N]{\text{DFT}} X(K) \omega_N^{Kl}$$

$$Y(K) = x((n-2))_4 \xleftrightarrow[N]{\text{DFT}} X(K) \omega_4^{2K}$$

$$(l=2, N=4)$$

$$Y(0) = X(0) \omega_4^{2 \cdot 0} = 1 \times 10 = \underline{\underline{10}}$$

$$Y(1) = X(1) \omega_4^2 = (-2+j2)(-1) = \underline{\underline{2-j2}}$$

$$Y(2) = X(2) \omega_4^4 = X(2) \omega_4^0 = (-2)(1) = -2$$

$$Y(3) = \omega_4^6 \cdot X(3) = \omega_4^2 \cdot X(3) = (-1)(-2-j2) = \underline{\underline{2+j2}}$$

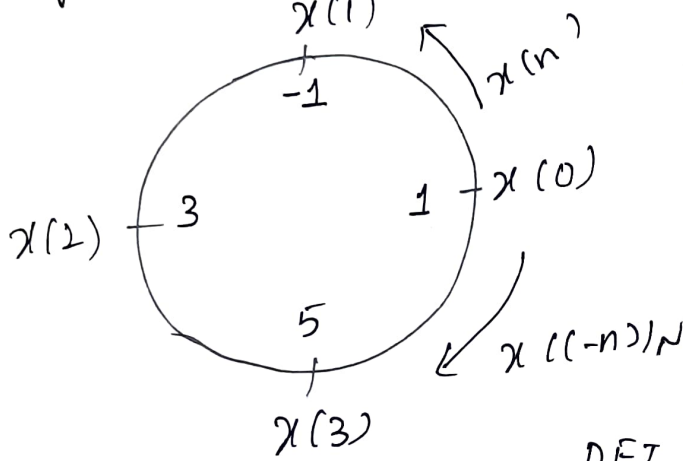
$$Y(K) = \{10, 2-j2, -2, 2+j2\}$$

(H) Time Reversal Property

WKT $x(n)$ is written on the circle at equidistance in anticlockwise diren. If the same seqⁿ $x(n)$ is written in clockwise diren. Then we will get the time reversal of $x(n)$ which is denoted by $x((-n))_N$

ex $x(n) = \{1, -1, 3, 5\}$

now $y(n) = x((-n))_4$
 $y(n) = \{1, 5, 3, -1\}$



statement: if $x(n) \xleftrightarrow[N]{DFT} X(k)$

Then $y(n) = x((-n))_N = x(N-n) \xleftrightarrow[N]{DFT} Y(k) = X((-k))_N = X(N-k)$

proof: [DFT is periodic over periods 'N'. $x(N-k)$ is equivalent to folding $x(k)$]

$Y(k) = DFT\{x((-n))_N\} = DFT\{x(N-n)\}$

$DFT\{x(N-n)\} \triangleq \sum_{n=0}^{N-1} x(N-n) e^{-j\frac{2\pi}{N}kn} \rightarrow (1)$

put $m = N-n$ then $n = N-m$

$= \sum_{m=N}^1 x(m) e^{-j\frac{2\pi}{N}k(N-m)} \rightarrow (2)$

$$= \sum_{m=1}^N x(m) e^{j\frac{2\pi}{N}km} \cdot e^{-j\frac{2\pi}{N}kN}$$

$$= e^{-j\frac{2\pi}{N}kN} = e^{-j2\pi k} = 1 \text{ always} \rightarrow \textcircled{3}$$

$$= \sum_{m=1}^N x(m) e^{j\frac{2\pi}{N}km}$$

$$= \sum_{m=0}^{N-1} x(m) e^{j\frac{2\pi}{N}km} \rightarrow \textcircled{4}$$

on the basis of eq (3) we can write

$$e^{-j2\pi m} = \cos(2\pi m) - j \sin(2\pi m) = 1 \quad \forall m$$

hence if we multiply R.H.S of eq (4) by $e^{-j2\pi m}$ its meaning will not change

$$\text{DFT}\{x(N-n)\} = \sum_{m=0}^{N-1} x(m) e^{j\frac{2\pi}{N}km} \cdot e^{-j2\pi m}$$

let us rearrange $e^{-j2\pi m}$ as $e^{-j\frac{2\pi}{N}mN}$.

$$\text{Then} \\ = \sum_{m=0}^{N-1} x(m) e^{j\frac{2\pi}{N}km} \cdot e^{-j\frac{2\pi}{N}mN}$$

$$= \sum_{m=0}^{N-1} x(n) e^{-j\frac{2\pi}{N}m(N-k)}$$

by definition of DFT, the R.H.S of the above eq is

$$\text{DFT}\{x(N-n)\} = X(N-k) \text{ hence the proof} \\ = X((N-k))_N$$

(1) Obtain the DFT of the foll seqn
 $x_1(n) = \{1, 1, 2, 3\}$ & also find DFT of
 $x_2(n) = x_1((-n))_4$.

$$x_1(n) = \{1, 1, 2, 3\}$$

$$X_1(k) = \{7, -1+j2, -1, -1-j2\}$$

~~$X_2(k)$~~
 WKT $x((-n))_N \xrightarrow[N]{\text{DFT}} X((-k))_N$

$$X_2(k) = \text{DFT}\{x_1((-n))_4\} = X_1((-k))_4$$

$$X_2(0) = X_1((-0))_4 = X_1(0) = 7$$

$$X_2(1) = X_1((-1))_4 = X_1(4-1) \\ = X_1(3) = -1-j2$$

$$X_2(2) = X_1((-2))_4 = X_1(4-2) \\ = X_1(2) = -1$$

$$X_2(3) = X_1((-3))_4 = X_1(4-3) \\ = X_1(1) = -1+j2$$

$$X_2(k) = \{7, -1-j2, -1, -1+j2\}$$

⑤ circular frequency shift :

(10) (28)

if $x(n) \xleftrightarrow[N]{DFT} X(k)$

then $x(n) e^{j\frac{2\pi}{N}dn} \xleftrightarrow[N]{DFT} X((k-d))_N$

Proof :

$Y(n) = IDFT \{ X((k-d))_N \}$

$= \frac{1}{N} \left[\sum_{k=0}^{N-1} X((k-d))_N e^{j\frac{2\pi}{N}kn} \right] \rightarrow (1)$

$= \frac{1}{N} \left[\underbrace{\sum_{k=0}^{d-1} X((k-d))_N e^{j\frac{2\pi}{N}kn}}_{I \text{ - term}} + \underbrace{\sum_{k=d}^{N-1} X(k-d) e^{j\frac{2\pi}{N}kn}}_{II \text{ - term}} \right]$

II - term $\rightarrow (2)$

I - term

$\sum_{k=0}^{d-1} X((k-d))_N e^{j\frac{2\pi}{N}kn}$

$= \sum_{k=0}^{d-1} X(N-d+k) e^{j\frac{2\pi}{N}kn}$

put $m = N-d+k$

$= \sum_{m=N-d}^{N-1} X(m) e^{j\frac{2\pi}{N}n[m+d-N]}$

$= \sum_{m=N-d}^{N-1} X(m) e^{j\frac{2\pi}{N}n(m+d)} \cdot e^{-j\frac{2\pi}{N}n \cdot N} \quad \text{--- } = 1$

$= \sum_{m=N-d}^{N-1} X(m) e^{j\frac{2\pi}{N}n(m+d)} \rightarrow (3)$

II - term

$$\sum_{k=l}^{N-1} x(k-l) e^{j\frac{2\pi}{N}kn}$$

put $m = k - l \quad \therefore k = m + l$

$$= \sum_{m=0}^{N-l-1} x(m) e^{j\frac{2\pi}{N}n(m+l)} \rightarrow \textcircled{H}$$

$$y(n) = \text{IDFT} \{ X((k-l))_N \} = \frac{1}{N} [\text{II term} + \text{I term}]$$

$$= \frac{1}{N} \left[\sum_{m=0}^{N-l-1} x(m) e^{j\frac{2\pi}{N}n(m+l)} + \sum_{m=N-l}^{N-1} x(m) e^{j\frac{2\pi}{N}n(m+l)} \right]$$

$$= \frac{1}{N} \sum_{m=0}^{N-1} x(m) e^{j\frac{2\pi}{N}n(m+l)}$$

$$= \frac{1}{N} \sum_{m=0}^{N-1} x(m) e^{j\frac{2\pi}{N}nm} \cdot e^{j\frac{2\pi}{N}nl}$$

$$X((k-l))_N = x(n) \cdot e^{j\frac{2\pi}{N}nl}$$

$$= x(n) W_N^{-nl} \text{ hence}$$

the proof

Q find 4-point DFT of the seq'n given below. $x(n) = \{1, -1, -1, 1\}$. Also find $y(n)$ if $Y(K) = X((K-2))_4$

Soln

$$\text{DFT } \{x(n)\} = X(K) = \sum_{n=0}^3 x(n) \omega_4^{Kn}$$

$$X(K) = 1 - \omega_4^K - \omega_4^{2K} + \omega_4^{3K}$$

$$X(K) = \{0, 2+2j, 0, 2-j^2\}$$

Recalling the shifting property

$$\text{DFT } \{ \omega_N^{-dn} x(n) \} = X((K-d))_N$$

$$\text{IDFT } \{ X((K-d))_N \} = x(n) \cdot \omega_N^{dn}$$

$$\therefore Y(K) = X((K-2))_4$$

$$\therefore y(n) = \omega_4^{-2n} x(n)$$

$$y(0) = \omega_4^{-0} x(0) = [\omega_4^0]^* x(0) = 1 * 1 = \underline{\underline{1}}$$

$$y(1) = \omega_4^{-2} x(1) = [\omega_4^2]^* x(1) = -1 * -1 = 1$$

$$y(2) = \omega_4^{-0} x(2) = [\omega_4^0]^* x(2) = 1 * -1 = -1$$

$$y(3) = \omega_4^{-2} x(3) = [\omega_4^2]^* x(3) = -1 * 1 = -1$$

$$y(n) = \{ \underline{\underline{1, 1, -1, -1}} \}$$

(6) circular convolution property

Before studying the property, let us discuss the concept of circular-convolution.

Let

$x_1(n)$ & $x_2(n)$ be two N -length seq^s.
The circular convolution of these sequence's defined as

$$y(n) = \sum_{m=0}^{N-1} x_1(m) x_2((n-m))_N \longrightarrow \textcircled{1}$$

; $0 \leq n \leq N-1$

this operation involves 2-length N 's seq^s & so it is also referred as N -point circular convolution & is denoted as

$$x_1(n) \textcircled{N} x_2(n) = x_2(n) \textcircled{N} x_1(n) \longrightarrow \textcircled{2}$$

(i) $x_1(n) = \{1, 2, 0, 1\}$, $x_2(n) = \{2, 2, 1, 1\}$
find $y(n) = x_1(n) \textcircled{N} x_2(n)$.

The 4-point circular convolution of these 2 seqs is given by

(12)

$$y(n) = x_1(n) \otimes x_2(n) \\ = x_1(n) \textcircled{4} x_2(n); \quad 0 \leq n \leq 3$$

$$= \sum_{m=0}^3 x_1(m) x_2((n-m))_4$$

n=0

$$y(0) = \sum_{m=0}^3 x_1(m) x_2((0-m))_4$$

$$= x_1(0) x_2((0-0))_4 + x_1(1) x_2((0-1))_4 + x_1(2) x_2((0-2))_4 + x_1(3) x_2((0-3))_4$$

$$= x_1(0) x_2(0) + x_1(1) x_2(3) + x_1(2) x_2(2) + x_1(3) x_2(1)$$

$$= 1 \cdot 2 + 2 \cdot 1 + 0 \cdot 1 + 1 \cdot 2$$

$$= \underline{\underline{6}}$$

n=1

$$y(1) = \sum_{m=0}^3 x_1(m) x_2((1-m))_4$$

$$= x_1(0) x_2((1-0))_4 + x_1(1) x_2((1-1))_4 + x_1(2) x_2((1-2))_4 + x_1(3) x_2((1-3))_4$$

$$= x_1(0) x_2(1) + x_1(1) x_2(0) + x_1(2) x_2(3) + x_1(3) x_2(2)$$

$$= 1 \cdot 2 + 2 \cdot 2 + 0 \cdot 1 + 1 \cdot 1$$

$$= \underline{\underline{7}}$$

n=2

$$y(2) = \sum_{m=0}^3 x_1(m) x_2((2-m))_4$$

$$= x_1(0) x_2(2) + x_1(1) x_2(1) + x_1(2) x_2(0) + x_1(3) x_2(3)$$

$$= \underline{\underline{6}} \\ y(3) = \sum_{m=0}^3 x_1(m) x_2((3-m))_4 = x_1(0) x_2(3) + x_1(1) x_2(2) + x_1(2) x_2(1) + x_1(3) x_2(0) \\ = 5$$

$$y(n) = [6, 7, 6, 5]$$

mainly there are 2 methods to find the circular convolution of 2 length-N seq

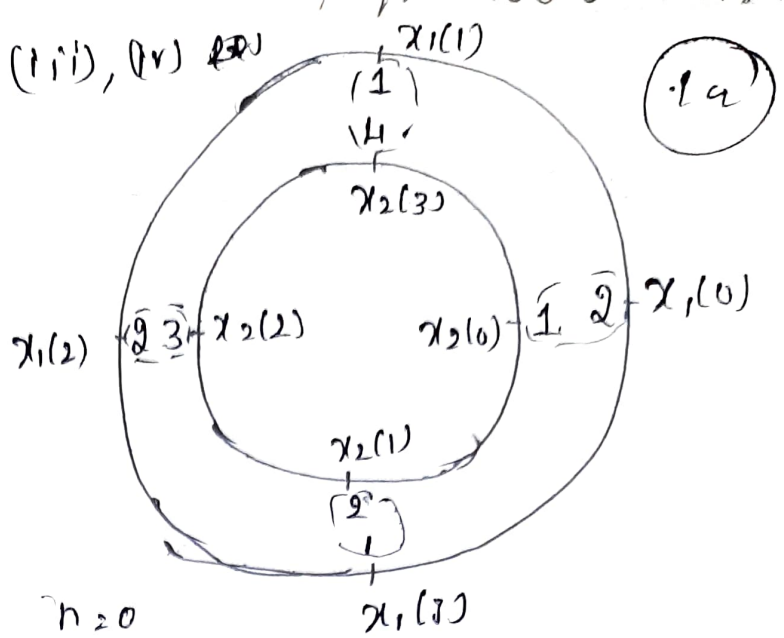
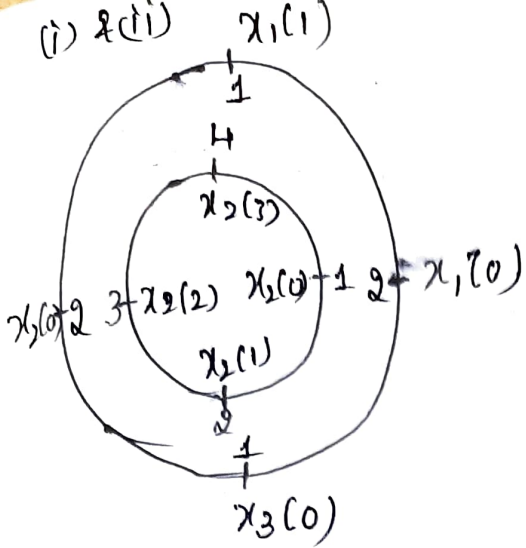
13

(a) Concentric circle method:

Given 2 length-N seqs $x_1(n)$ & $x_2(n)$, the N-point circular convolution of these 2 seqs $y(n) = x_1(n) \otimes x_2(n)$ can be performed by following steps:

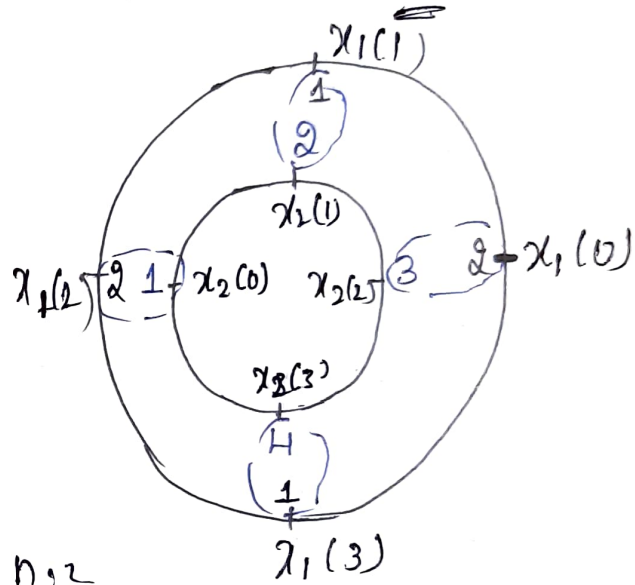
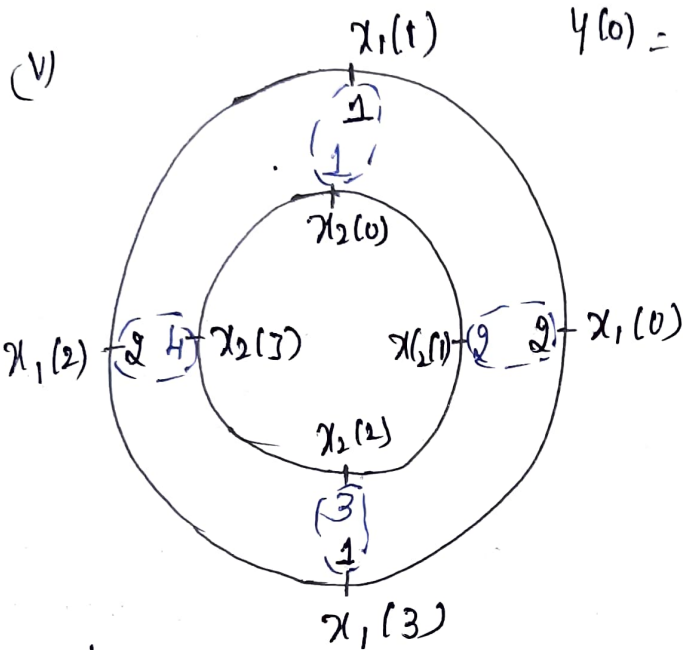
- (i) Write N samples of $x_1(n)$ at equidistant around an outer circle in anticlockwise direction
- (ii) Starting at the same point as $x_1(n)$ circle, write N samples of $x_2(n)$ at equidistant around the inner circle in clockwise direction
- (iii) Multiply corresponding samples on the 2 circles & sum the products to get the O/P.
- (iv) Rotate the inner circle by one sample in anticlockwise direction & repeat step (iii) to obtain the next samples of the O/P
- (v) Repeat step (iv) for N-1 times to get all the samples of the O/P $y(n)$.

eg. $x_1(n) = \{2, 1, 2, 1\}$
& $x_2(n) = \{1, 2, 3, 4\}$



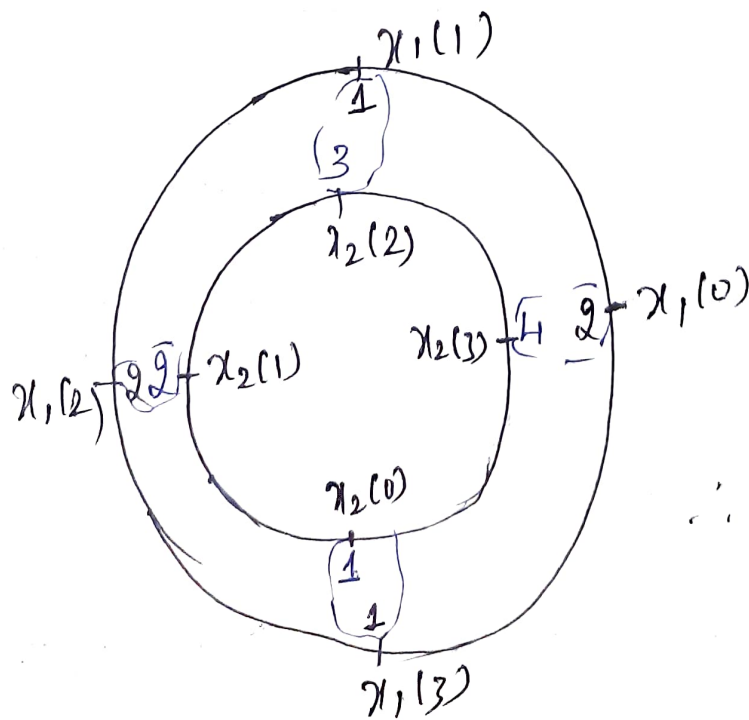
(19)

$n=0$
 $\psi(0) = 2 + 4 + 6 + 2 = 14$



$n=1$
 $\psi(1) = 4 + 1 + 8 + 3 = 16$

$n=2$
 $\psi(2) = 6 + 2 + 2 + 4 = 14$



$n=3$
 $\psi(3) = 8 + 3 + 4 + 1 = 16$

$\therefore \psi(n) = \{ 14, 16, 14, 16 \}$

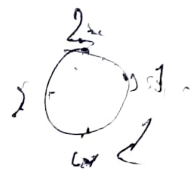
① Compute the circular convolution of the 2 seqs given below $x(n) = \{1, 4, 2, 6\}$ $h(n) = \{1, 2, 3, 4\}$

Let $y_c(n) = x(n) \circledast h(n) = \sum_{m=0}^3 x(m) h[(n-m)]_4$

② circul.
n

	$x(m)$	$h[(n-m)]_4$	$y(n)$
0	$\{1, 4, 2, 6\}$	$\{1, 4, 3, 2\}$	$1+16+6+12 = 35$
1	$\{1, 4, 2, 6\}$	$\{2, 1, 4, 3\}$	$2+4+8+18 = 32$
2	$\{1, 4, 2, 6\}$	$\{3, 2, 1, 4\}$	$3+8+2+24 = 37$
3	$\{1, 4, 2, 6\}$	$\{4, 3, 2, 1\}$	$4+12+4+6 = 26$

$y_c(n) = \{35, 32, 37, 26\}$



③ linear conv

$x(n) = \{1, 4, 2, 6\}$ & $h(n) = \{1, 2, 3, 4\}$

$y_d(n) = x(n) * h(n)$
 $= [\delta(n) + 4\delta[n-1] + 2\delta[n-2] + 6\delta[n-3]]$
 $* [\delta(n) + 2\delta[n-1] + 3\delta[n-2] + 4\delta[n-3]]$

$= \delta(n) + 6\delta(n-1) + 13\delta(n-2) + 26\delta(n-3)$
 $+ 34\delta(n-4) + 26\delta(n-5) + 24\delta(n-6)$

$y_d(n) = \{1, 6, 13, 26, 34, 26, 24\}$
 $= \{1+34, 6+26, 13+24, 26\} = \{35, 32, 37, 26\}$

Circular conv = linear conv + Aliasing

(b) matrix multiplication method

In this method, the N -point circular convolution of 2 length- N seqs $x_1(n)$ & $x_2(n)$ can be obtained as below. (16)

$$\begin{bmatrix} y(0) \\ y(1) \\ y(2) \\ \vdots \\ y(N-1) \end{bmatrix} = \begin{bmatrix} x_1(0) & x_1(N-1) & x_1(N-2) & \dots & x_1(1) \\ x_1(1) & x_1(0) & x_1(N-1) & \dots & x_1(2) \\ x_1(2) & x_1(1) & x_1(N) & \dots & x_1(3) \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ x_1(N-1) & x_1(N-2) & x_1(N-3) & \dots & x_1(0) \end{bmatrix} \begin{bmatrix} x_2(0) \\ x_2(1) \\ x_2(2) \\ \vdots \\ x_2(N-1) \end{bmatrix}$$

The seq'n $x_1(n)$ is written in each column separately by circularly shifting the samples of ~~it~~ it in a $N \times N$ matrix. A such a matrix is called circulant matrix.

Let $x_1(n) = \{2, 1, 2, 1\}$ $x_2(n) = \{1, 2, 3, 4\}$

$$\begin{bmatrix} y(0) \\ y(1) \\ y(2) \\ y(3) \end{bmatrix} = \begin{bmatrix} 2 & 1 & 2 & 1 \\ 1 & 2 & 1 & 2 \\ 2 & 1 & 2 & 1 \\ 1 & 2 & 1 & 2 \end{bmatrix} \begin{bmatrix} 1 \\ 2 \\ 3 \\ 4 \end{bmatrix}$$

$$= \begin{bmatrix} 2 \cdot 1 + 1 \cdot 2 + 2 \cdot 3 + 1 \cdot 4 \\ 1 \cdot 1 + 2 \cdot 2 + 1 \cdot 3 + 2 \cdot 4 \\ 2 \cdot 1 + 1 \cdot 2 + 2 \cdot 3 + 1 \cdot 4 \\ 1 \cdot 1 + 2 \cdot 2 + 1 \cdot 3 + 2 \cdot 4 \end{bmatrix} = \begin{bmatrix} 14 \\ 16 \\ 14 \\ 16 \end{bmatrix}$$

$\therefore y(n) = x_1(n) \textcircled{4} x_2(n)$
 $= \{14, 16, 14, 16\}$

circular convolution Property

$$\text{if } x_1(n) \xleftrightarrow[N]{\text{DFT}} X_1(k)$$

$$\& x_2(n) \xleftrightarrow[N]{\text{DFT}} X_2(k)$$

(U.K)

(12)

$$\text{Then } x_1(n) \otimes x_2(n) \xleftrightarrow[N]{\text{DFT}} X_1(k) \cdot X_2(k)$$

multiplication of 2 DFTs is equivalent to circular convolution of their seqs in time domain

Proof

By defn.

$$X_1(k) \triangleq \sum_{n=0}^{N-1} x_1(n) e^{-j\frac{2\pi}{N}kn}, k=0, \dots, N-1 \rightarrow (1)$$

$$\& X_2(k) \triangleq \sum_{n=0}^{N-1} x_2(n) e^{-j\frac{2\pi}{N}kn}, k=0, \dots, N-1 \rightarrow (2)$$

$$\text{let } X_3(k) = X_1(k) \cdot X_2(k) \rightarrow (3)$$

let $x_3(m)$ be the seqⁿ whose DFT is $X_3(k)$

$$\text{Then } x_3(m) = \frac{1}{N} \sum_{k=0}^{N-1} X_3(k) e^{j\frac{2\pi}{N}km}$$

substituting eq (3) in ~~(4)~~ we get

$$x_3(m) = \frac{1}{N} \sum_{k=0}^{N-1} X_1(k) \cdot X_2(k) e^{j\frac{2\pi}{N}km} \rightarrow (4)$$

substituting (1) & (2) in eq (4)

$$x_3(m) = \frac{1}{N} \sum_{k=0}^{N-1} \left[\sum_{n=0}^{N-1} x_1(n) e^{-j\frac{2\pi}{N}kn} \right] \left[\sum_{l=0}^{N-1} x_2(l) e^{-j\frac{2\pi}{N}kl} \right] e^{j\frac{2\pi}{N}km}$$

here all the 3 summations have different indices since they are independent. Rearranging the summations & terms in above eqn as follows. (18)

$$x_3(m) = \frac{1}{N} \sum_{n=0}^{N-1} x_1(n) \sum_{l=0}^{N-1} x_2(l) \left\{ \sum_{k=0}^{N-1} e^{j\frac{2\pi}{N}k(m-n-l)} \right\}$$

→ (5)

Now let us consider the standard eqn

$$\sum_{k=0}^{N-1} a^k = \begin{cases} N, & a=1 \\ \frac{1-a^N}{1-a}, & a \neq 1 \end{cases} \rightarrow (6)$$

here now $a = e^{j\frac{2\pi}{N}(m-n-l)}$

when $(m-n-l) = \text{multiple of } N$ i.e. $N, 2N, 3N$

Then $a = 1$ \because $a e^{j\frac{2\pi}{N}N} = e^{j\frac{2\pi}{N}2N}$
 $= e^{j2\pi 2} = 1$

Thus for the 1st condition, we can write it for $a=1$

$$\sum_{k=0}^{N-1} a^k = \sum_{k=0}^{N-1} e^{j\frac{2\pi}{N}k(m-n-l)} = N, \text{ when } (m-n-l) \text{ is integer multiple of } N$$

→ (7)

Now let us consider the 2nd condition. i.e. for $a \neq 1$

$$\begin{aligned} \sum_{k=0}^{N-1} a^k &= \sum_{k=0}^{N-1} e^{j\frac{2\pi}{N}k(m-n-l)} \text{ when } (m-n-l) \text{ is not integer multiple of } N \\ &= \frac{1-a^N}{1-a} \\ &= \frac{1 - e^{j\frac{2\pi}{N}(m-n-l)N}}{1 - e^{j\frac{2\pi}{N}(m-n-l)}} \end{aligned}$$

$$e^{j2\pi(m-n-l)} = 1 \text{ always}$$

(12)

$$\sum_{k=0}^{N-1} e^{j\frac{2\pi k}{N}(m-n-l)} = 0, \text{ when } (m-n-l) \text{ is not multiple of } N.$$

Thus

$$\sum_{k=0}^{N-1} e^{j\frac{2\pi k}{N}(m-n-l)} = \begin{cases} N, & \text{when } (m-n-l) \text{ is multiple of } N \\ 0, & \text{o.w} \end{cases}$$

→ (8)

substituting these in eq (5)

$$x_3(m) = \frac{1}{N} \sum_{n=0}^{N-1} x_1(n) \sum_{l=0}^{N-1} x_2(l) \cdot N \quad \text{when } (m-n-l) \text{ is integer multiple of } N$$

$$= \sum_{n=0}^{N-1} x_1(n) \sum_{l=0}^{N-1} x_2(l) \rightarrow (9)$$

$(m-n-l)$ is a multiple of N

∴ we can write $(m-n-l) = pN$.

p is an integer & may be +ve or -ve
we can write for our convenience of

$$m-n-l = -pN$$

$$l = m-n+pN \rightarrow (10)$$

Sub in eq (9)

$$= \sum_{n=0}^{N-1} x_1(n) \cdot x_2(m-n+pN) \rightarrow (10)$$

since $x_2(m-n+PN)$ is a periodic seqⁿ with period N .

then $x_2(m)$ is shifted circularly by ' n ' samples

$$x_2(m-n+PN) = x_2(m-n, \text{modulon } N) \\ = x_2((m-n)_N)$$

∴ eq (10) becomes

$$x_3(m) = \sum_{n=0}^{N-1} x_1(n) x_2((m-n)_N), \quad m=0, \dots, N-1$$

↳ (11)

let us compare eq (11) with linear convⁿ

$$y(n) = \sum_{k=-\infty}^{\infty} x(k) h(n-k)$$

eq (11) appears like convolution operⁿ but the seqⁿ is shifted circularly hence it is called circular convolution

$$x_3(m) = x_1(n) \textcircled{N} x_2(n) = \sum_{n=0}^{N-1} x_1(n) x_2((m-n)_N)$$

(1) find the circular convolution using DFT & IDFT for the foll seqⁿ

$$x_1(n) = \{2, 3, 1, 1\}$$

$$x_2(n) = \{1, 3, 5, 3\}$$

use convolⁿ property

$$X_1(k) = \sum_{n=0}^{N-1} x_1(n) e^{-j\frac{2\pi}{N}kn}$$

$$= \{7, 1-j2, -1, 1+j2\}$$

$$X_2(k) = \{19, -4, 0, -4\}$$

$$Y(k) = Y_1(k) \cdot Y_2(k) = \{84, -4+j8, 0, -4-j8\}$$

$$Y(0) = Y_1(0) \cdot Y_2(0) = \dots$$

Now take IDFT on $Y(k)$

$$Y(n) = \frac{1}{N} \sum_{k=0}^{N-1} Y(k) e^{j \frac{2\pi}{N} kn}$$

$$= \{19, 17, 23, 25\}$$

$$Y(n) = X_1(n) \textcircled{4} X_2(n)$$

(2) Given the seqⁿ $X_1(n) = \{1, 2, 3, 1\}$ & $X_2(n) = \{4, 3, 2, 2\}$. find $Y(n)$

such that $Y(k) = X_1(k) \cdot X_2(k)$

solⁿ given $X_1(n) = \{1, 2, 3, 1\}$

$X_2(n) = \{4, 3, 2, 2\}$

$$Y(k) = X_1(k) \cdot X_2(k)$$

taking IDFT

$$Y(n) = X_1(n) \textcircled{N} X_2(n)$$

$$\begin{bmatrix} Y(0) \\ Y(1) \\ Y(2) \\ Y(3) \end{bmatrix} = \begin{bmatrix} 1 & 1 & 3 & 1 \\ 2 & 1 & 1 & 3 \\ 3 & 2 & 1 & 1 \\ 1 & 3 & 2 & 1 \end{bmatrix} \begin{bmatrix} 4 \\ 3 \\ 2 \\ 2 \end{bmatrix} = \begin{bmatrix} 17 \\ 19 \\ 22 \\ 19 \end{bmatrix}$$

modulation property (or) multiplication of 2 seqns

If $x_1(n) \xleftrightarrow{DFT} X_1(k)$

u.c (22)

& $x_2(n) \xleftrightarrow{DFT} X_2(k)$

then $y(n) = x_1(n) \cdot x_2(n) \xleftrightarrow{DFT} Y(k) = \frac{1}{N} [X_1(k) \otimes X_2(k)]$

proof

consider 2 seqns $x_1(n)$ & $x_2(n)$

$x_1(n) \triangleq \frac{1}{N} \sum_{k=0}^{N-1} X_1(k) e^{j\frac{2\pi}{N}kn} \rightarrow (1)$
 $0 \leq n \leq N-1$ [14/12/17]

$x_2(n) \triangleq \frac{1}{N} \sum_{k=0}^{N-1} X_2(k) e^{j\frac{2\pi}{N}kn} \rightarrow (2)$
 $0 \leq n \leq N-1$ (k=1)

let $y(n) = x_1(n) \cdot x_2(n) \rightarrow (3)$

The N-point DFT of $y(n)$ is

$Y(k) = \sum_{n=0}^{N-1} y(n) e^{-j\frac{2\pi}{N}kn} \rightarrow (4)$

sub eq (3) in (4)

$Y(k) = \sum_{n=0}^{N-1} x_1(n) \cdot x_2(n) e^{-j\frac{2\pi}{N}kn} \rightarrow (5)$

sub eq (1) & (2) in eq (5)

$Y(k) = \frac{1}{N^2} \sum_{n=0}^{N-1} \left[\sum_{k=0}^{N-1} X_1(k) e^{j\frac{2\pi}{N}kn} \right] \left[\sum_{l=0}^{N-1} X_2(l) e^{j\frac{2\pi}{N}ln} \right] e^{-j\frac{2\pi}{N}kn} \rightarrow (6)$

Replace k by m & l in eq (1) & (2)

$= \frac{1}{N^2} \sum_{n=0}^{N-1} \left[\sum_{m=0}^{N-1} X_1(m) e^{j\frac{2\pi}{N}mn} \right] \left[\sum_{l=0}^{N-1} X_2(l) e^{j\frac{2\pi}{N}ln} \right] e^{-j\frac{2\pi}{N}kn}$

$= \frac{1}{N^2} \sum_{m=0}^{N-1} X_1(m) \sum_{l=0}^{N-1} X_2(l) \sum_{n=0}^{N-1} e^{-j\frac{2\pi}{N}n[k-m-l]} \rightarrow (7)$

The ^{last} summation in eq (2)

(23)

$$\sum_{n=0}^{N-1} a^n = \frac{1-a^N}{1-a}, \quad a \neq 1$$

$$\text{or } \sum_{n=0}^{N-1} a^n = N, \quad (k-m-1) = PN$$

$$= 0, \quad \text{o.l.}$$

∴ eq (2) becomes

$$Y(k) = \frac{1}{N} \sum_{m=0}^{N-1} X_1(m) \sum_{l=0}^{N-1} X_2(l)$$

$$\text{or wkt } k-m-l = -PN \quad l = k-m+PN$$

$$= \frac{1}{N} \sum X_1(m) X_2((k-m))_N$$

$$= \frac{1}{N} [X_1(k) \otimes X_2(k)]$$

(1) Let seqs $x_1(n) = \{1, 1, 2, 1\}$ & $x_2(n) = \{1, 0, 1, 0\}$
 find 4-point DFT of $y(n) = x_1(n) \cdot x_2(n)$

solⁿ find $X_1(k) = \{5, -1, 1, -1\}$

$$X_2(k) = \{2, 0, 2, 0\}$$

using moduli property
 $y(n) = x_1(n) \cdot x_2(n)$

$$Y(k) = \frac{1}{4} [X_1(k) \otimes X_2(k)]$$

$$= \frac{1}{4} \begin{bmatrix} 5 & -1 & 1 & -1 \\ -1 & 5 & -1 & 1 \\ 1 & -1 & 5 & 1 \\ -1 & 1 & -1 & 5 \end{bmatrix} \begin{bmatrix} 2 \\ 0 \\ 2 \\ 0 \end{bmatrix} = \begin{bmatrix} 3 \\ -1 \\ 3 \\ -1 \end{bmatrix}$$

2. EFFICIENT COMPUTATION OF FFT

2.1 Introduction:

WRT, DFT is widely used DSP algorithm. DFT also plays an important role in many applications of DSP processing including linear-filtering, spectrum analysis etc. A major reason for this its importance is the existence of efficient algorithms for computation of DFT.

The different computationally efficient algorithms are discussed here for the evaluating the DFT.

There are 2 approaches for evaluating the DFT in a computationally efficient manner

(i) Divide & Conquer approach:

* In this approach, the N -point DFT is reduced to the computation of smaller DFTs from which the larger DFT is computed.

(N -composite no) ~~to be the C.~~
* The computational algorithm called Fast-Fourier Transform (FFT) algorithms for computing the DFT where the size N is power of 2. This is known as Radix-2 FFT algorithms.

(ii) A linear filtering approach:

* It is based on the formulation of DFT as a linear filtering operation on the data.

* In this approach the algorithms like ~~Greif~~ ~~Greif~~ ~~Greif~~ Goertzel & Chirp-Z transform are used.

9.2 Efficient computation of the DFT

Direct computation of DFT

From the defⁿ of DFT, we have

$$X(k) = \sum_{n=0}^{N-1} x(n) \omega_N^{kn} \rightarrow (1)$$

$$0 \leq k \leq N-1$$

$x(n)$ may be real / complex

$\omega_N \rightarrow$ Twiddle factor which is complex no.

The above summation includes multiplications & summing of complex nos.

~~$X(k) = x(0)\omega_N^0 + x(1)\omega_N^k + \dots + x(N-1)\omega_N^{(N-1)k}$~~

$X(k) = x(0) \boxed{x} \omega_N^0 \oplus x(1) \boxed{x} \omega_N^k \oplus x(2) \boxed{x} \omega_N^{2k} + \dots \oplus x(N-1) \boxed{x} \omega_N^{(N-1)k}$

Complex multiplications (pointing to \boxed{x})

Complex addition (pointing to \oplus)

for any values of k , we have

$N \rightarrow$ no. of complex multiplications $\rightarrow (1)$

$(N-1) \rightarrow$ additions

\therefore For evaluation of $x(k)$ from 0 to $N-1$ requires

no. of complex multiplications = $N \times N = N^2$

no. of complex additions = $(N-1) \times N = N(N-1)$

$\rightarrow (2)$

Also eq ① can be written as if

$$x(n) = x_R(n) + j x_I(n)$$

$$\omega_N^{kn} = \text{Re}(\omega_N^{kn}) + j \text{Im}(\omega_N^{kn})$$

$$X(k) = \sum_{n=0}^{N-1} \{ x_R(n) + j x_I(n) \} \{ \text{Re}(\omega_N^{kn}) + j \text{Im}(\omega_N^{kn}) \}$$

$$= \sum_{n=0}^{N-1} x_R(n) \text{Re}(\omega_N^{kn}) + j x_R(n) \text{Im}(\omega_N^{kn}) + j x_I(n) \text{Re}(\omega_N^{kn}) - x_I(n) \text{Im}(\omega_N^{kn})$$

$$= \sum_{n=0}^{N-1} (x_R(n) \text{Re}(\omega_N^{kn}) - x_I(n) \text{Im}(\omega_N^{kn})$$

$$+ j \{ x_I(n) \text{Im}(\omega_N^{kn}) + x_I(n) \text{Re}(\omega_N^{kn}) \}$$

$$= x_R(n) \boxed{\times} \text{Re}(\omega_N^{kn}) - x_I(n) \boxed{\times} \text{Im}(\omega_N^{kn}) + j [x_R(n) \boxed{\times} \text{Im}(\omega_N^{kn}) + x_I(n) \boxed{\times} \text{Re}(\omega_N^{kn})]$$

4 real xions

2-real addition

NOTE 1. * subtraction is also counted as a addition in DSP since it requires almost same time as addition.

* $a + jb$ is never executed \because it is just a way of representing complex no.

* One complex multiplication is converted into $\left. \begin{array}{l} \rightarrow 4 \text{ Real xions} \\ \rightarrow 2 \dots \text{tions} \end{array} \right\}$

\rightarrow (2)

* for each values of k .

~~one~~ N complex multiplication is converted into $\left. \begin{array}{l} \rightarrow 4 \cdot N \text{ Real xion} \\ \rightarrow 2 \cdot N \text{ Real xion} \end{array} \right\}$

\rightarrow (3)

* k values from 0 to $N-1$

for complete DFT complex multiplications are converted into $\left. \begin{array}{l} \rightarrow 4N^2 = [4N \cdot N] \text{ Real xions} \\ \rightarrow 2N^2 \text{ Real + ion} \end{array} \right\}$

\rightarrow (5)

* now let us see how many complex additions are converted into

$(a+jb) + (c+jd)$
 $\underbrace{\hspace{10em}}_{2 \text{ complex nos}}$

$$(a+jb) + (c+jd) = (a+c) + j(b+d)$$

one complex addition \rightarrow $(a+c)$ and $j(b+d)$
 \rightarrow 2 real additions

* one complex addition is converted into \rightarrow 2 real additions \rightarrow (6)

* for each values of k , there are $(N-1)$ complex additions

hence $(N-1)$ complex additions is converted into \rightarrow $2(N-1)$ real additions \rightarrow (7)

* k varies from 0 to $N-1$
then

hence

total

for complete DFT complex additions are converted into	}	$= 2 \times N-1 \times N$	→ ⑧
		$= 2[N^2 - N]$	
		$= 2N^2 - 2N$	

* Total real additions are from eq ⑤

Total real additions in computation of DFT	$= 2N^2 - 2N + 2N$	→ ⑨
	$= 4N^2 - 2N$	
	$= N[4N - 2]$	

Thus for direct computation of N -point DFT requires the foll arithmetic operⁿ:

- * N complex multiplications for each value of k
- * N^2 complex multiplications for all values of k
- * $(N-1)$ complex additions for each values of k
- * $N(N-1)$ complex additions for all values of k

(OR)

- * $4N$ real multiplications for each value of k
- * $4N^2$ real multiplications for all values of k
- * $(4N-2)$ real additions for each values of k
($2N-2+2N = 4N-2$)
- * $N(4N-2)$ real additions for all values of k .

- * ∞^0 The no of arithmetic operns in direct computation of DFT is large & thus it is time consuming.
- * The total no of operations pres very rapidly as N ↑.
- * Hence it is of practical interest to develop more efficient fast algorithms for computing the DFT.

Periodicity property of ω_N

① $\omega_N^{k+N} = \omega_N^k$

Proof!

$$\omega_N = e^{-j\frac{2\pi}{N}}$$

$$\omega_N^{k+N} = e^{-j\frac{2\pi}{N}(k+N)}$$

$$= e^{-j\frac{2\pi}{N}k} \cdot e^{-j\frac{2\pi}{N}N} = 1$$

$$= e^{-j\frac{2\pi}{N}k}$$

$$= \omega_N^k$$

② $\omega_N^2 = \omega_{N/2}$

$$\omega_N = e^{-j\frac{2\pi}{N}}$$

replace N by $N/2$

$$\frac{\omega_N}{2} = e^{-j\frac{2\pi}{N/2}}$$

$$= e^{-j\frac{2\pi}{N} \cdot 2}$$

$$= \underline{\underline{\omega_N^2}}$$

③ Symmetry

$$\omega_N^{k+\frac{N}{2}} = -\omega_N^k$$

Proof!

$$\omega_N = e^{-j\frac{2\pi}{N}}$$

$$\omega_N^{k+\frac{N}{2}} = e^{-j\frac{2\pi}{N}(k+\frac{N}{2})}$$

$$= e^{-j\frac{2\pi}{N}k} \cdot e^{-j\frac{2\pi}{N} \cdot \frac{N}{2}}$$

$$= e^{-j\frac{2\pi}{N}k} \cdot e^{-j\pi}$$

$$e^{-j\pi} = -1 \text{ always}$$

$$= -e^{-j\frac{2\pi}{N}k}$$

$$= -\omega_N^k //$$

Q.3 Radix-2. FFT Algorithm:

- * By employing divide & conquer approach, a computationally efficient algorithm to evaluate DFT can be developed.
- * This approach depends on the decomposition of N-point DFT into successively smaller size DFTs.

* Let $N = r_1 \cdot r_2 \cdot r_3 \dots r_v$
 where $r_1 = r_2 = r_3 = \dots r_v = r$
 then

$$N = r^v$$

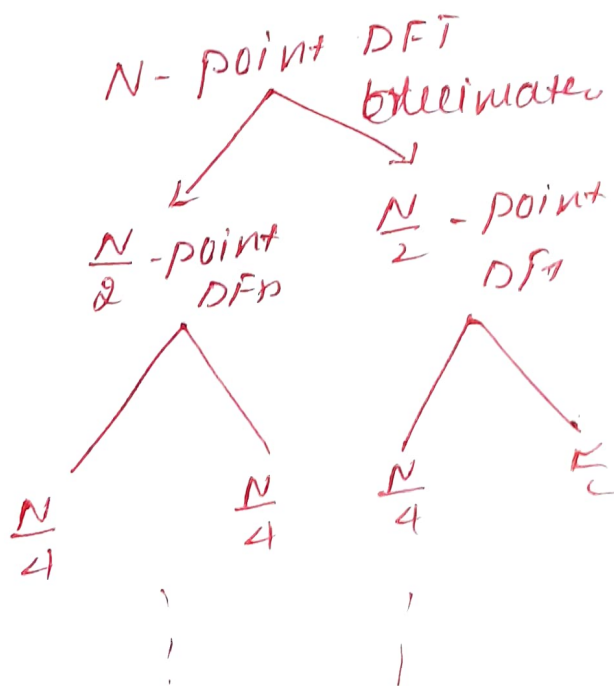
here the no r is called radix of FFT algorithm

- * here the most widely used is radix-2 FFT algorithms when $r=2$, ~~as~~ are explained.
- * There are 2 types of radix-2 FFT algorithms they are
 - (a) Radix-2: Decimation in Time FFT algorithm [DIT-FFT] algorithm
 - (b) Radix-2: Decimation in Frequency FFT algo. [DIF-FFT] algorithm

9.2

Decimation in Time - FFT Algorithm [DIT-FFT]

- Let us assume $x(n)$ is a length- N sequence & N is assumed as a power of 2 (N^V)
 Let $V=2$. $\therefore N^2$. $N = 2^P$
- In this algorithm, the N -point DFT is decimated (broken) into two $\frac{N}{2}$ -point DFTs.
 Each $\frac{N}{2}$ -point DFT is decimated into two $\frac{N}{4}$ -point DFTs & this decimation is continued until 2-point DFTs are obtained.



2-point DFTs are obtained

This approach is called as divide & conquer approach

* consider that the given length N seqⁿ $x(n) = \{ x(0), x(1), x(2), \dots, x(N-1) \}$

consider that the given length - N seqⁿ is

$$x(n) = \{ x(0), x(1), x(2), \dots, x(\frac{N}{2}-1), \dots, x(N-1) \}$$

I-stage:

Decimate this seqⁿ $x(n)$ into two seqⁿ of length $N/2$. one considered as even-indexed values of $x(n)$ & other as odd-indexed values of $x(n)$.

Even-indexed seqⁿ : $\{ x(0), x(2), x(4), \dots, x(N-2) \}$

Odd-indexed seqⁿ : $\{ x(1), x(3), \dots, x(N-1) \}$

WKT by defⁿ of DFT,

$$X(k) \triangleq \sum_{n=0}^{N-1} x(n) \omega_N^{kn} \rightarrow \textcircled{1}$$

$$0 \leq k \leq N-1$$

decimating $x(n)$ in eq $\textcircled{1}$ into even & odd indexed seqⁿs, we get

$$X(k) = \sum_{\substack{n=0 \\ n: \text{even}}}^{N-2} x(n) \omega_N^{kn} + \sum_{\substack{n=1 \\ n: \text{odd}}}^{N-1} x(n) \omega_N^{kn} \rightarrow \textcircled{2}$$

substituting $n = 2r$ in first summation & $n = 2r+1$ in second summation, we get

$$X(k) = \sum_{r=0}^{\frac{N}{2}-1} x(2r) \omega_N^{2kr} + \sum_{r=0}^{\frac{N}{2}-1} x(2r+1) \omega_N^{k(2r+1)} \rightarrow \textcircled{3}$$

NOTE: $\omega_N^{k2r} = \omega_{N/2}^{kr}$

$$= \sum_{r=0}^{\frac{N}{2}-1} g(r) \omega_N^{2kr} + \sum_{r=0}^{\frac{N}{2}-1} h(r) \omega_N^{2kr} \cdot \omega_N^k$$

$$= \sum_{r=0}^{\frac{N-1}{2}} g(r) \omega_N^{kr} + \omega_N^k \sum_{r=0}^{\frac{N-1}{2}} h(r) \omega_N^{kr} \rightarrow (4)$$

$\frac{N}{2}$ -point DFT of even indexed seqⁿ
 $\frac{N}{2}$ -point DFT of odd-indexed seqⁿ

if $G(k)$ & $H(k) \rightarrow \frac{N}{2}$ -point DFTs of even & odd indexed seqⁿ

$$0 \leq k \leq \frac{N}{2} - 1$$

& they are also periodic with a period $N/2$

we have

$$\begin{aligned} G(k) &= G(k - \frac{N}{2}) \\ &\& H(k) &= H(k - N/2) \end{aligned} \rightarrow (5)$$

eq (4) can be written as

$$X(k) = G(k) + \omega_N^k H(k) \rightarrow (6)$$

$$0 \leq k \leq \frac{N}{2} - 1$$

Using eq (5) \therefore for $\frac{N}{2} \leq k \leq N-1$

eq (6) can be written as

$$X(k) = G(k - \frac{N}{2}) + \omega_N^k H(k - \frac{N}{2}) \rightarrow (7)$$

$$; \frac{N}{2} \leq k \leq N-1$$

$$X(k) = \begin{cases} G(k) + \omega_N^k H(k), & 0 \leq k \leq \frac{N}{2} - 1 \\ G(k - \frac{N}{2}) + \omega_N^k H(k - \frac{N}{2}), & \frac{N}{2} \leq k \leq N-1 \end{cases}$$

consider $N = 2^4 \therefore N = 8$. Then $0 \leq k \leq 7$
 from eq (8) we get

$$X(k) = \begin{cases} G(k) + W_8^k H(k), & k=0, 1, \dots, \frac{N}{2}-1 \\ G(k+\frac{N}{2}) + W_8^k H(k+\frac{N}{2}), & k=\frac{N}{2}, \dots, N-1 \end{cases} \rightarrow \textcircled{9}$$

$$X(k) = \begin{cases} G(k) + W_8^k H(k), & k=0, 1, 2, 3 & \frac{N}{2}-1 = 4-1 = 3 \\ G(k+4) + W_8^k H(k+4), & k=4, 5, 6, 7 \end{cases} \rightarrow \textcircled{9}$$

$$X(0) = G(0) + W_8^0 H(0)$$

$$X(1) = G(1) + W_8^1 H(1)$$

$$X(2) = G(2) + W_8^2 H(2)$$

$$X(3) = G(3) + W_8^3 H(3)$$

$$X(4) = G(0) + W_8^4 H(0)$$

$$X(5) = G(1) + W_8^5 H(1)$$

$$X(6) = G(2) + W_8^6 H(2)$$

$$X(7) = G(3) + W_8^7 H(3).$$

using these sets of eqⁿ we obtain the flow graph after 1st stage decimation of 8-point DFT

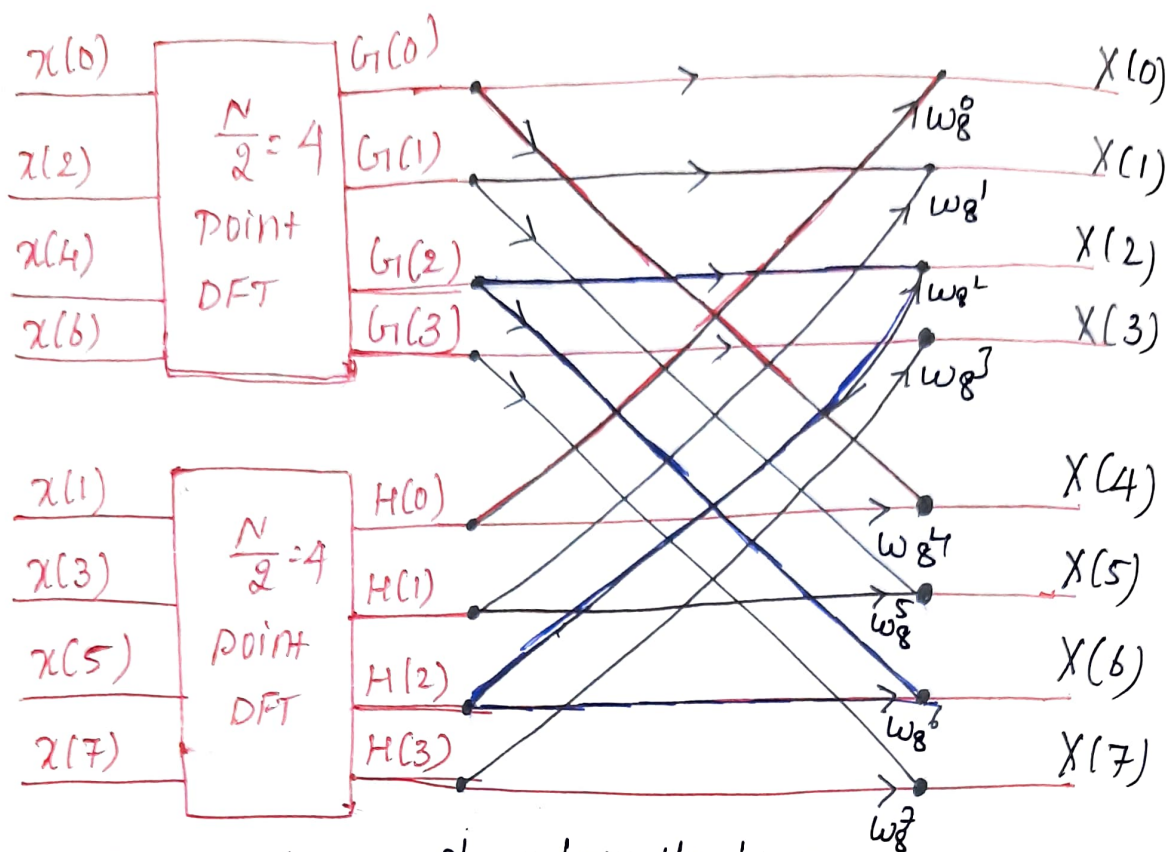


fig 0 Flow-graph after the first stage decomposition in DIT-FFT algorithm for $N=8$.

* In general, the no of complex multiplications required to evaluate the N -point DFT with the first stage decomposition is given by:

$$\begin{aligned}
 \alpha &= \left(\frac{N}{2}\right)^2 + \left(\frac{N}{2}\right)^2 + N \quad \text{no of complex xions required to multiply } \omega_N^k. \\
 &\quad \uparrow \quad \quad \quad \uparrow \\
 &\quad \text{no of} \quad \quad \quad \text{no of complex} \\
 &\quad \text{complex} \quad \quad \quad \text{multiplication required} \\
 &\quad \text{xions required} \quad \quad \quad \text{for direct computation} \\
 &\quad \text{for direct computation} \quad \quad \quad \text{of } \frac{N}{2} \text{ point DFT} \\
 &\quad \text{of } \frac{N}{2} \text{ point DFT} \quad \quad \quad \text{of } \frac{N}{2} \text{ point DFT} \\
 &\quad G(k) \quad \quad \quad \quad \quad \quad \quad H(k)
 \end{aligned}$$

$$= 2 \cdot \frac{N^2}{4} + N = \frac{N^2}{2} + N$$

* Thus the no of complex xions is reduced from N^2 to $\frac{N^2}{2} + N$

$$\begin{aligned}
 &8^2 = 64 \\
 &64 + 8 = 32 + 8 \quad 81
 \end{aligned}$$

IInd stage:- each $\frac{N}{2}$ point seqⁿ are further decimated into seqⁿ's of length $\frac{N}{4}$.

we have

$$G_1(K) = \sum_{n=0}^{N/2-1} g(n) W_{\frac{N}{2}}^{K n} \rightarrow (10)$$

decimated $g(n)$ of eq (10) into even & odd-indexed seqⁿ.

$$G_1(K) = \sum_{r=0}^{N/2-2} g(r) W_{\frac{N}{2}}^{K r} + \sum_{r=1}^{N/2-1} g(r) W_{\frac{N}{2}}^{K r} \rightarrow (11)$$

Substituting $r=2l$ in 1st summation & $r=2l+1$ in 2nd summation we get

$$G_1(K) = \sum_{l=0}^{N/4-1} g(2l) W_{\frac{N}{2}}^{2K l} + \sum_{l=0}^{N/4-1} g(2l+1) \frac{W_{\frac{N}{2}}^{K(2l+1)}}{2}$$

$$= \sum_{l=0}^{N/4-1} g(2l) W_{\frac{N}{2}}^{2K l} + \sum_{l=0}^{N/4-1} g(2l+1) W_{\frac{N}{2}}^{2K l} \cdot W_{\frac{N}{2}}^K$$

$$= \underbrace{\sum_{l=0}^{N/4-1} g(2l) W_{\frac{N}{4}}^{K l}}_{A(K)} + W_{\frac{N}{2}}^K \underbrace{\sum_{l=0}^{N/4-1} g(2l+1) W_{\frac{N}{4}}^{K l}}_{B(K)}$$

$$G_1(K) = A(K) + W_{\frac{N}{2}}^K B(K) \rightarrow (12)$$

$$0 \leq K \leq \frac{N}{4} - 1$$

$A(K) \rightarrow \frac{N}{4}$ -point DFT of even indexed seqⁿ of $g(n)$
 $B(K) \rightarrow \dots \dots \dots$ odd $\dots \dots \dots g(n)$

1114

$$H(K) = C(K) + W_{N/2}^K D(K) \rightarrow (13)$$

Since $A(K)$, $B(K)$, $C(K)$, & $D(K)$ are periodic with a period 4 we can write

$$G(K) = \begin{cases} A(K) + W_{\frac{N}{2}}^K B(K), & K = 0, 1, \dots, \frac{N}{4} - 1, \\ A(K + \frac{N}{4}) + W_{\frac{N}{2}}^K B(K + \frac{N}{4}), & K = \frac{N}{4}, \frac{N}{4} + 1, \dots, \frac{N}{2} - 1 \end{cases} \rightarrow (14)$$

$$H(K) = \begin{cases} C(K) + W_{\frac{N}{2}}^K D(K) & \dots K = 0, 1, \dots, \frac{N}{4} - 1 \\ C(K + \frac{N}{4}) + W_{\frac{N}{2}}^K D(K + \frac{N}{4}), & K = \frac{N}{4}, \dots, \frac{N}{2} - 1 \end{cases} \rightarrow (15)$$

for $N=8$.

$$G(K) = \begin{cases} A(K) + W_4^K B(K), & K = 0, 1 \\ A(K+2) + W_4^K B(K+2), & K = 2, 3 \end{cases}$$

$$H(K) = \begin{cases} C(K) + W_4^K D(K), & K = 0, 1 \\ C(K+2) + W_4^K D(K+2), & K = 2, 3 \end{cases}$$

$$G(0) = A(0) + W_4^0 B(0)$$

$$G(1) = A(1) + W_4^1 B(1)$$

$$G(2) = A(0) + W_4^2 B(0)$$

$$G(3) = A(1) + W_4^3 B(1)$$

$$H(0) = C(0) + W_4^0 D(0)$$

$$H(1) = C(1) + W_4^1 D(1)$$

$$H(2) = C(0) + W_4^2 D(0)$$

$$H(3) = C(1) + W_4^3 D(1)$$

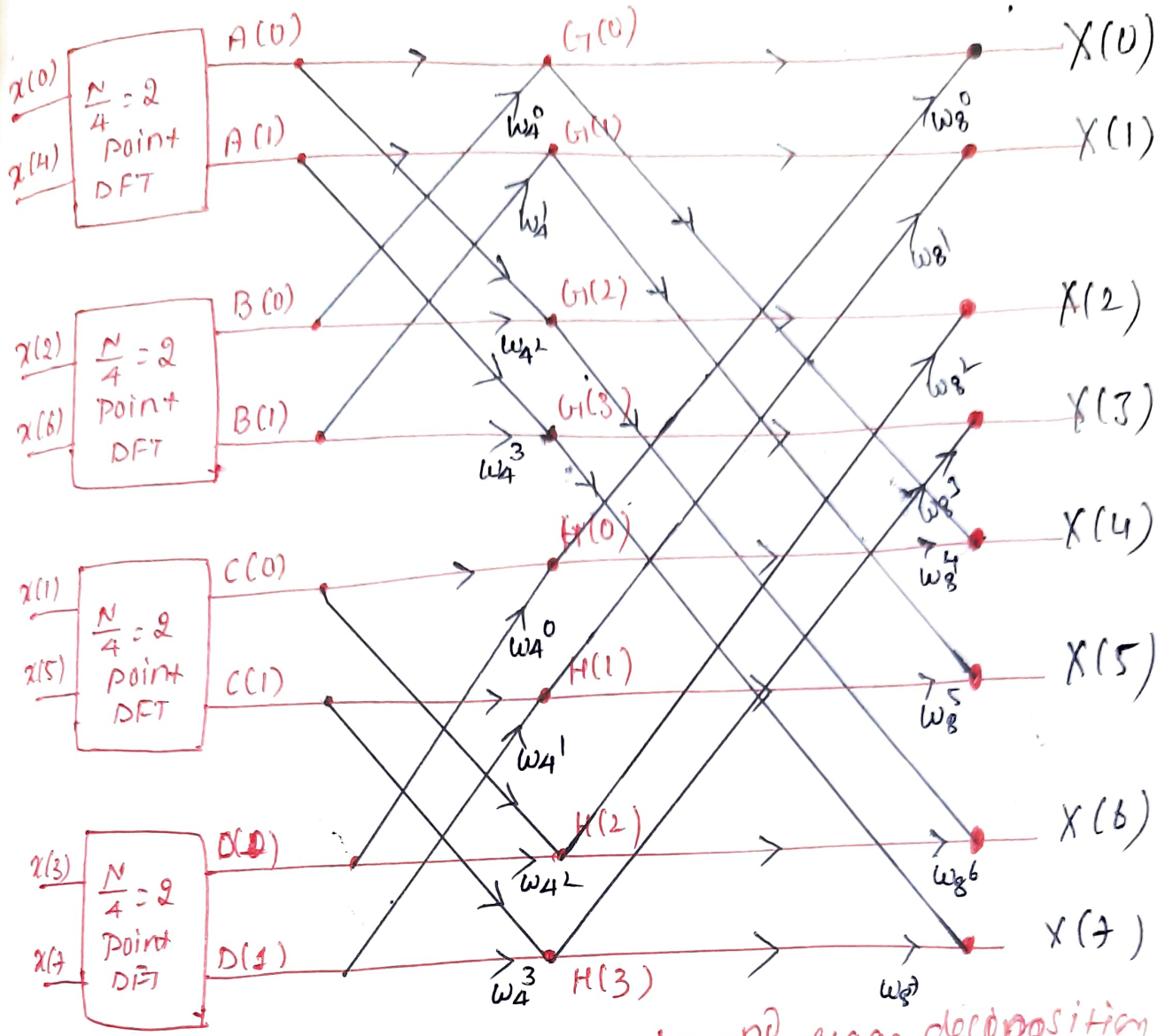


fig 2 flow-graph after the 2nd stage decimation in DIT-FFT algorithm for $N=8$

* In general, the no of complex multiplications required after 2nd stage decimation is given below.

$$\beta = \underbrace{\left(\frac{N}{4}\right)^2 + \left(\frac{N}{4}\right)^2 + \left(\frac{N}{4}\right)^2 + \left(\frac{N}{4}\right)^2}_{\text{no of complex xions required for direct computation of } 4 \times \frac{N}{4} \text{ point DFT}} + \underbrace{\left(\frac{N}{2}\right) + \left(\frac{N}{2}\right)}_{\text{no of complex xions required for multiply the factor } w_{N/2}^k} + N$$

$$= 4 \left(\frac{N}{4}\right)^2 + 2 \left(\frac{N}{2}\right) + N$$

$$\beta = \frac{N^2}{4} + 2N$$

* Continuing this process of decimation, we can represent each $\frac{N}{4}$ -point DFT as a combination of 2 two $\frac{N}{8}$ -point DFT & so on.

* N is a power of 2, i.e., $N = 2^V$
 this process is continued until there are $V = \log_2 N$ stages.

* In the above example for $N=8$, the 8-point DFT computation has been reduced to a computation of 2-point DFTs after the second stage decimation.

* Let us consider the 2-point DFT of $x(0)$ & $x(4)$ we have

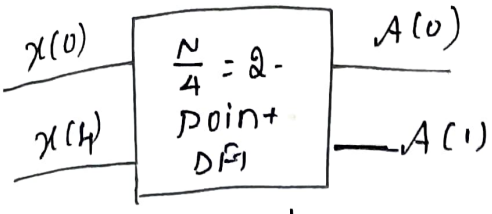
$$A(k) = \sum_{n=0}^{\frac{N}{4}-1} x(n) W_{\frac{N}{4}}^{kn} ; 0 \leq k \leq 1$$

$$= \sum_{n=0}^1 x(n) W_{\frac{N}{4}}^{kn}$$

$$A(0) = x(0) W_{\frac{N}{4}}^0$$

$$A(k) = x(0) W_{\frac{N}{4}}^0 + x(1) W_{\frac{N}{4}}^k$$

$$A(0) = x(0) + x(4) W_{N/4}$$



can be written as in fig (3)

$$A(k) = \sum_{n=0}^1 x(n) w_{\frac{N}{4}}^{kn}$$

$$= x(0) w_{\frac{N}{4}}^0 + x(1) w_{\frac{N}{4}}^k$$

$$A(k) = x(0) + x(1) w_2^k$$

$$A(0) = x(0) + w_2^0 x(1) \rightarrow (16)$$

$$A(1) = x(0) + w_2^1 x(1) \rightarrow (17)$$

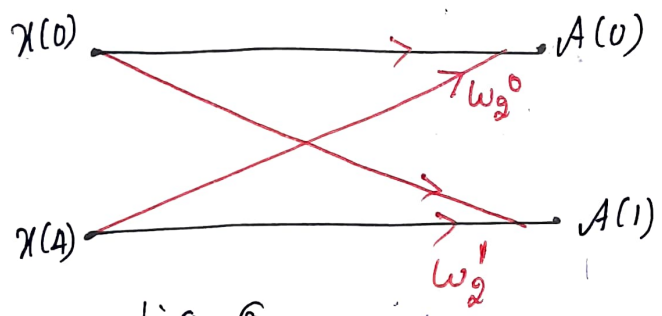


fig (3).

Inserting the flow-graph shown in fig (3) in fig (2) The complete flow-graph for computation of 8-point DFT is as shown in fig (4)

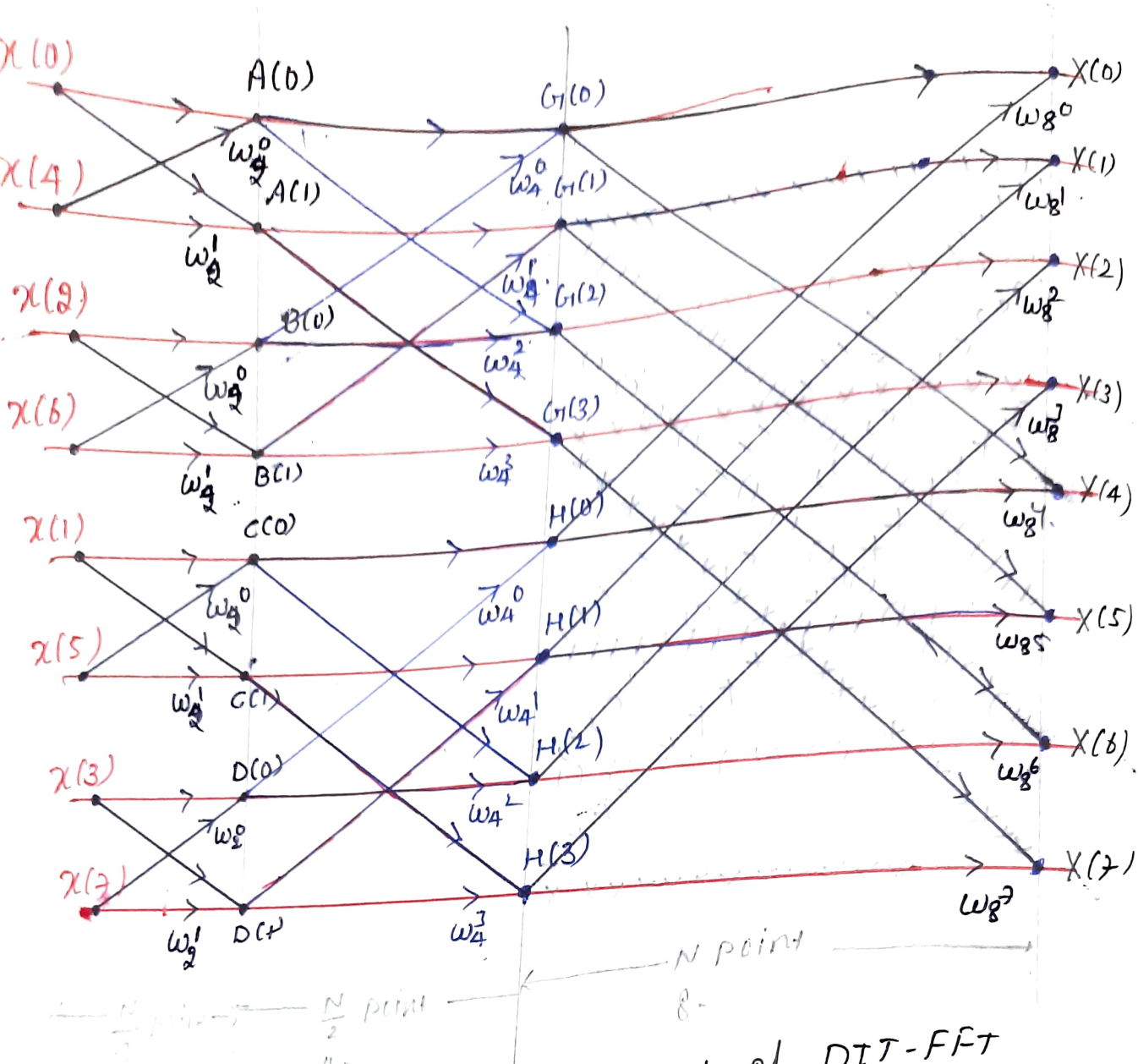


fig (4) The complete flow graph of DIT-FFT Algorithm for $N=8$

- * from the flow graph shown above
- * for each stage there are 8 complex multiplications & 8 complex additions.
- * In general for computation of N -point DFT using DIT-FFT alg, there are N complex multiplication & N complex additions for each stage
- * since there are $V = \log_2^N$ stages. A total of $\therefore N \log_2^N$ complex mults & additions are required to compute the N -point DFT using DIT-FFT alg

* In-place Computation :-

from fig (4) we have the foll points:

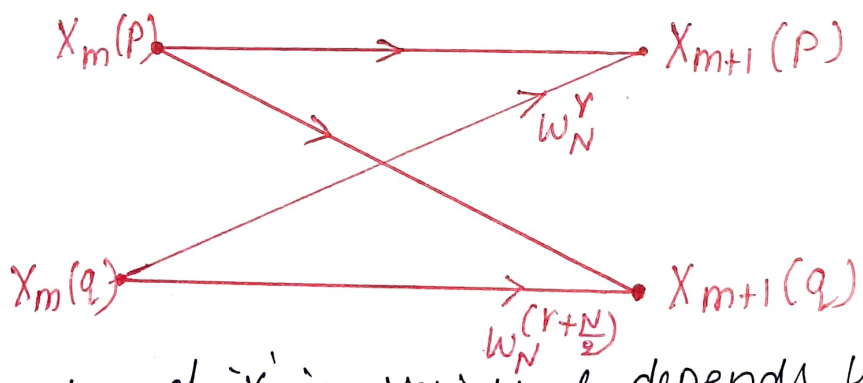
(a) The input data appear in 'bit reversed' order as illustrated below

	bit reverse
000	000 → 0
001	100 → 4
010	010 → 2
011	110 → 6
100	001 → 1
101	101 → 5
110	011 → 3
111	111 → 7

(b) Each basic computational block in the diagram is called a 'butterfly' because of its diagram

(c) The N-point DFT $X(k)$ are appears in the normal order (or) freq domain (or)

* if ~~w~~ let $m \rightarrow$ represent the stage
 p & $q \rightarrow$ position no. in the stages
 each butterfly in fig (4) can be represented as shown in fig (5) below



* The value of 'r' is variable & depends upon the position of the butterfly.

* The o/p's $X_{m+1}(p)$ & $X_{m+1}(q)$ of the butterfly at stage (m+1) are calculated in terms of $X_m(p)$ & $X_m(q)$ which are the o/p values from the mth stage & no other i/p

$$X_{m+1}(p) = X_m(p) + W_N^r \cdot X_m(q) \rightarrow 18$$

$$X_{m+1}(q) = X_m(p) - W_N^{(r+N/2)} \cdot X_m(q) \rightarrow 19$$

* This kind of computation is known as in-place computation.

* Further Reduction: (using Cooley-Tukey algorithm)

from eq (18) & (19) we have

$$X_{m+1}(p) = X_m(p) + W_N^r X_m(q)$$

$$X_{m+1}(q) = X_m(p) + W_N^{(r+\frac{N}{2})} X_m(q)$$

Also we have

$$W_N^{(r+\frac{N}{2})} = W_N^r \cdot W_N^{N/2}$$

$$W_N^{N/2} = e^{-j2\pi \cdot \frac{N}{2} \cdot \frac{1}{N}} = e^{-j2\pi} = -1 \text{ always}$$

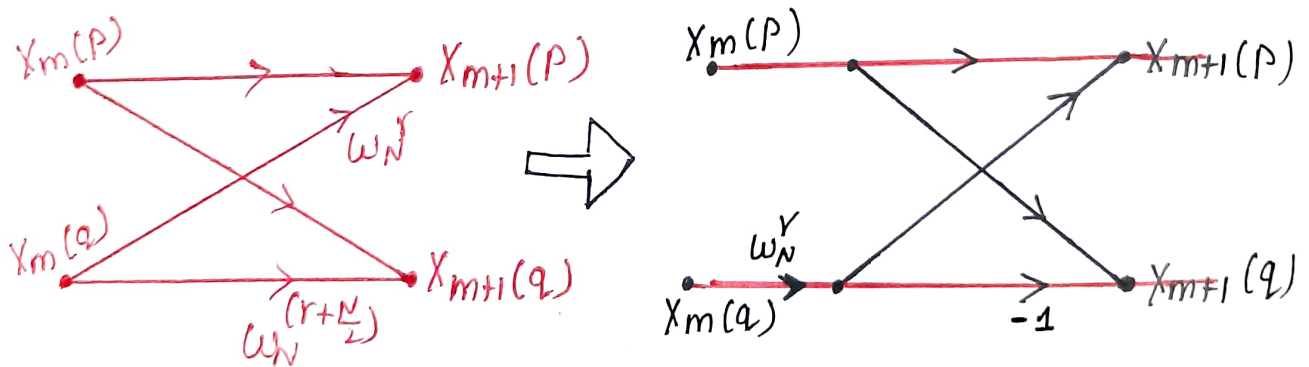
$$W_N^{(r+\frac{N}{2})} = -W_N^r \rightarrow (20)$$

∴ eq (18) & (19) can be written as

$$X_{m+1}(p) = X_m(p) + W_N^r X_m(q) \rightarrow (21)$$

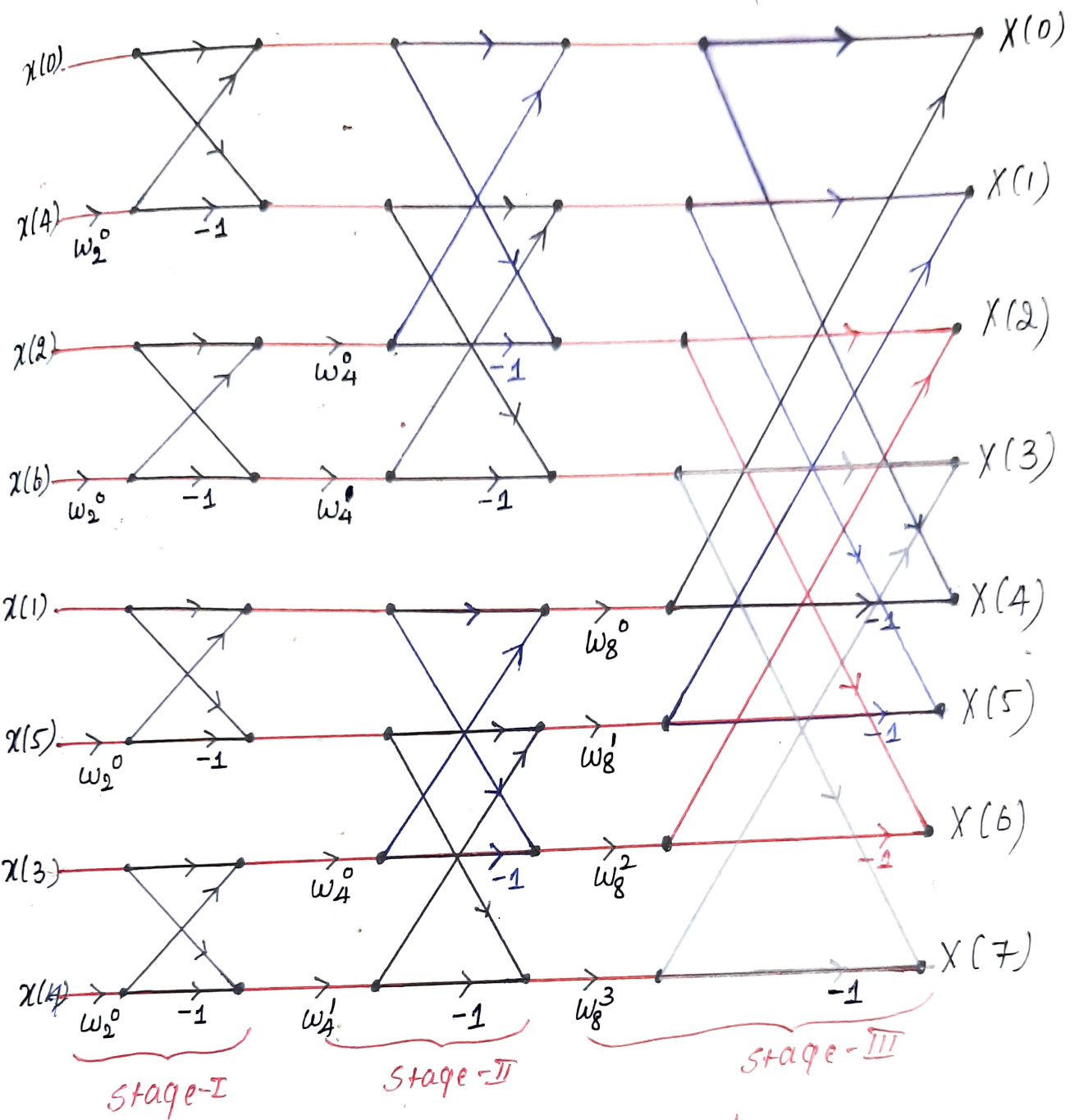
$$X_{m+1}(q) = X_m(p) - W_N^r X_m(q) \rightarrow (22)$$

using eq (21) & (22) the butterfly shown in fig (5) can be modified as shown in fig (6).



* After further reduction the total no of complex multiplications are reduced to $\frac{N}{2} \times \log_2 N$

* The reduced complete - 8-point DIT-FFT Alg is shown in fig (7) below.



8-point DIT-FFT flow graph

Note: - $w_2^0 = w_4^0 = w_8^0$
 $w_4^1 = w_8^2$

$$w_2^0 = 1$$

$$w_4^0 = 1$$

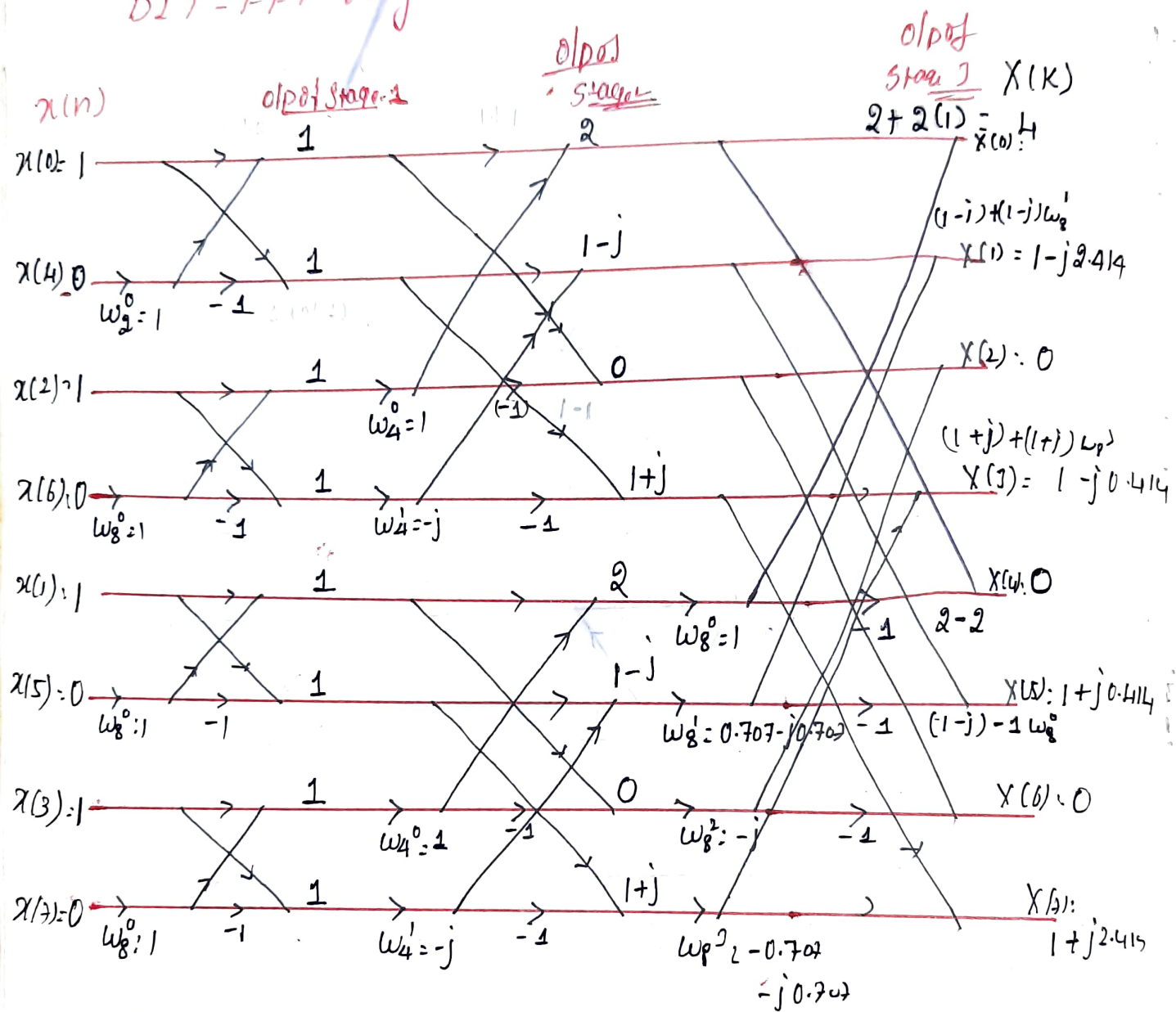
$$w_8^0 = 1$$

$$w_4^1 = -j$$

$$w_8^1 = 0.707 - j0.707$$

$$w_8^2 = -j ; w_8^3 = -0.707 - j0.707$$

(1) Compute 8-point DFT of the seqn $x(n)$
 $x(n) = \{1, 1, 1, 1, 0, 0, 0, 0\}$ using
 DIT-FFT Algorithm.



$$\begin{array}{r} 0.707 \\ 0.707 \\ \hline 1.414 \end{array}$$

$$\begin{aligned} &(1-j) + (0.707-j0.707)(1-j) \\ &1-j + 0.707 - j0.707 - 0.707j - 0.707 \\ &1 - j2.414 \end{aligned}$$

$$X(K) = \{ 4, 1-j2.414, 0, 1-j0.414, 0, 1+j0.414, 0, 1+j2.414 \}$$

② find the 8-point DFT of the seqn
 $x(n) = 2^n, 0 \leq n \leq 7$, use DIT-FFT Alg.

Solⁿ
 $x(n) = \{0, 2, 4, 8, 16, 32, 64, 128\}$

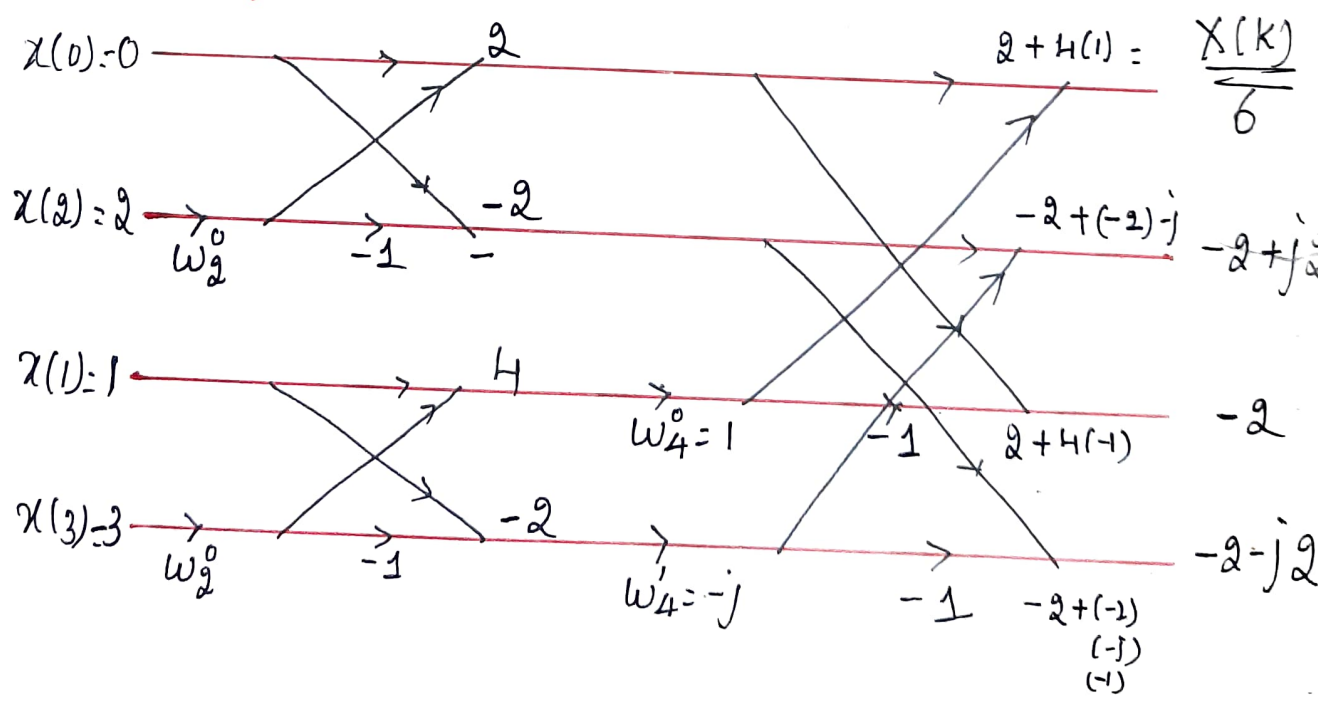
o/p of I-stage: $\{17, -15, 68, -60, 34, -30, 136, -120\}$

o/p of II-stage: $\{85, -15+j60, -51, -15+j60, 170, -30+j120, -102, -30+j120\}$

o/p of III stage

$$X(k) = \left\{ \begin{array}{l} 255, 48.63 + j166.05, -51 + j102, \\ -78.63 + j46.05, -85, -78.63 - j46.05 \\ -51 - j102, 48.63 - j166.05 \end{array} \right\}$$

③ given $x(n) = \{0, 1, 2, 3\}$ find its DFT using DIT-FFT Alg. $N=4$



$\therefore X(k) = \{6, -2+j2, -2, -2-j2\}$

Decimation in frequency

DIF-FFT Algorithm :-

The ~~the~~ ~~the~~ DIT algorithm is based on the decomposition of the DFT computation by forming smaller & smaller subsequences of seqⁿ $x(n)$. where as in DIF-algorithm, the o/p seqⁿ $X(k)$ is divided into smaller & smaller subseqs.

I-stage:- The decimation in freq-FFT Algorithm is obtained by breaking $X(k)$ as follows:

$$X(k) = \sum_{n=0}^{N-1} x(n) \omega_N^{kn} \quad ; \quad 0 \leq k \leq N-1 \quad \text{--- (1)}$$

splitting the i/p seqⁿ into a length of $\frac{N}{2}$ samples

$$X(k) = \sum_{n=0}^{\frac{N}{2}-1} x(n) \omega_N^{kn} + \sum_{n=\frac{N}{2}}^{N-1} x(n) \omega_N^{kn}$$

substitute ~~let~~ $r = n - \frac{N}{2}$ in second summation

$$= \sum_{n=0}^{\frac{N}{2}-1} x(n) \omega_N^{kn} + \sum_{r=0}^{\frac{N}{2}-1} x(r + \frac{N}{2}) \cdot \omega_N^{k(r + \frac{N}{2})}$$

since r is a dummy variable, it can be replaced by n we get

$$= \sum_{n=0}^{\frac{N}{2}-1} x(n) \omega_N^{kn} + \sum_{n=0}^{\frac{N}{2}-1} x(n + \frac{N}{2}) \cdot \omega_N^{k(n + \frac{N}{2})}$$

$$= \sum_{n=0}^{\frac{N}{2}-1} x(n) \omega_N^{kn} + \sum_{n=0}^{\frac{N}{2}-1} x(n + \frac{N}{2}) \cdot \omega_N^{kn} \cdot \omega_N^{k \cdot \frac{N}{2}} \quad \text{--- (2)}$$

WKT $W_N^{KN/2} = e^{-j2\pi \cdot K \cdot \frac{N}{2}} = e^{-j2\pi} = (-1)^K$

∴ eq (2) reduces to

$$X(K) = \sum_{n=0}^{\frac{N}{2}-1} x(n) W_N^{Kn} + \sum_{n=0}^{\frac{N}{2}-1} x(n + \frac{N}{2}) W_N^{Kn} (-1)^K$$

$$= \sum_{n=0}^{\frac{N}{2}-1} [x(n) + (-1)^K x(n + \frac{N}{2})] W_N^{Kn} \rightarrow (3)$$

The decimation in freq is now obtained by getting even & odd terms of $X(K)$.

(i) even term: substitute $K = 2r$ in eq (3)

~~$X(K) = \dots$~~

$$X(2r) = \sum_{n=0}^{\frac{N}{2}-1} [x(n) + (-1)^{2r} x(n + \frac{N}{2})] W_N^{2rn}$$

$[(-1)^{2r} = 1 \forall r]$
 & $W_N^{2rn} = W_{N/2}^{rn}$

$$= \sum_{n=0}^{\frac{N}{2}-1} [x(n) + x(n + \frac{N}{2})] W_{N/2}^{rn}$$

$$X(2r) = \sum_{n=0}^{N/2-1} g(n) W_{\frac{N}{2}}^{rn} \quad 0 \leq r \leq \frac{N}{2}-1 \rightarrow (4)$$

iii) $k = 2r+1$ in eq (3)

$$X(2r+1) = \sum_{n=0}^{\frac{N}{2}-1} \left[x(n) + (-1)^{2r+1} x\left(n + \frac{N}{2}\right) \right] W_N^{(2r+1)n}$$

$$(-1)^{2r+1} = -1 \quad \forall r$$

$$W_N^{(2r+1)n} = W_N^{2rn} \cdot W_N^n$$

$$\therefore \sum_{n=0}^{\frac{N}{2}-1} \left[x(n) - x\left(n + \frac{N}{2}\right) \right] W_N^{\frac{rn}{2}} \cdot W_N^n$$

$$X(2r+1) = \sum_{n=0}^{\frac{N}{2}-1} h(n) W_N^{\frac{rn}{2}} ; \quad 0 \leq r \leq \frac{N}{2}-1 \quad \rightarrow (5)$$

$$\begin{aligned} g(n) &= x(n) + x\left(n + \frac{N}{2}\right) \\ h(n) &= \left[x(n) - x\left(n + \frac{N}{2}\right) \right] W_N^n \end{aligned} \quad \rightarrow (6)$$

for $N=8$ eq (6)

$$g(n) = x(n) + x(n+4)$$

$$g(0) = x(0) + x(4)$$

$$g(1) = x(1) + x(5)$$

$$g(2) = x(2) + x(6)$$

$$g(3) = x(3) + x(7)$$

($n = 0$ to 3)

becomes

$$h(n) = \left[x(n) - x(n+4) \right] W_8^n$$

$$h(0) = \left[x(0) - x(4) \right] W_8^0$$

$$h(1) = \left[x(1) - x(5) \right] W_8^1$$

$$h(2) = \left[x(2) - x(6) \right] W_8^2$$

$$h(3) = \left[x(3) - x(7) \right] W_8^3$$

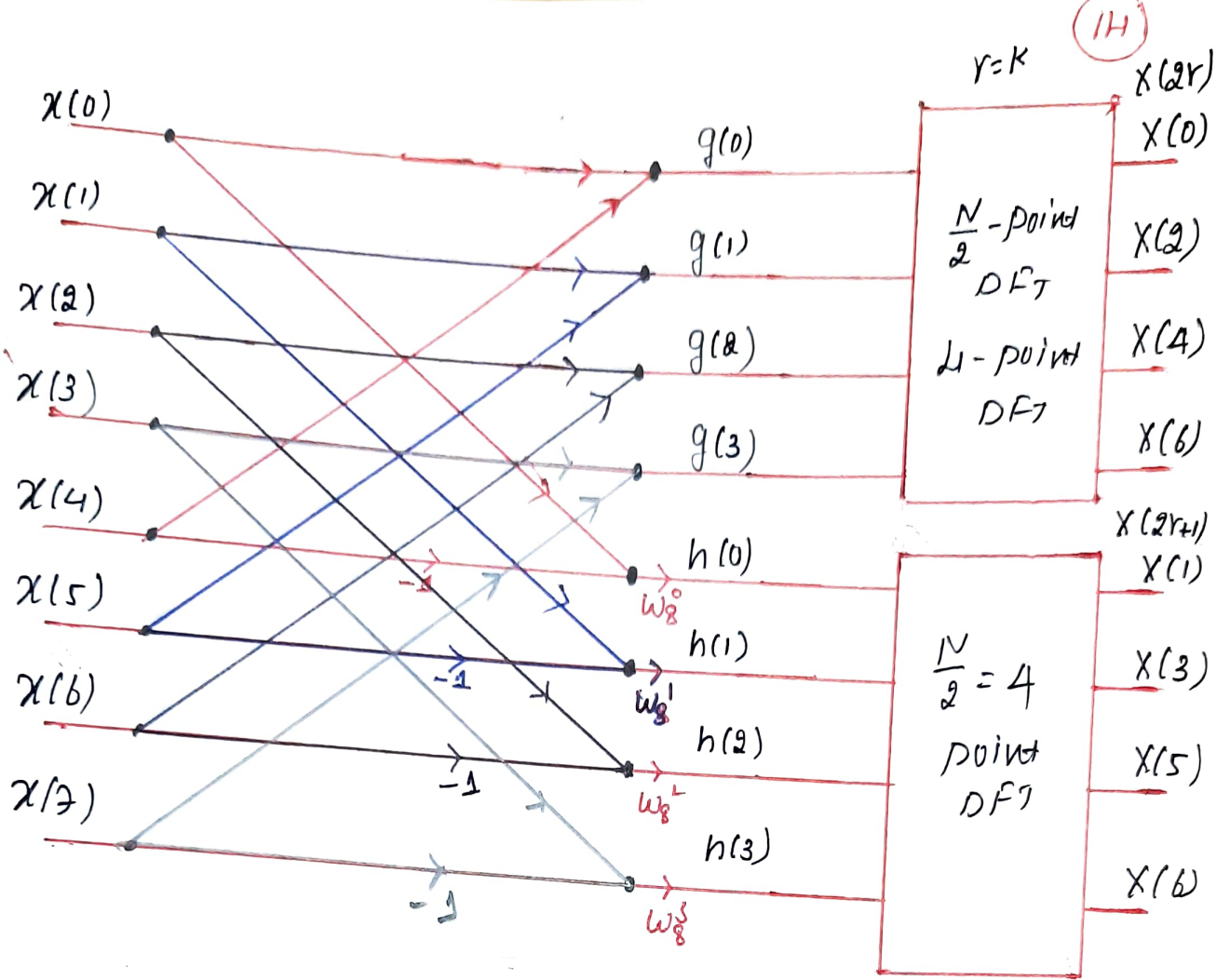


fig ① signal flow graph after 1st stage decimation of DIF-FFT Alg for $N=8$

* no of complex xlions = $2 \cdot \left(\frac{N}{2}\right)^2$

II - stage decimation

consider eq ④

$$X(2r) = \sum_{n=0}^{\frac{N}{2}-1} g(n) w_{\frac{N}{2}}^{rn} ; 0 \leq r \leq \frac{N}{2}-1$$

$$g(n) = x(n) + x\left(n + \frac{N}{2}\right)$$

splitting the ip seq $g(n)$ of length $\frac{N}{4}$

$$= \sum_{n=0}^{\frac{N}{4}-1} g(n) w_{\frac{N}{2}}^{rn} + \sum_{n=\frac{N}{4}}^{\frac{N}{2}-1} g(n) w_{\frac{N}{2}}^{rn}$$

substituting $x' = n - \frac{N}{4}$ in 2nd summation

$$\therefore n = x' + \frac{N}{4}$$

$$X(2r) = \sum_{n=0}^{\frac{N}{4}-1} g(n) w_{\frac{N}{2}}^{rn} + \sum_{x'=0}^{\frac{N}{4}-1} g(x' + \frac{N}{4}) w_{\frac{N}{2}}^{r(x' + \frac{N}{4})}$$

Since x' is a dummy variable, it can be replaced by n

$$= \sum_{n=0}^{\frac{N}{4}-1} g(n) w_{\frac{N}{2}}^{rn} + \sum_{n=0}^{\frac{N}{4}-1} g(n + \frac{N}{4}) \cdot w_{\frac{N}{2}}^{rn} \cdot w_{\frac{N}{2}}^{r \cdot \frac{N}{4}}$$

$$w_{\frac{N}{2}}^{rn} =$$

$$w_{\frac{N}{2}}^{r \cdot N/4} = w_N^{r \cdot 2 \cdot \frac{N}{4}} = w_N^{r \cdot \frac{N}{2}} = e^{-j2\pi \cdot \frac{nr}{N} \cdot \frac{N}{2}} = e^{-j\pi nr}$$

$$= (-1)^r$$

$$= \sum_{n=0}^{\frac{N}{4}-1} g(n) w_{\frac{N}{2}}^{rn} + \sum_{n=0}^{\frac{N}{4}-1} g(n + \frac{N}{4}) w_{\frac{N}{2}}^{rn} \cdot (-1)^r$$

$$X(2r) = \sum_{n=0}^{\frac{N}{4}-1} [g(n) + (-1)^r g(n + \frac{N}{4})] w_{\frac{N}{2}}^{rn} \rightarrow \textcircled{7}$$

substituting $r = 2r$ in eq $\textcircled{7}$ to get even terms

$$X(4r) = \sum_{n=0}^{\frac{N}{4}-1} [g(n) + (-1)^{2r} g(n + \frac{N}{4})] w_{\frac{N}{2}}^{2r \cdot n}$$

$$[(-1)^{2r} = 1 \quad \forall r$$

$$w_{\frac{N}{2}}^{2rn} = w_{\frac{N}{4}}^{rn}]$$

$$= \sum_{n=0}^{\frac{N}{4}-1} [g(n) + g(n + \frac{N}{4})] w_{\frac{N}{4}}^{rn}$$

$$X(4r) = \sum_{n=0}^{\frac{N}{4}-1} a(n) W_{\frac{N}{4}}^{rn} \quad 0 \leq r \leq \frac{N}{4}-1 \quad \rightarrow (8)$$

to get odd term substitute $r=2r+1$ in eq (7)

$$X(4r+2) = \sum_{n=0}^{\frac{N}{4}-1} \left[g(n) + (-1)^{2r+1} g\left(n + \frac{N}{4}\right) \right] W_{N/2}^{(2r+1)n}$$

$$= \sum_{n=0}^{\frac{N}{4}-1} \left[g(n) + (-1)^{2r+1} g\left(n + \frac{N}{4}\right) \right] W_{N/2}^{2rn} \cdot W_{N/2}^n$$

$$(-1)^{2r+1} = -1 \quad \forall r$$

$$W_{N/2}^{2rn} = W_{N/4}^{rn}$$

$$W_{N/2}^n = W_N^{2n}$$

$$\therefore \sum_{n=0}^{\frac{N}{4}-1} \left[g(n) - g\left(n + \frac{N}{4}\right) \right] \left(W_{\frac{N}{2}}^n \right) W_{\frac{N}{4}}^{rn}$$

$\rightarrow W_N^{2n}$

$$X(4r+4) = \sum_{n=0}^{\frac{N}{4}-1} b(n) \cdot W_{\frac{N}{4}}^{rn} \quad 0 \leq r \leq \frac{N}{4}-1 \quad \rightarrow (9)$$

$$b(n) = \left[g(n) - g\left(n + \frac{N}{4}\right) \right] W_{\frac{N}{2}}^n \quad \rightarrow (10)$$

$$a(n) = \left[g(n) + g\left(n + \frac{N}{4}\right) \right]$$

iii) consider eq (5) $X(4r+1)$ & $X(4r+3)$

$$c(n) = \left[h(n) + h\left(n + \frac{N}{4}\right) \right]$$

$$d(n) = \left[h(n) - h\left(n + \frac{N}{4}\right) \right] W_{\frac{N}{2}}^n \quad \rightarrow (11)$$

$$\underline{N=8}$$

$$n = 0 + 0 \frac{N}{4} - 1 = 0 + 0 \frac{8}{4} - 1$$

$$a(n) = g(n) + g(n + \frac{N}{4}) = g(n) + g(n+2) = 0 - 1$$

$$a(0) = g(0) + g(2)$$

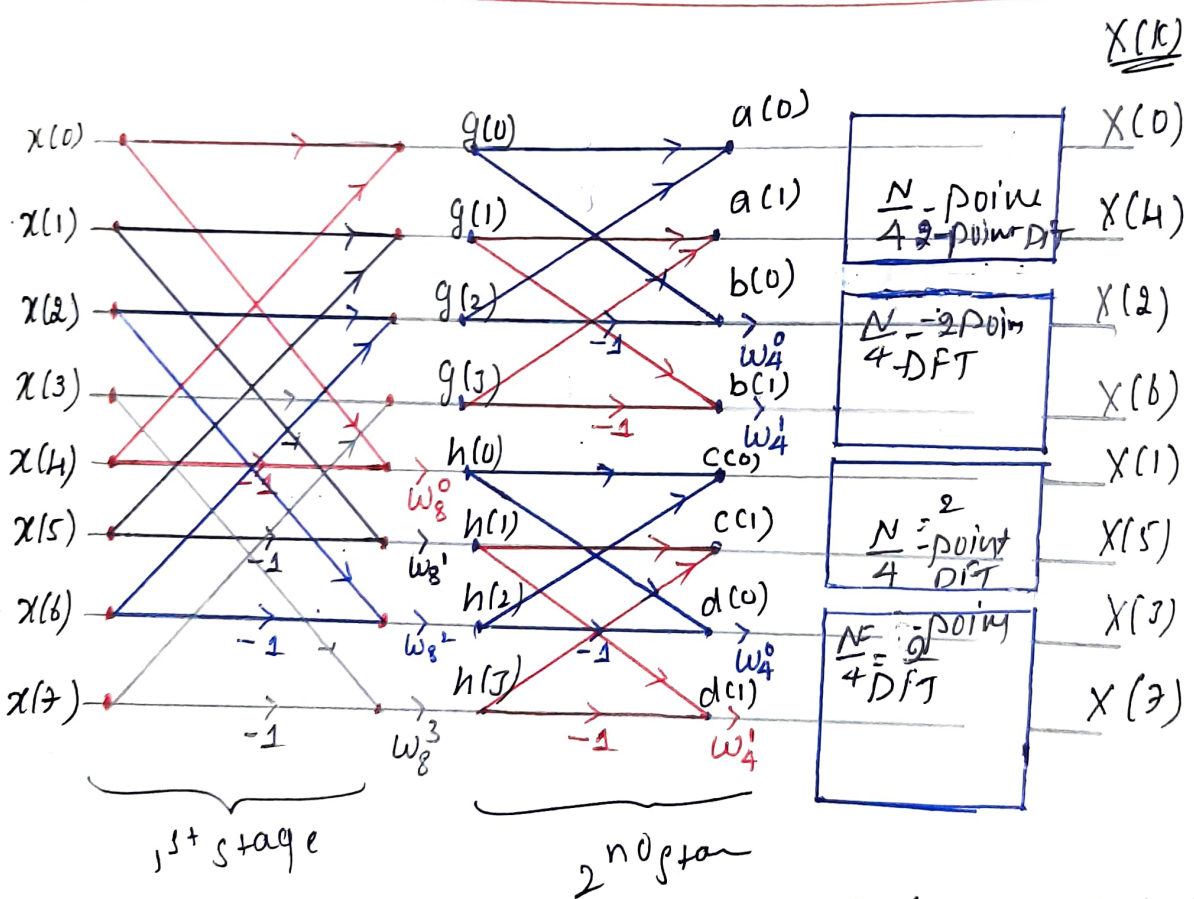
$$a(1) = g(1) + g(3)$$

$$b(n) = [g(n) - g(n + \frac{N}{4})] w_{\frac{N}{4}}^{n} = [g(n) - g(n+2)] w_4^n$$

$$b(0) = [g(0) - g(2)] w_4^0 \quad b(1) = [g(1) - g(3)] w_4^1$$

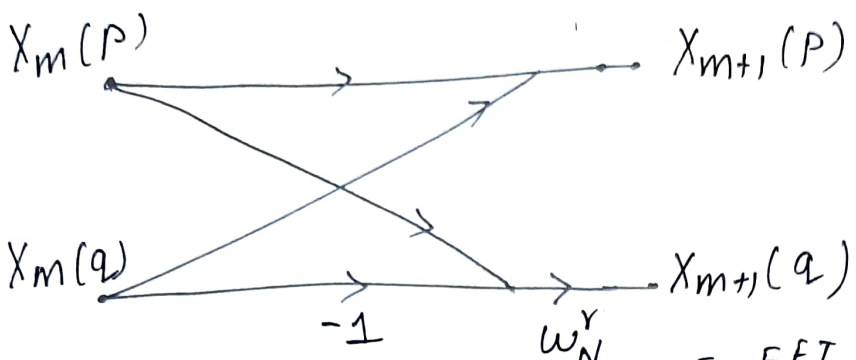
$$c(0) = h(0) + h(2) \quad c(1) = h(1) + h(3)$$

$$d(0) = [h(0) - h(2)] w_4^0 \quad d(1) = [h(1) - h(3)] w_4^1$$



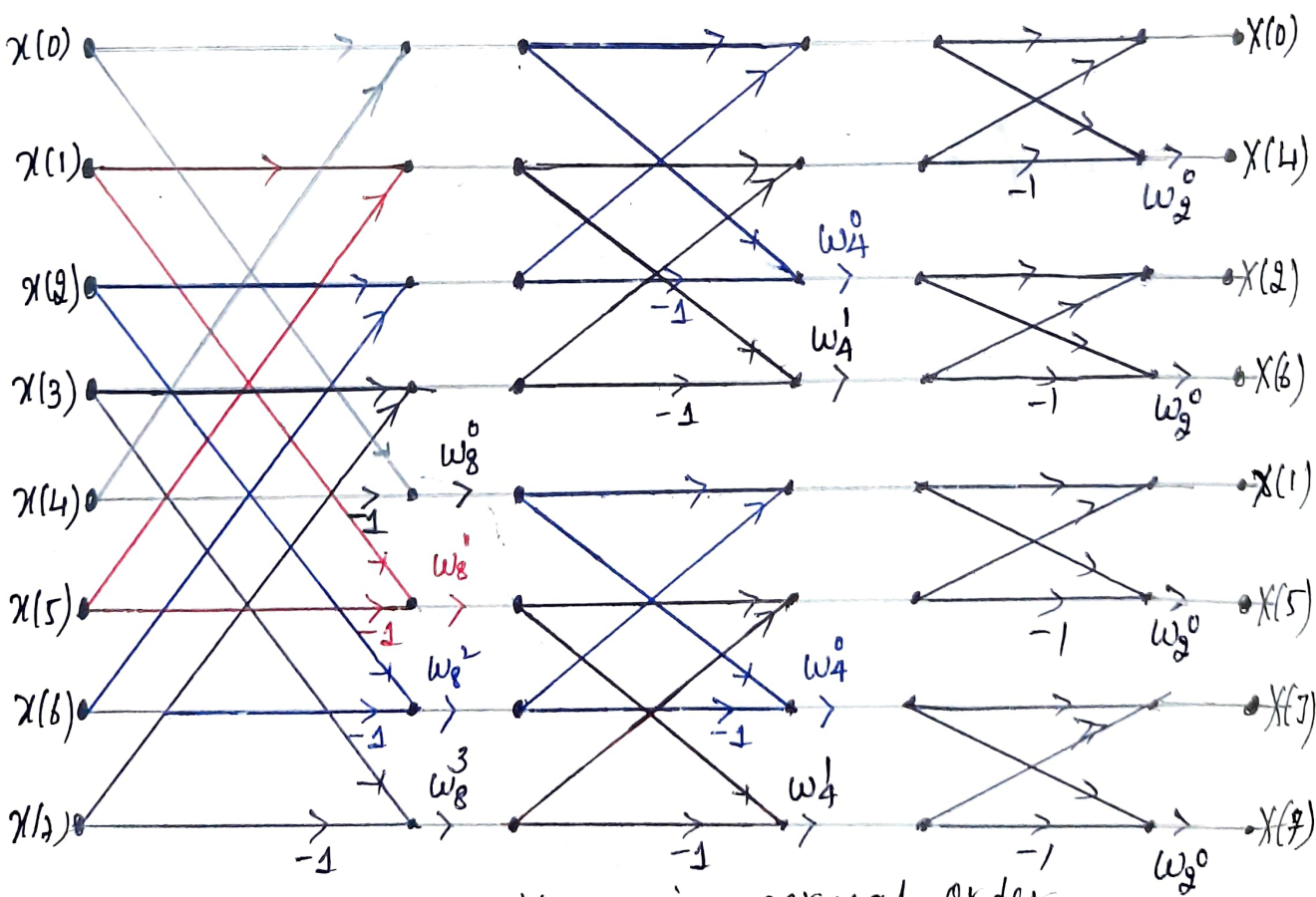
In similar way each $\frac{N}{4}$ -point DFT is decimated into two $\frac{N}{8}$ -point DFT. Continuing this process, down to 2-point DFT, the N-point DFT will be available at the output of 2-point transform in bit-reversed order 99

The complete process ~~consists~~ consists of $V = \log_2^N$ stages of decimation, where each stage involves $\frac{N}{2}$ butterfly of the type shown below



basic butterfly for DIF-FFT Alg

* The DIF-FFT Alg requires $\frac{N}{2} \log_2^N$ complex additions & $N \log_2^N$ complex additions



* we observe that i/p is in normal order while as the o/p is in bit reversed order

*
$$X_{m+1}(p) = X_m(p) + X_m(q)$$
$$X_{m+1}(q) = [X_m(p) - X_m(q)] w_N^r$$

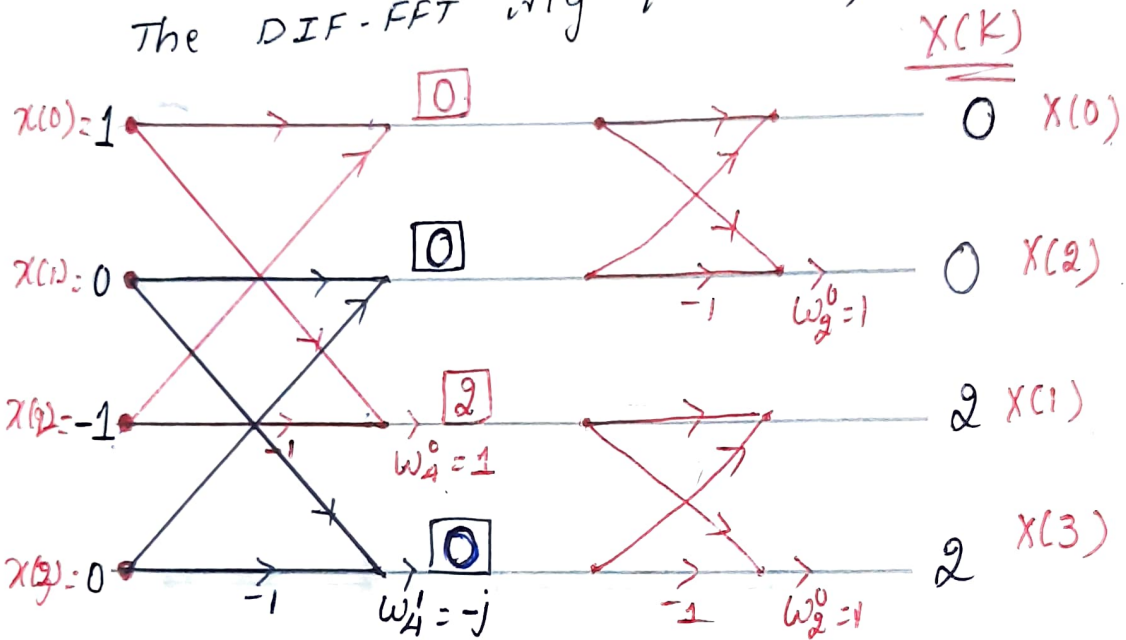
(i) Compute the DFT of the seqⁿ $x(n) = \cos n\frac{\pi}{2}$, where $N=4$ using DIF-FFT Alg.

Soln:

$$x(n) = \cos n\frac{\pi}{2}$$

$$x(n) = \{1, 0, -1, 0\}$$

The DIF-FFT Alg for $N=4$



$$X(k) = \{0, 2, 0, 2\}$$

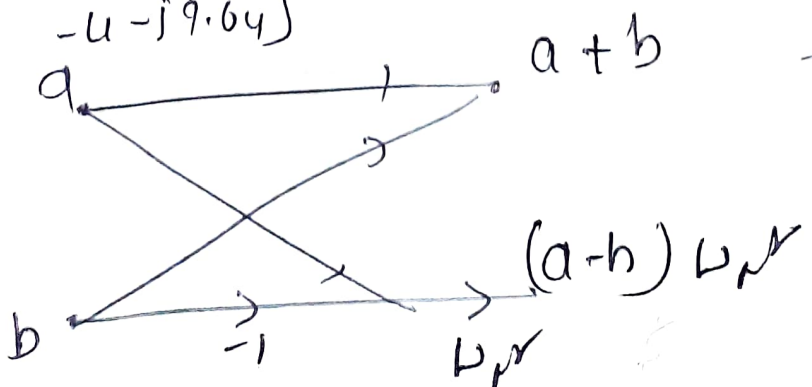
(2) $x(n) = n+1 \quad 0 \leq n \leq 7 \quad \underline{N=8}$

$$w_8^0 = 1, \quad w_8^1 = 0.707 - j0.707, \quad w_8^2 = -j$$

$$w_8^3 = -0.707 - j0.707$$

$$x(n) = \{1, 2, 3, 4, 5, 6, 7, 8\}$$

$$X(k) = \{36, -4 + j9.64, -4 + j, -4 + j1.64, -4, -4 - j1.64, -4 - j, -4 - j9.64\}$$



INVERSE DFT USING FFT ALGORITHM:-

We have

$$\text{IDFT } x(n) \triangleq \frac{1}{N} \sum_{k=0}^{N-1} X(k) W_N^{-kn} ; 0 \leq n \leq N-1$$

& ~~can~~ DFT $X(k) \triangleq \sum_{n=0}^{N-1} x(n) W_N^{kn} ; 0 \leq k \leq N-1$

by comparing eq ① & ② we find that the only difference bet' 2 eqn's is the factor $\frac{1}{N}$ & sign of power of factor W_N .

∴ an FFT Alg for computing DFT can be used to converted to an FFT algorithm for computing the IDFT by making the foll changes.

- (i) Reverse the diren of flow-graph
- (ii) change the sign of the power of the factor W_N .
- (iii) Replace $x(n)$ by $X(k)$ & vice versa
- (iv) multiply the o/p by factor $\frac{1}{N}$

→ I - { 6, 8, 10, 12, -4, -2.828 + j2.828, 4j, 2.828 + j2.828 }

II - { 16, 20, -4, 4j, -4 + 4j, 5.656j, -4, -4j, 5.656j }

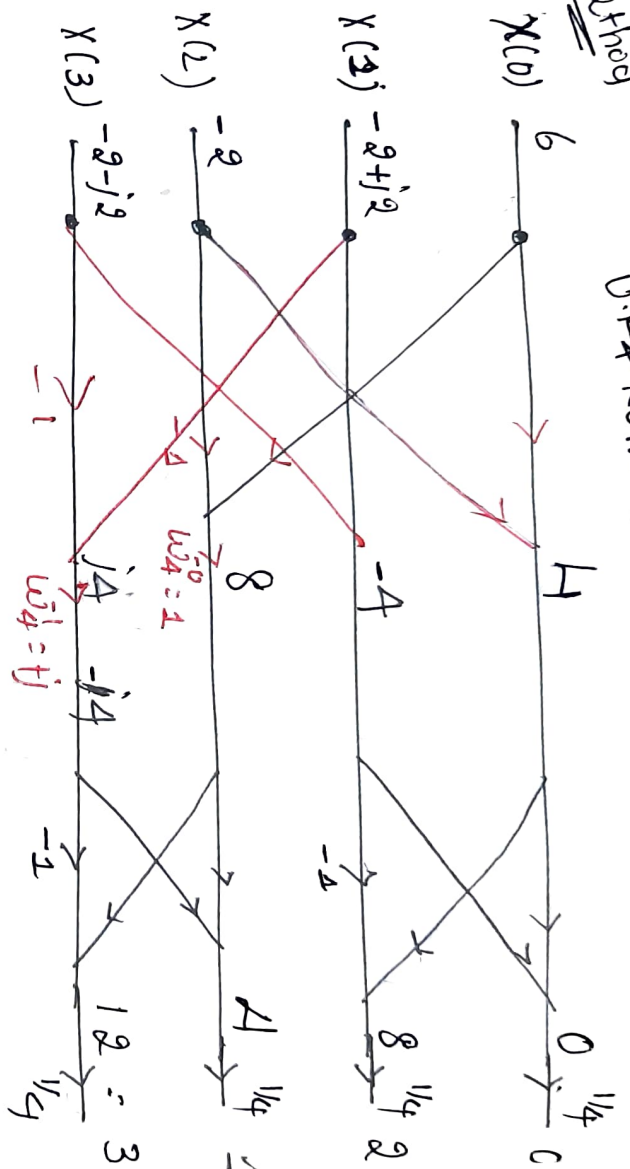
III - { 36, -4, -4 + 4j, -4 - 4j, -4 + j9.64, -4 - j9.64, -4 + j1.04, -4 - j1.04 }

Q1 Find the IDFT of the seqⁿ given below

using DIT-FFT Alg $X(K) = \{6, -2+j2, -2, -2-j2\}$

U.K.R Kunal Swamy

I-method



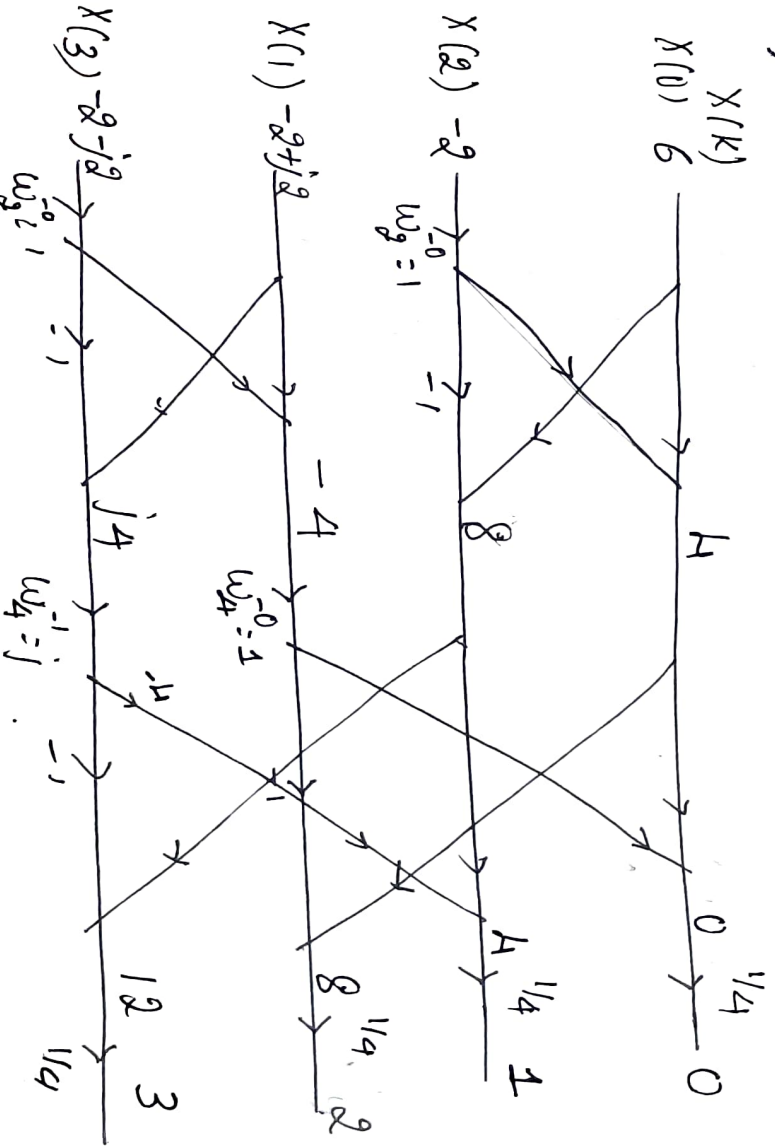
$6 - 2 = 4$

$-2 + j2 - 2 - j2 = -4$

$-2 + j2 + 2 + j2 = j4$

$X(N) = \{0, 1, 2, 3\}$

II-method C.R.R



$X(N) = \{0, 1, 2, 3\}$

III - method

$$x(n) = \frac{1}{N} \sum_{k=0}^{N-1} x(k) \omega_N^{-kn}$$

$$x^*(n) = \frac{1}{N} \sum_{k=0}^{N-1} x^*(k) \omega_N^{kn}$$

$$N \cdot x^*(n) = \sum_{k=0}^{N-1} x^*(k) \omega_N^{kn}$$

$$= y(n)$$

$$\therefore x^*(n) = \frac{1}{N} y(n)$$

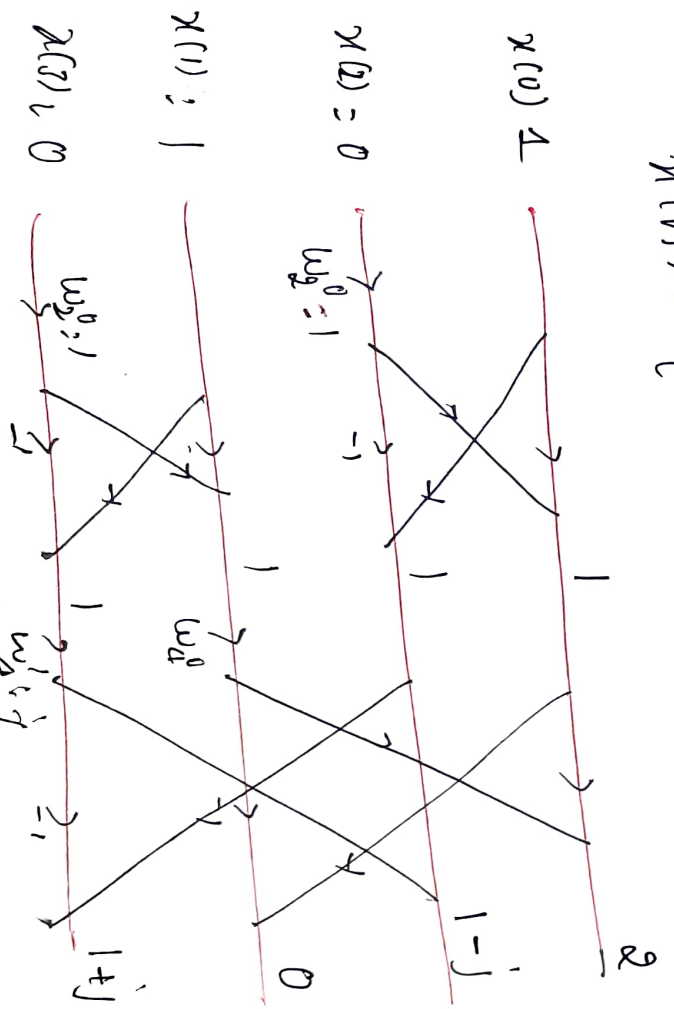
to get IDFTs

$$x(n) = \frac{1}{N} y^*(n)$$

② @ compute DFT of the seqn $x(n)$ given below using DIT-FFT alg. ~~use~~ found in part (a)

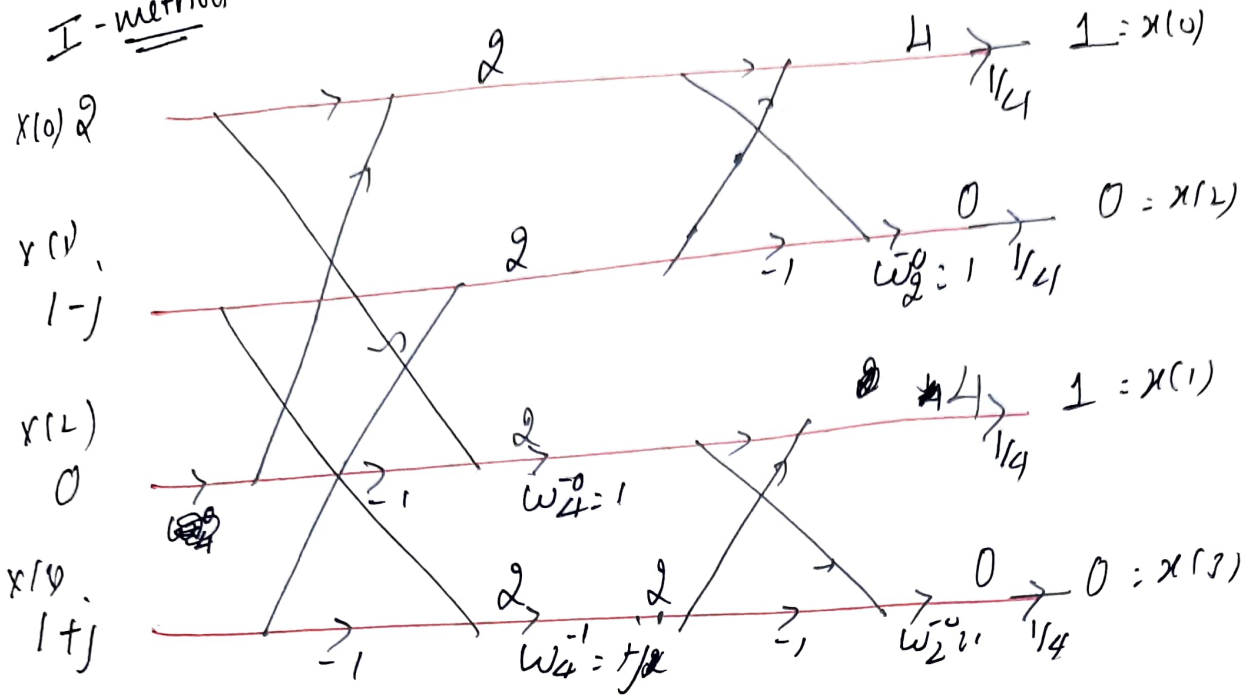
⑤ for the $x(n) = \{1, 1, 0, 0\}$ find $x(n)$ using DIT-Alg

$$x(n) = \{1, 1, 0, 0\}$$



$$X(K) = \{ 2, 1-j, 0, 1+j \}$$

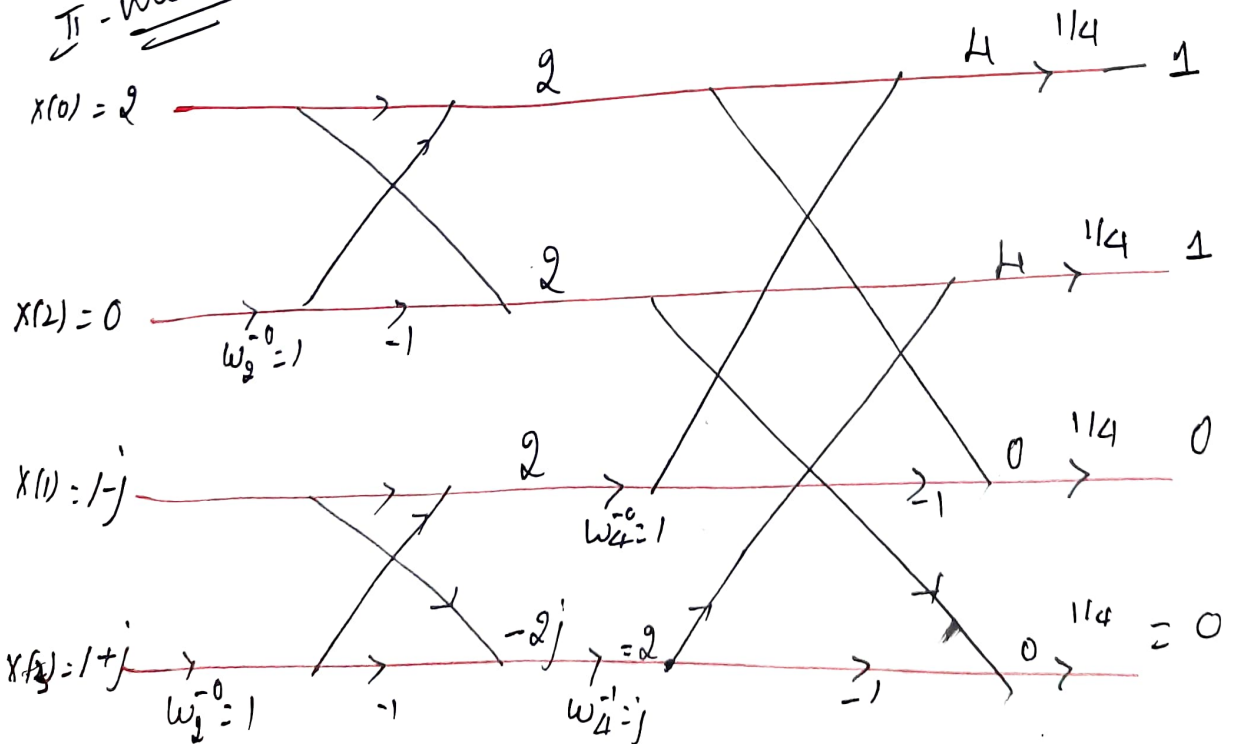
I-method



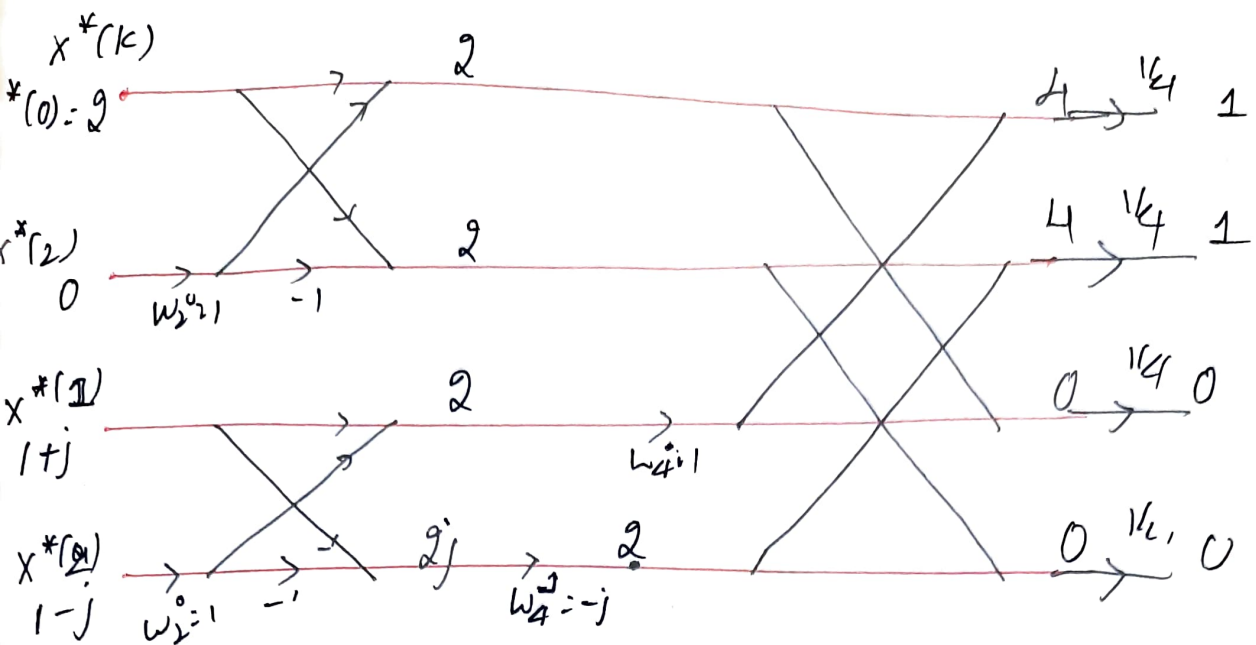
$$\begin{aligned}
 1-j + 1+j &: \\
 1-j - 1-j &= -2j(j) \\
 &= -2j^2 \\
 &= 2
 \end{aligned}$$

$$x(n) = \{ 1, 1, 0, 0 \}$$

II-method



$$\begin{aligned}
 1-j - 1-j & \\
 1-j + 1+j & \\
 x(n) &: \{ 1, 1, 0, 0 \}
 \end{aligned}$$

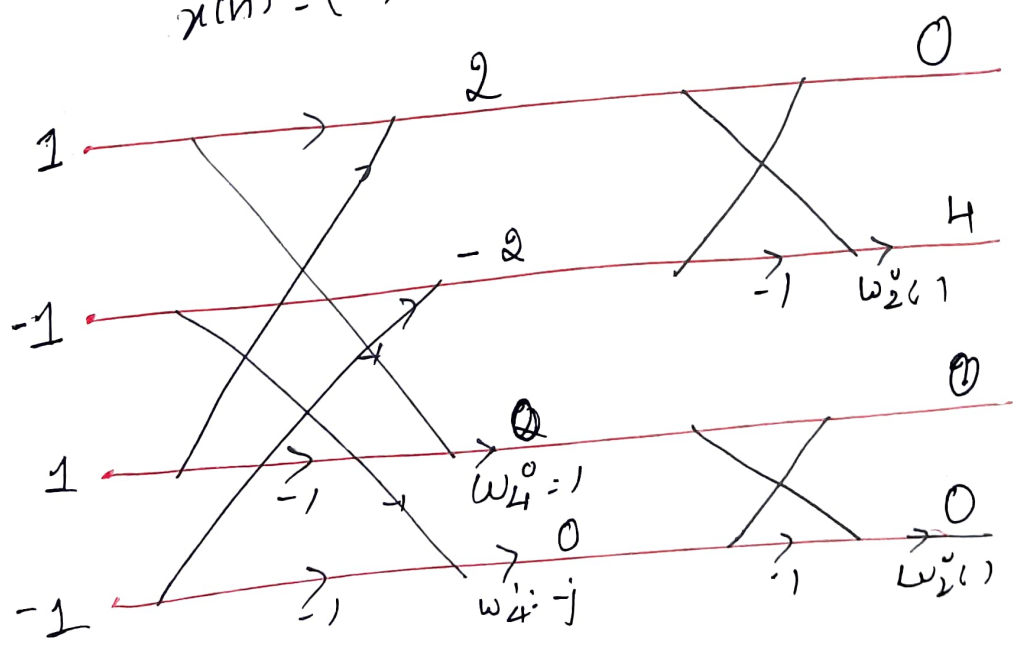


$$\begin{aligned}
 (1+j)(1-j) &= 2j(-j) \\
 &= -2j^2 \\
 &= 2
 \end{aligned}$$

$$\begin{aligned}
 (1+j)(-1+j) &= 2j(-j) \\
 &= 2
 \end{aligned}$$

② Find DFT of $4e^{j\pi n}$ using DIF. Obtain IDFT. $x(n) = (-1)^n$

$x(n) = \{1, -1, 1, -1\}$



$$X(k) = \{0, 0, 4, 0\}$$

① Find circular convolution of the seqs given below using ~~DFT~~ FFT Alg

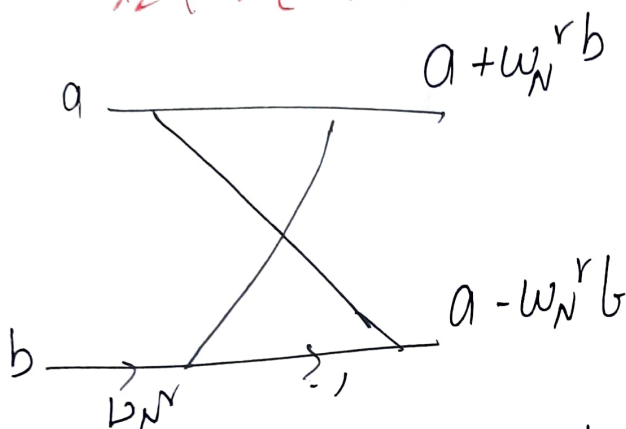
$$x_1(n) = \{1, 0, 1, 0\}$$

$$x_2(n) = \{1, 1, 1, 1\}$$

② find circular convolution of the seqs given below

$$x_1(n) = \{1, 0, 1, 0\}$$

$$x_2(n) = \{1, 1, 1, 1\}$$



find (i) $x_1(k) = \{2, 0, 2, 0\}$

$I \rightarrow \{2, 0, 0, 0\}$

$II \rightarrow \{2, 0, 2, 0\}$

(2) $x_2(k) = \{4, 0, 0, 0\}$

$I \rightarrow \{2, 0, 2, 0\}$

$2I \rightarrow \{4, 0, 0, 0\}$

III $y(n)$

$y(k) = x_1(k) \cdot x_2(k) = \{8, 0, 0, 0\}$
2 DFT using DIT-FFT

$y(n) = \{2, 2, 2, 2\}$

$I \rightarrow \{8, 8, 0, 0\}$ $\{2, 2, 2, 2\}$

to find

N-point DFT of 2-real seqn using
single N-point DFT

Let

$g(n) \rightarrow$ real seqn of length N

$h(n) \rightarrow$ " " " " " "

their DFTs $G(k)$ & $H(k)$

Let $x(n) = g(n) + j h(n) \rightarrow$ (1)

$g(n) = \text{Re}\{x(n)\} = x_R(n) \rightarrow$ (2)

$h(n) = \text{Im}\{x(n)\} = x_I(n) \rightarrow$ (3)

taking DFT on eq (2) & (3)
using symmetry properties of
real seqn

$$G(k) = \frac{1}{2} [X(k) + X^*((-k))_N]$$

$$H(k) = \frac{1}{2j} [X(k) - X^*((-k))_N]$$

$$X^*((-k))_N = X^*(N-k)$$

determine 4-point DFTs of the 2 real
seqns $g(n) = \{1, 2, 0, 1\}$ $h(n) = \{2, 2, 1, 1\}$
using a single 4-point DFT

$$x(n) = g(n) + j h(n)$$

$$x(n) = \{1+j2, 2+j2, j, 1+j\}$$

$$X(k) = \{4+j6, 2, -2, j2\}$$

$$X^*(K) = \{4-j6, 2, -2, -j2\}$$

$$X^*(4-K) = \{4-j6, -j2, -2, 2\}$$

$$G(K) = \{4, 1-j, -2, 1+j\}$$

$$H(K) = \{6, 1-j, 0, 1+j\}$$

② To find $2N$ -point DFT of a real seqⁿ using a single N -point DFT

Let $v(n) \rightarrow$ a real seqⁿ of length $2N$ & its DFT $V(K)$.

Let $g(n) \neq h(n) \rightarrow$ are real seqⁿ of length N

$$g(n) = v(2n) \quad 0 \leq n \leq N-1$$

$$h(n) = v(2n+1)$$

$$x(n) = g(n) + j h(n)$$

$$v(k) = \sum_{n=0}^{2N-1} v(n) \omega_{2N}^{kn}$$

$$v(k) = G((K))_N + \omega_{2N}^{Kn} H((K))_N$$

$$0 \leq k \leq 2N-1$$

find 8-point DFT of the seqⁿ
 $x(n) = \{1, 2, 2, 2, 0, 1, 1, 1\}$ using
 a single 4-point DFTs

Solⁿ $x(n) = \{1, 2, 2, 2, 0, 1, 1, 1\}$

$$g(n) = x(2n) = \{1, 2, 0, 1\}$$

$$h(n) = x(2n+1) = \{2, 2, 1, 1\}$$

Let $x(n) = g(n) + j h(n)$

$$= \{1+j2, 2+j2, j, 1+j\}$$

$$X(k) = \{4+j6, 2, -2, j^2\}$$

$$X^*(k) = \{4-j6, 2, -2, -j^2\}$$

$$X^*((1-k))_4 = \{4-j6, -j^2, -2, 2\}$$

$$G(k) = \{4, 1-j, -2, 1+j\}$$

$$H(k) = \{6, 1-j, 0, 1+j\}$$

$$v(k) = G(k)_N + W_{2N}^{kN} H(k)_N$$

$$V(k) = G(k)_4 + W_8^k H(k)_4$$

$$\begin{Bmatrix} 10, & 1-j^2 \cdot 4/4^2, & -2, & 1-j \cdot 0.4/4^2, \\ -2, & 1+j \cdot 0.4/4^2, & -2, & 1+j^2 \cdot 4/4^2 \end{Bmatrix}$$

Computation of DFT via Filtering methods

① Chirp Z-transform:

* WKT DFT of an N -point discrete time signal $x(n)$ is viewed as Z -transform of $x(n)$ evaluated at N equally spaced points on the unit circle

* Here we are interested in evaluating Z -transform on different contours

in the complex-plane including the unit circle & spiral contour.

* TO derive chirp Z-transform of the seqⁿ $x(n)$

consider the Z -transform of values ~~which~~ z_k^{-1} of Z_k^{-1}

$$X(Z_k) = \sum_{n=0}^{N-1} x(n) Z_k^{-n} \quad \cdot \quad k=0, 1, \dots, N-1$$

①

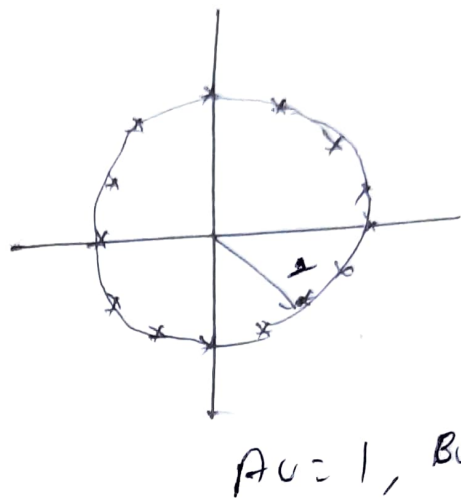
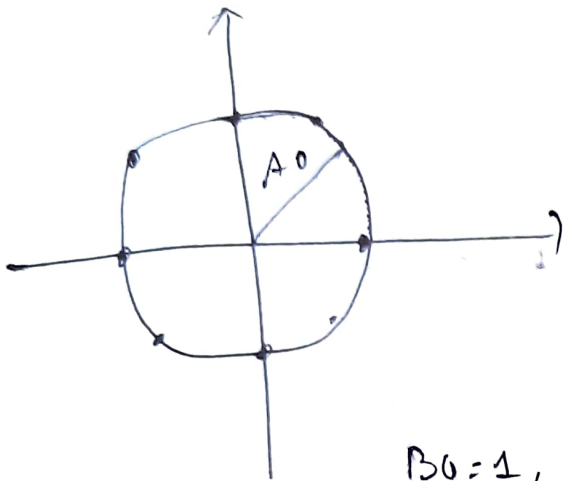
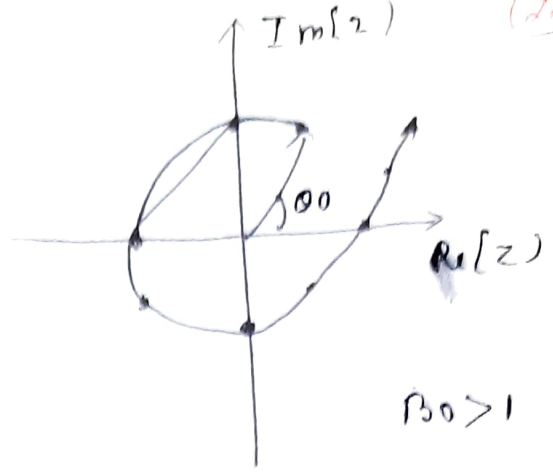
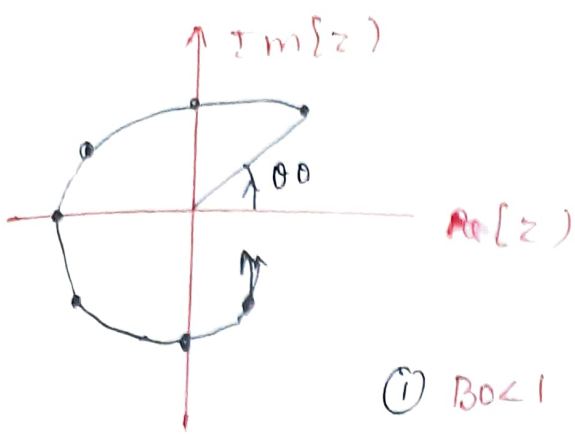
let $Z_k \rightarrow$ points on the spiral centered about origin.

$$Z_k = AB^{-k} \rightarrow \textcircled{2}$$

where $A = Ae^{j\theta_0} \rightarrow$ defines point at which spiral starts

$B = B_0 e^{-j\phi_0} \rightarrow$ defines angular separation

betⁿ points Z_k & determines the rate at which the spiral moves



- * The above figs shows typical cases
- * for $B_0 < 1$, the seq of points z_k spirals towards the origin
- * for $B_0 > 1$, the spiral is away from the origin
- * for $B_0 = 1, A_0 < 1$, the seq of points lies on a circle of radius A_0
- * $A_0 = 1, B_0 = 1, \phi = \frac{2\pi}{N}$ corresponds to DFT & seqn of points lies on the unit circle

substituting eq (2) in (1)

Rachin
Madan

$$X(z_k) = \sum_{n=0}^{N-1} x(n) A^{-n} B^{+kn} \rightarrow (3)$$

we can express eq (3) in convolution form

by consider $nk = \frac{1}{2} [n^2 + k^2 - (n-k)^2]$

$$X(z_k) = \sum_{n=0}^{N-1} x(n) A^{-n} B^{\frac{n^2}{2}} B^{\frac{k^2}{2}} B^{-\frac{(n-k)^2}{2}} B^{-\frac{(n-k)^2}{2}}$$

(4)

(5)

let us define 2 funs
g(n) & h(n) as follows

$$g(n) = x(n) A^{-n} B^{n^2/2}$$

~~$$h(k) = B^{-k^2/2}$$~~

$$h(n) = B^{-n^2/2}$$

$$\therefore g(n) = x(n) A^{-n} \frac{1}{B^{-n^2/2}}$$

$$= x(n) A^{-n} \cdot \frac{1}{h(n)}$$

$$h(k) = B^{-k^2/2}$$

$$h(n-k) = B^{-(n-k)^2/2}$$

$$\therefore X(z_k) =$$

$$X(z_k) = \sum_{n=0}^{N-1} g(n) \cdot \frac{1}{h(k)} \cdot h(n-k)$$

$$= \frac{1}{h(k)} \cdot \sum_{n=0}^{N-1} g(n) \cdot h(n-k)$$

replace n by k & vice versa

$$X(z_n) = \frac{1}{h(n)} \sum_{k=0}^{N-1} g(k) \cdot h(k-n)$$

$$= \frac{1}{h(n)} [g(n) * h(-n)]$$

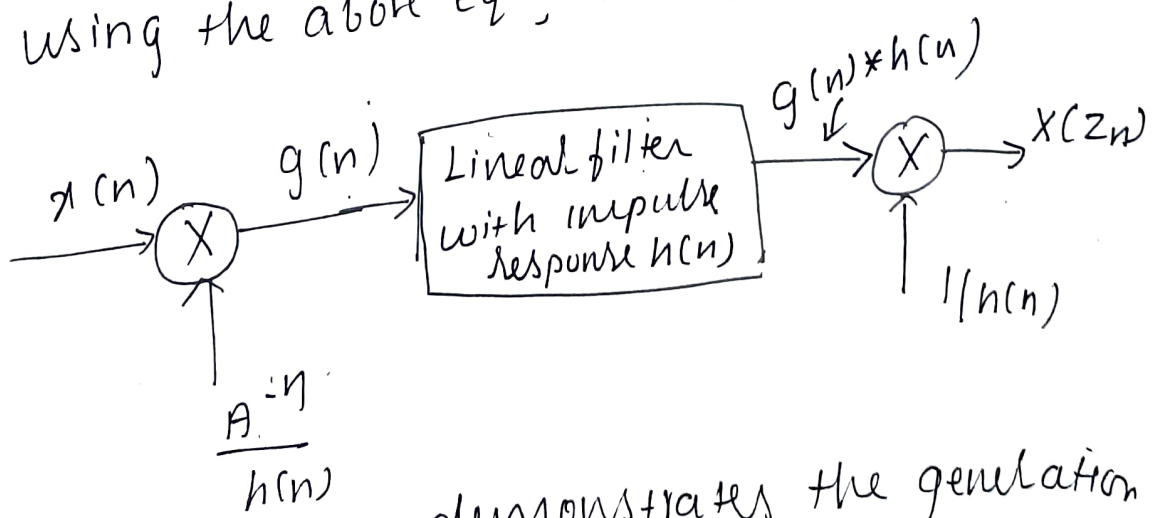
but $h(-n) = h(n)$ as $h(n)$ is even funⁿ

$$x(n) * h(n) = \sum_{k=0}^{N-1} x(k) h(n-k)$$

$$X(z_n) = \frac{1}{h(n)} [g(n) * h(n)]$$

↳ (6)

using the above eqⁿ, the foll block dia



The above B.D demonstrates the generation of chirp-Z-transform which is the DFT for the conditions specified in fig (4)

if $A_0 = 1$ & $B_0 = 1$ we get

$$h(n) = B^{-n^2/2}$$

$$= [B_0 e^{-j\phi_0}]^{-n^2/2}$$

$$= e^{+j\frac{\phi_0 n^2}{2}} = e^{-j\phi_0} e^{j\frac{n\phi_0 n}{2}} \rightarrow \textcircled{7}$$

comparing eq $\textcircled{7}$ with $e^{j\omega n}$ we get

$$\omega = \frac{n\phi_0}{2}$$

Eq $\textcircled{7}$ can be thought of as a complex exponential seqⁿ with a linearly rising freq ω . Such a signal is called chirp signals in radar system hence the name chirp Z-transform

* All the opⁿs illustrated above can be carried out digitally however convolution opⁿ for chirp Z-transform can be implemented by means of

Charge transfer devices (CTD) & such devices are available commercially. The CTD imp^l-mentation appears to be very cheap but not 100% efficient

$$X(k) = \sum_{m=-\infty}^{+\infty} x(m) \omega_N^{-k(N-m)} \rightarrow (2)$$

let us define a seqⁿ

$$y_k(n) = \sum_{m=-\infty}^{\infty} x(m) \omega_N^{-k(n-m)} \rightarrow (3)$$

then above eqⁿ (2) becomes

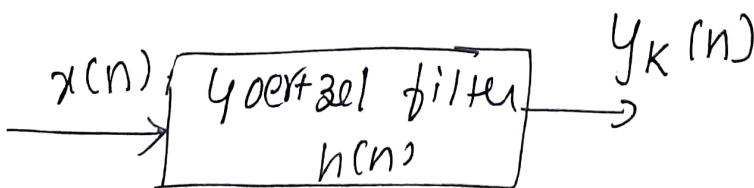
$$X(k) = y_k(n) \Big|_{n=N} \rightarrow (4)$$

let $h(n) = \omega_N^{-nk}$ as a impulse response of a filter known as Goertzel's filter then eq (3) becomes

$$y_k(n) = \sum_{m=-\infty}^{\infty} x(m) \cdot h(n-m)$$

$$= x(n) * h(n) \rightarrow (5)$$

eq (5) is ~~not~~ realized using the B.D shown below



* let .

Let us now proceed to find the transfer fun
 of $y[n]$ over $x[n]$ filter

(25)

$$\frac{Y_K(z)}{X_K(z)} = H_K(z) = \sum \{h(n)\}$$

$$= \sum_{n=0}^{\infty} h(n) z^{-n}$$

$$= \sum_{n=0}^{\infty} (\omega_N^{-nK}) z^{-n}$$

$$= \sum_{n=0}^{\infty} (\omega_N^{-K} z^{-1})^n$$

$$\text{Let } \sum_{n=0}^{\infty} a^n = \frac{1}{1-a} \quad |a| < 1$$

$$H_K(z) = \frac{Y_K(z)}{X(z)} = \frac{1}{1 - \omega_N^{-K} z^{-1}}$$

$$Y_K(z) [1 - \omega_N^{-K} z^{-1}] = X(z)$$

$$Y_K(z) - \omega_N^{-K} z^{-1} Y_K(z) = X(z)$$

+ taking inverse z- transform

$$y_K(n) - \omega_N^{-K} y_K(n-1) = x(n)$$

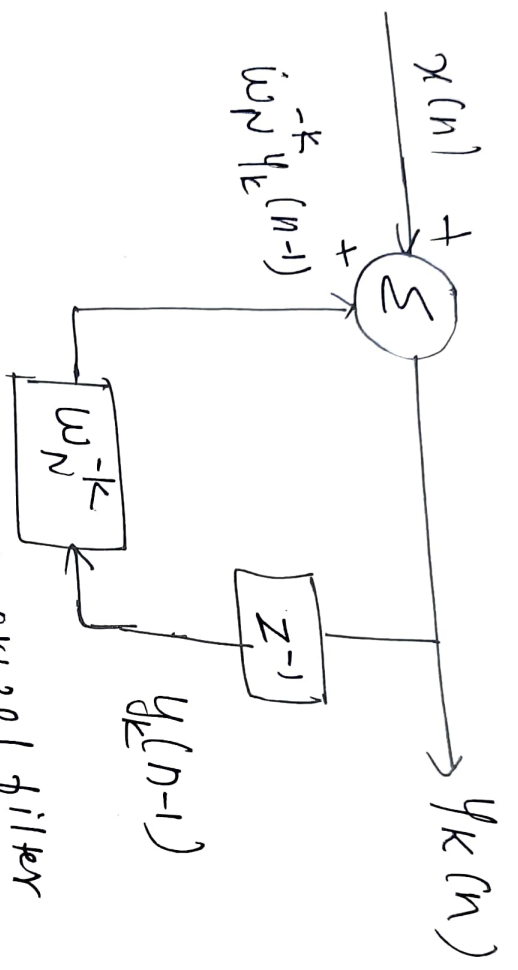
$$\therefore \boxed{y_K(n) = x(n) + \omega_N^{-K} y_K(n-1)} \rightarrow \textcircled{6}$$

take delay = 1 hence it is 1st order filter

Assuming $y_k^{(n-1)} = 0$ eq (5) is solved recursively to find $y_k^{(n)}$ then from eq (3)

$$x_k^{(n)} = y_k^{(n)}$$

using eq (3) we can show the B.P.O of 1st order goertzel filter as below



B.P.O of 1st order goertzel filter

the final O/P of 1st order goertzel filter requires N complex operations to compute the O/P at the N^{th} sample.

* The complex operation in the above eqn can be reduced by combining complex conjugate poles

$$H_k(z) = \frac{1}{1 - w_N^{-k} z^{-1}}$$

$$H_K(z) = \frac{1}{1 - \omega_N^{-k} z^{-1}} \times \frac{(1 - \omega_N^k z^{-1})}{(1 - \omega_N^k z^{-1})}$$

only conjugates

$$= \frac{1 - \omega_N^k z^{-1}}{1 - \omega_N^k z^{-1} - \omega_N^k z^{-1} + z^{-2}}$$

$$= \frac{1 - \omega_N^k z^{-1}}{1 - z^{-1} [\omega_N^{-k} + \omega_N^k] + z^{-2}}$$

WKT

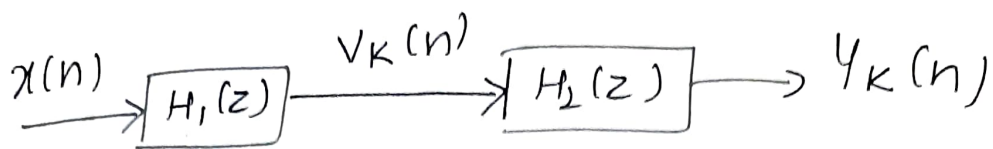
$$[\omega_N^{-k} + \omega_N^k] = e^{+j\frac{2\pi}{N}k} + e^{-j\frac{2\pi}{N}k}$$

$$= 2 \cos \frac{2\pi}{N}k$$

$$H_K(z) = \frac{1 - \omega_N^k z^{-1}}{1 - z^{-1} 2 \cos \frac{2\pi}{N}k + z^{-2}}$$

$$H_K(z) = \underbrace{H_1(z)}_{\text{poles of } H_K(z)} \cdot \underbrace{H_2(z)}_{\text{zeros of } H_K(z)}$$

$$H_K(z) = \frac{1}{1 - 2z^{-1} \cos \frac{2\pi}{N}k + z^{-2}} (1 - \omega_N^k z^{-1})$$



$$H_1(z) = \frac{V_k(z)}{X(z)} = \frac{1}{1 - 2z^{-1} \cos \frac{2\pi}{N} k + z^{-2}}$$

$$V_k(z) [1 - 2z^{-1} \cos \frac{2\pi}{N} k + z^{-2}] = X(z)$$

~~IZT~~

$$v_k(z) - 2z^{-1} v_k(z) \cos \frac{2\pi}{N} k + v_k(z) z^{-2} = X(z)$$

Taking IZT

$$v_k(n) - 2 v_k(n-1) \cos \frac{2\pi}{N} k + v_k(n-2) = x(n)$$

$$v_k(n) = x(n) + 2 v_k(n-1) \cos \frac{2\pi}{N} k - v_k(n-2)$$

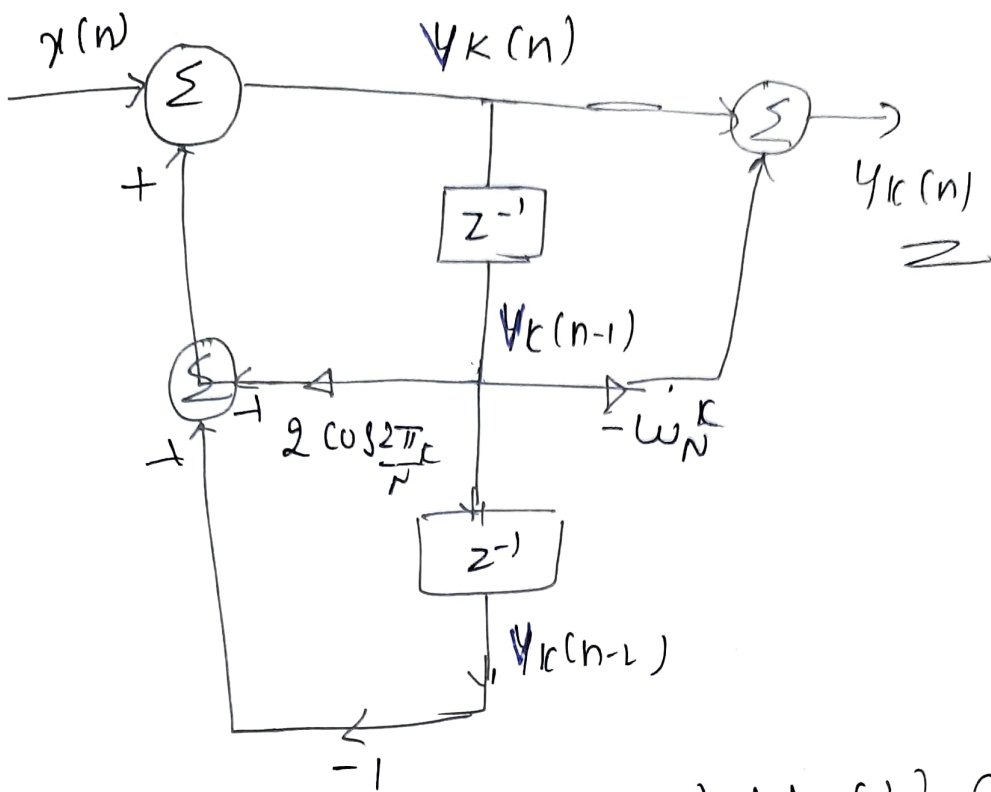
$$H_2(z) = \frac{Y_k(z)}{V_k(z)} = 1 - \omega_N^k z^{-1}$$

$$Y_k(z) = V_k(z) [1 - \omega_N^k z^{-1}]$$

IZT

$$y_k(n) = v_k(n) - \omega_N^k v_k(n-1)$$

B.D of π order filter



Initial conditions $y_k(-1)$ & $y_k(-2)$ are assumed to be 0

The direct form π order Goertzel filter is shown in above fig

Let $x(k) = y_k(n) |_{n=N}$ iterated

The recursive eqn is ~~is~~ iteration for $n = 0$ to $n = N$ & the final o/p is computed only once at a time $n = N$

each iteration requires one real xions & 2 additions finally for real seq $x(n)$ requires $(N+1)$ real xions & one complex xions to get $x(k)$ & also $x(N-k)$

when $x(k)$ is to be computed
 at some value of k , otherwise
 FFT alg are preferred

① compute $x(2)$ & $x(3)$ given
 $x(n) = \{2, 0, 2, 0\}$ use Goertzel Alg

Soln:

According to Goertzel Alg.

$$X(k) = Y_k(N) = Y_k(n)/n=N$$

$$= Y_k(4)$$

↳

$$Y_k(n) = \omega_N^{-k} Y_k(n-1) + x(n)$$

(i) $x(2)$ $k = 2$

$$Y_2(n) = \omega_4^{-2} Y_2(n-1) + x(n)$$

$$\omega_4^2 = -1$$

$$= -Y_2(n-1) + x(n)$$

initial $Y_2(-1) = 0$

$$Y_2(0) = -Y_2(-1) + x(0) = -0 + 2 = 2$$

$$Y_2(1) = -Y_2(0) + x(1) = -2 + 0 = -2$$

$$Y_2(2) = -Y_2(1) + x(2) = 2 + 2 = 4$$

$$Y_2(3) = -Y_2(2) + x(3) = 0 - 4 = -4$$

$$\therefore X(2) = Y_2(4) \quad \bigcirc$$

1111

compute $X(2)$ & $X(3)$ given

$$x(n) = \{ \underset{\uparrow}{2}, 0, 2, 0 \} \text{ use Goertzel}$$

According to Goertzel Alg

$$X(K) = Y_K(n) / n = N \\ = Y_K(4)$$

$X(2)$ {

$$K = 2, N = 4$$

$$X(2) = Y_K(4) = Y_2(4)$$

WKT

$$Y_K(n) = x(n) + w_N^{-K} Y_K(n-1)$$

$$Y_2(n) = x(n) + w_4^{-2} Y_2(n-1)$$

$$w_4^{-2} = -1$$

$$Y_2(n) = x(n) - Y_2(n-1) \longrightarrow \textcircled{1}$$

initial $Y_2(-1) = 0$

$$Y_2(0) = x(0) - Y_2(-1) = 2 - 0 = 2$$

$$Y_2(1) = x(1) - Y_2(0) = 0 - 2 = -2$$

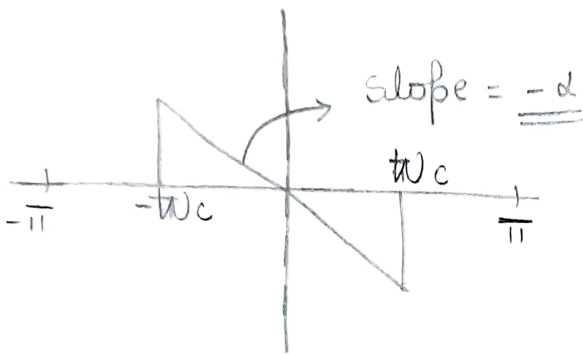
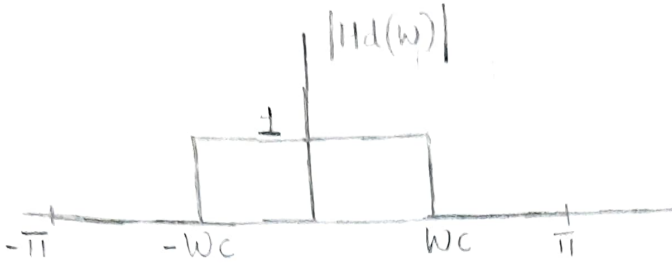
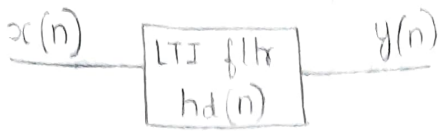
$$Y_2(2) = x(2) - Y_2(1) = 2 - (-2) = 4$$

$$Y_2(3) = x(3) - Y_2(2) = 0 - 4 = -4$$

$$Y_2(4) = x(4) - Y_2(3) = 0 - (-4) = 4$$

$$\boxed{X(2) = Y_2(4) = 4}$$

FINITE IMPULSE RESPONSE FILTER (FIR)



IIR filters were designed to give a desired magnitude response without regards to phase response. In many applications, a linear phase is required throughout pass band. In order to preserve shape of stop band signal assume desired LPF, frequency response

defined by

$$H_d(w) = \begin{cases} e^{-j\alpha w}, & 0 < |w| < w_c \\ 0, & w_c < |w| < \pi \end{cases}$$

$$|H_d(w)| = \begin{cases} 1, & 0 < |w| < w_c \\ 0, & w_c < |w| < \pi \end{cases}$$

$$\angle H_d(w) = \begin{cases} -w\alpha, & 0 < |w| < w_c \\ 0, & w_c < |w| < \pi \end{cases}$$

Using eqn (2) and (3), mag and phase spectra shown in fig (2) and (3) are drawn. DATE / / 200

O/P of filter shown in fig (1) in frequency domain is

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

Sub eqn (1) in (4), we get

$$Y(\omega) = \begin{cases} X(\omega) e^{-j\omega d} & , 0 < |\omega| < \omega_c \\ 0 & , \omega_c < |\omega| < \pi \end{cases}$$

Taking IDTF on b.s we get

$$y(n) = \underline{\underline{x(n-d)}}$$

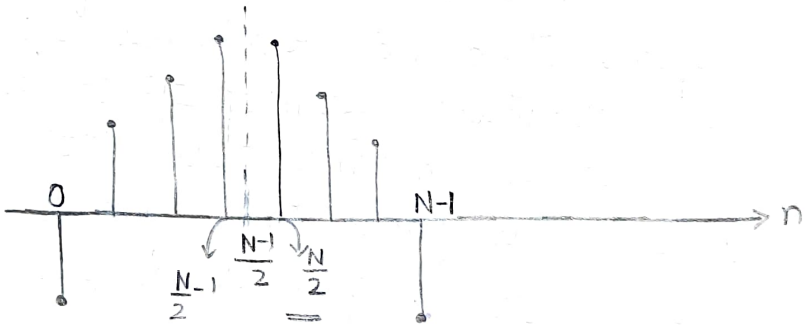
Above eqn means linear phase filter did not alter the shape of I/P sgl. simply translated it to right by an amount d . If the phase response had not been linear, O/P sgl would be a distorted version of I/P sgl $x(n)$. It can be shown that causal IIR filter cannot give linear phase. Only special type of FIR filter can give linear phase.

THEOREM FOR LINEAR PHASE:

Statement: Let $h(n)$ represent impulse response of a discrete time system. Then necessary and sufficient conditions for existence of linear phase are as follows:

- ① $h(n)$ must be of finite duration
- ② $h(n)$ must be either symmetric or antisymmetric about its midpt
- ③ For ~~partic~~ practical realizatn, $h(n)$ must be causal sequence

Case 1: Let $h(n)$ be symmetric about its midpt. Let N be even.



Even symmetry of $h(n)$ about its midpt is explained mathematically by the following discrete eqn

$$h(n) = h(N-1-n) \rightarrow \text{①}$$

WKT $H(z) = \sum_{n=0}^{N-1} h(n) z^{-n}$

$$H(z) = \sum_{n=0}^{N/2-1} h(n) z^{-n} + \sum_{n=N/2}^{N-1} h(n) z^{-n}$$

let us substitute $m = N-1-n$ to the second summation on RHS of above eqn.

then the above eqn becomes

$$H(z) = \sum_{n=0}^{N/2-1} h(n) z^{-n} + \sum_{n=N/2-1}^0 h(n) z^{-n}$$

$$H(z) = \sum_{n=0}^{N/2-1} h(n) \bar{z}^n + \sum_{m=N/2}^N h(N-1-m) \bar{z}^{(N-1-m)}$$

Since m is a dummy variable, it can be replaced by n . Acc. above eqn becomes

$$H(z) = \sum_{n=0}^{N/2-1} h(n) \bar{z}^n + \sum_{n=0}^{N/2-1} h(N-1-n) \bar{z}^{(N-1-n)} \rightarrow (2)$$

Sub eqn (1) in eqn (2), we get

$$H(z) = \sum_{n=0}^{N/2-1} h(n) \left[\bar{z}^n + \bar{z}^{(N-1-n)} \right]$$

Since $\sum_{n=-\infty}^{\infty} |h(n)| < \infty$, FIR system under consideration is stable. Hence, its frequency response is obtained by letting $z = e^{j\omega}$ in $H(z)$

$$H(e^{j\omega}) = H(\omega) = \sum_{n=0}^{N/2-1} h(n) \left[e^{-j\omega n} + e^{-j\omega(N-1-n)} \right]$$

$$H(\omega) = 2e^{-j\omega \left(\frac{N-1}{2}\right)} \sum_{n=0}^{N/2-1} h(n) \left[e^{-j\omega \left(n - \left(\frac{N-1}{2}\right)\right)} + e^{j\omega \left(n - \left(\frac{N-1}{2}\right)\right)} \right]$$

$$\Rightarrow H(\omega) = e^{-j\omega \left(\frac{N-1}{2}\right)} \sum_{n=0}^{N/2-1} 2h(n) \cos \left[\omega \left[n - \left(\frac{N-1}{2}\right) \right] \right]$$

$$\Rightarrow H(\omega) = e^{-j\omega \left(\frac{N-1}{2}\right)} \sum_{n=0}^{N/2-1} 2h(n) \cos \left[\omega \left[\left(\frac{N-1}{2}\right) - n \right] \right] \rightarrow (3)$$

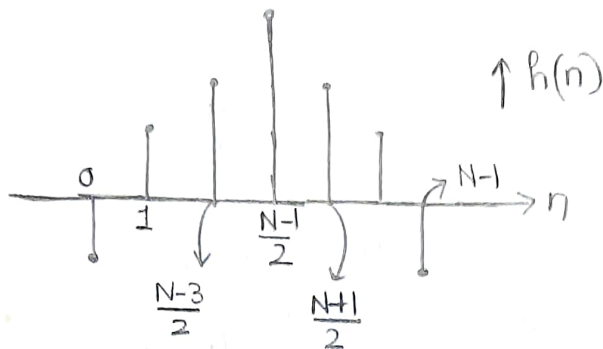
Comparing eqn (3) with $H(\omega) = H_m(\omega) e^{j\theta(\omega)}$ we get

$$H_m(\omega) = \sum_{n=0}^{N/2-1} 2h(n) \cos \left[\omega \left[\left(\frac{N-1}{2}\right) - n \right] \right]$$

$$\theta(\omega) = \begin{cases} -\omega \left(\frac{N-1}{2}\right) + 0 & ; \text{ if } H_m(\omega) > 0 \\ -\omega \left(\frac{N-1}{2}\right) + \pi & ; \text{ if } H_m(\omega) < 0 \end{cases}$$

Hence phase is linear

Case 2: Let $h(n)$ be even symmetric about its mid pt and N be odd.



Even symmetry of $h(n)$ can be explained mathematically by following discrete relation

$$h(n) = h(N-1-n)$$

WKT

$$H(z) = \sum_{n=0}^{N-1} h(n) z^{-n}$$

$$z^{2N-2-N-1} z^{\frac{N-1}{2}}$$

$$H(z) = \sum_{n=0}^{\frac{N-3}{2}} h(n) z^{-n} + h\left(\frac{N-1}{2}\right) z^{-\left(\frac{N-1}{2}\right)} + \sum_{n=\frac{N+1}{2}}^{N-1} h(n) z^{-n}$$

Letting $m = N-1-n$ in the 3rd term on RHS of above eqn, we get

$$H(z) = \sum_{n=0}^{\frac{N-3}{2}} h(n) z^{-n} + h\left(\frac{N-1}{2}\right) z^{-\left(\frac{N-1}{2}\right)} + \sum_{m=\frac{N+1}{2}}^0 h(N-1-m) z^{-(N-1-m)}$$

Since m is a dummy variable, it can be replaced by 'n'. Acc, above eqn becomes

$$H(z) = h\left(\frac{N-1}{2}\right) z^{-\left(\frac{N-1}{2}\right)} + \sum_{n=0}^{\frac{N-3}{2}} h(n) z^{-n} + \sum_{m=0}^{\frac{N-3}{2}} h(N-1-n) z^{-(N-1-n)}$$

Sub eqn (1) in eqn (2) and then combining two summations, we get

$$H(z) = h\left(\frac{N-1}{2}\right) z^{-\left(\frac{N-1}{2}\right)} + \sum_{n=0}^{\frac{N-3}{2}} h(n) \left[z^{-n} + z^{-(N-1-n)} \right]$$

Since impulse response is absolutely summable FIR system under consideration is stable, Hence its impulse response is found by letting $z = e^{j\omega}$ in $H(z)$

$$H(e^{j\omega}) = H(\omega) = h\left(\frac{N-1}{2}\right) + \sum_{n=0}^{(N-3)/2} h(n) \left\{ e^{-j\omega n} + e^{-j\omega(N-1-n)} \right\}$$

$$H(\omega) = e^{-j\omega \left(\frac{N-1}{2}\right)} \left[h\left(\frac{N-1}{2}\right) + \sum_{n=0}^{\frac{N-3}{2}} 2h(n) \frac{e^{-j\omega \left(\frac{N-1}{2} - n\right)} + e^{+j\omega \left(n - \left(\frac{N-1}{2}\right)\right)}}{2} \right]$$

$$H(\omega) = e^{-j\omega \left(\frac{N-1}{2}\right)} \left[h\left(\frac{N-1}{2}\right) + \sum_{n=0}^{\frac{N-3}{2}} 2h(n) \cos\omega \left(n - \left(\frac{N-1}{2}\right)\right) \right]$$

Since $\cos(-\theta) = \cos\theta$ above eqn can be

written as

$$H(\omega) = e^{-j\omega \left(\frac{N-1}{2}\right)} \left[h\left(\frac{N-1}{2}\right) + \sum_{n=0}^{\frac{N-3}{2}} 2h(n) \cos\omega \left(\frac{N-1}{2} - n\right) \right] \rightarrow (3)$$

Comparing eqn (3) with

$$H(\omega) = H_m(\omega) e^{j\theta(\omega)}, \text{ we get}$$

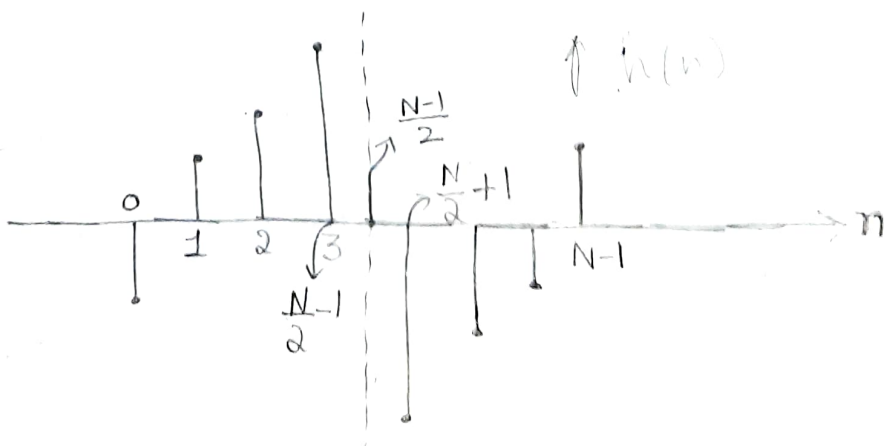
$$H_m(\omega) = h\left(\frac{N-1}{2}\right) + \sum_{n=0}^{\frac{N-3}{2}} 2h(n) \cos\omega \left(\frac{N-1}{2} - n\right)$$

and

$$\theta(\omega) = \begin{cases} -\omega \left(\frac{N-1}{2}\right) + 0, & \text{if } H_m(\omega) > 0 \\ -\omega \left(\frac{N-1}{2}\right) + \pi, & \text{if } H_m(\omega) < 0 \end{cases}$$

hence the phase is linear

Case 3: let $h(n)$ be the antisymmetry about its midpt and N be even



Odd symmetry of $h(n)$ about its midpt is explained mathematically by the discrete relation given below $h(n) = -h(N-1-n) \rightarrow \textcircled{1}$

WKT, $H(z) = \sum_{n=0}^{N-1} h(n) z^{-n}$

$$H(z) = \sum_{n=0}^{\frac{N-1}{2}} h(n) z^{-n} + \sum_{n=\frac{N}{2}}^{N-1} h(n) z^{-n}$$

Letting $m = N-1-n$ in second summation on RHS of above eqn gives

$$H(z) = \sum_{n=0}^{\frac{N}{2}-1} h(n) z^{-n} + \sum_{m=\frac{N}{2}-1}^0 h(N-1-m) z^{-(N-1-m)}$$

Since m is some dummy variable, it can be replaced by n . Acc above eqn becomes

$$H(z) = \sum_{n=0}^{\frac{N}{2}-1} h(n) z^{-n} + \sum_{n=0}^{\frac{N}{2}-1} h(N-1-n) z^{-(N-1-n)} \quad \textcircled{2}$$

Sub eqn $\textcircled{1}$ in eqn $\textcircled{2}$ and then combining two summations, we get

$$H(z) = \sum_{n=0}^{\frac{N}{2}-1} h(n) \left\{ z^{-n} - z^{-(N-1-n)} \right\}$$

Since the system under consideration is stable ($\sum_{n=-\infty}^{\infty} |h(n)| < \infty$), frequency response is obtained by letting $z = e^{j\omega}$ in $H(z)$

$$H(e^{j\omega}) = H(\omega) = \sum_{n=0}^{\frac{N}{2}-1} h(n) \left\{ e^{-j\omega n} - e^{-j\omega(N-1-n)} \right\}$$

$$H(e^{j\omega}) = H(\omega) = e^{-j\omega \left(\frac{N-1}{2}\right)} \sum_{n=0}^{\frac{N}{2}-1} (-2j) h(n) \left\{ \frac{e^{-j\omega \left(n - \left(\frac{N-1}{2}\right)\right)} + e^{j\omega \left(n - \left(\frac{N-1}{2}\right)\right)}}{(-2j)} \right\}$$

$$H(\omega) = e^{-j\omega \left(\frac{N-1}{2}\right)} \sum_{n=0}^{\frac{N}{2}-1} (-2e^{j\frac{n}{2}}) h(n) \sin \omega \left(n - \left(\frac{N-1}{2}\right)\right) \quad 131$$

Since $\sin(-\theta) = -\sin\theta$ we get

$$H(\omega) = e^{-j\omega\left(\frac{N-1}{2}\right)} \sum_{n=0}^{N/2-1} a e^{j\frac{\pi}{2}} h(n) \sin\left(\frac{N-1}{2} - n\right)$$

$$H(\omega) = e^{-j\omega\left(\frac{N-1}{2}\right) + j\frac{\pi}{2}} \sum_{n=0}^{N/2-1} a h(n) \sin\left(\frac{N-1}{2} - n\right) \rightarrow (3)$$

Comparing eqn (3) with

$$H(\omega) = H_r(\omega) e^{j\theta(\omega)}, \text{ we get}$$

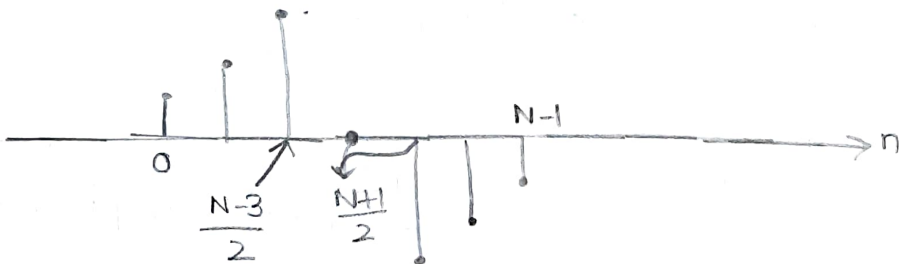
$$H_r(\omega) = e^{-j\omega\left(\frac{N-1}{2}\right) + j\frac{\pi}{2}} \sum_{n=0}^{N/2-1} a h(n) \sin\left(\frac{N-1}{2} - n\right)$$

and

$$\theta(\omega) = \begin{cases} -\omega\left(\frac{N-1}{2}\right) + \frac{\pi}{2} + 0, & \text{if } H_r(\omega) > 0 \\ -\omega\left(\frac{N-1}{2}\right) + \frac{\pi}{2} + \pi, & \text{if } H_r(\omega) < 0 \end{cases}$$

hence the phase is linear

Case 4: let $h(n)$ be odd symmetric about its mid pt and N be odd.



Odd symmetry of $h(n)$ is explained mathematically by the foll. discrete reln

$$h\left(\frac{N-1}{2}\right) = 0 \rightarrow (1) \text{ and } h(n) = -h(N-1-n) \rightarrow (2)$$

WKT

$$H(z) = \sum_{n=0}^{N-1} h(n) z^{-n}$$

$$H(z) = \sum_{n=0}^{N/2-1} h(n) z^{-n} + h\left(\frac{N-1}{2}\right) z^{-\left(\frac{N-1}{2}\right)} + \sum_{n=N/2}^{N-1} h(n) z^{-n}$$

Let $m = N-1-n$ in second summation

$$= \sum_{h=0}^{N-3} h(n) \bar{\alpha}^n + \sum_{m=0}^{N-1} h(N-1-m) \bar{\alpha}^{-(N-1-m)}$$

Since m is dummy variable, replace m by n

$$= \sum_{n=0}^{N-3} h(n) \bar{\alpha}^n + \sum_{m=0}^{N-1} h(N-1-m) \bar{\alpha}^{-(N-1-m)}$$

Sub eqn 1 in eqn 2, and combining the eqn, we get

$$= \sum_{n=0}^{N-3} h(n) \left[\bar{\alpha}^n - \bar{\alpha}^{-(N-1-n)} \right]$$

Since impulse response is absolute summable FIR system is stable, $\left[\sum_{n=-\infty}^{\infty} h(n) < \infty \right]$ Thus it's

frequency response is given by $H(z)$ replace z by $e^{j\omega}$

$$H(e^{j\omega}) = H(\omega) = \sum_{n=0}^{N-3} h(n) \left[e^{-j\omega n} - e^{-j\omega(N-1-n)} \right]$$

$$H(\omega) = e^{-j\omega \left(\frac{N-1}{2} \right)} \sum_{n=0}^{N-3} h(n) \left[e^{j\omega \left(n - \frac{N-1}{2} \right)} - e^{-j\omega \left(n - \frac{N-1}{2} \right)} \right]$$

$$H(\omega) = e^{-j\omega \left(\frac{N-1}{2} \right)} \sum_{n=0}^{N-3} -2j \sin \left(n - \frac{N-1}{2} \right) h(n)$$

$$H(\omega) = e^{-j\omega \left(\frac{N-1}{2} \right)} \sum_{n=0}^{N-3} 2 \sin \left(n - \frac{N-1}{2} \right) h(n)$$

$$H_r(\omega) = \sum_{n=0}^{N-3} 2 \sin \left(n - \frac{N-1}{2} \right) h(n)$$

$$\theta(\omega) = \begin{cases} -\omega \left(\frac{N-1}{2} \right) + \frac{\pi}{2}, & \text{if } H_r(\omega) > 0 \\ -\omega \left(\frac{N-1}{2} \right) + \frac{\pi}{2} + \pi, & \text{if } H_r(\omega) < 0 \end{cases}$$

phase is linear

★ Determine unit impulse response $h(n)$ of a linear phase FIR filter with length $N=4$.
 Given $H_r(0)=1$ and $H_r(\pi/2)=1/4$

5: let us assume $h(n)$ to be even symmetric about its mid pt. Then we have

$$H_n(j\omega) = \sum_{n=0}^{N/2-1} 2h(n) \cos \omega \left(\frac{N-1}{2} - n \right)$$

$$N=4$$

$$H_n(j\omega) = \sum_{n=0}^1 2h(n) \cos \omega (1.5 - n)$$

$$H_n(\omega) = 2h(0) \cos \omega (1.5) + 2h(1) \cos (0.5)\omega$$

$$\textcircled{1} H_n(0) = 1 \Leftrightarrow 1 = 2h(0) + 2h(1)$$

$$\textcircled{2} H_r\left(\frac{\pi}{2}\right) = \frac{1}{4} \Leftrightarrow \frac{1}{4} = 2h(0) \cos\left(1.5 \times \frac{\pi}{2}\right) + 2h(1) \cos\left(0.5 \frac{\pi}{2}\right)$$

$$h(0) = 0.1616$$

$$h(1) = 0.3384$$

Since $h(n)$ is even symmetric about its mid pt, full condtn must be true

$$h(N-1-n) = h(n)$$

$$h(2) = 0.3384$$

$$h(3) = 0.1616$$

$$h(n) = \{0.1616, 0.3384, 0.3384, 0.1616\}$$

★ The frequency response of an FIR filter is given by $H(\omega) = e^{-3j\omega} \{1.2 + 0.6 \cos^2 \omega + 0.8 \cos 2\omega + 0.4 \cos \omega\}$ determine Co-eff of FIR filter?

S: here $\frac{N-1}{2} = 3 \Rightarrow \underline{N=7}$

Since $H(\omega)$ has cosine terms it implies $h(n)$ is even symmetric about its mid pt.
Also N is odd.

Hence, $H(\omega) = e^{-j\omega \left(\frac{N-1}{2}\right)} \left[h\left(\frac{N-1}{2}\right) + \sum_{n=0}^{(N-3)/2} 2h(n) \cos\left[\omega\left(\frac{N-1}{2} - n\right)\right] \right]$

$$H(\omega) = e^{-j3\omega} \left[h(3) + \sum_{n=0}^2 2h(n) \cos\omega(3-n) \right]$$

$$\Rightarrow = e^{-3j\omega} \left[h(3) + 2h(0) \cos 3\omega + 2h(1) \cos 2\omega + 2h(2) \cos \omega \right]$$

↳ ②

⇒ Comparing eqn ① and eqn ②, we get

$$h(3) = 1.2$$

$$h(0) = 0.3$$

$$h(1) = 0.4$$

$$h(2) = 0.2$$

Since $h(n)$ is even symmetry a

$$h(n) = h(N-1-n)$$

$$\Rightarrow h(n) = h(6-n)$$

thus $h(4) = 0.2 = h(2)$

$$h(5) = 0.4 = h(1)$$

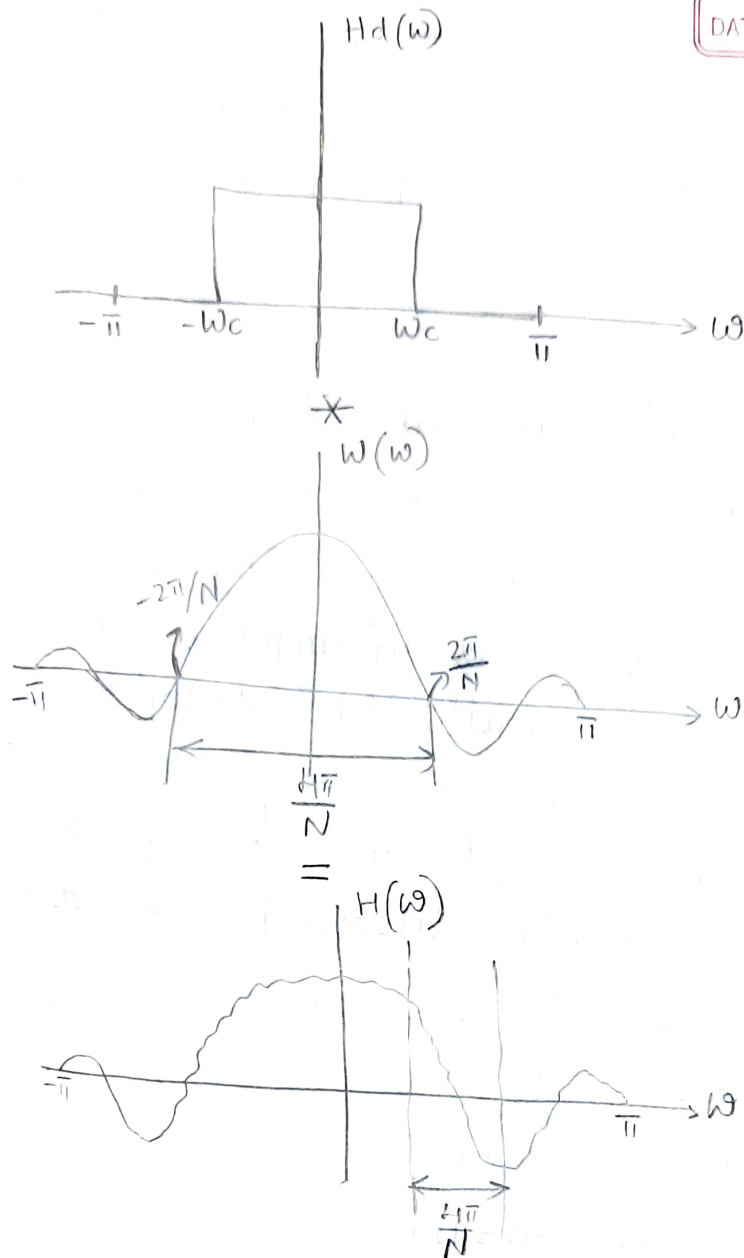
$$h(6) = 0.3 = h(0)$$

thus

$$h(n) = \left\{ \underset{\uparrow}{0.3}, 0.4, 0.2, 1.2, 0.2, 0.4, 0.3 \right\}$$

FIR FILTER DESIGN USING WINDOWS:

DATE / / 200



The Easiest way to design an FIR filter is to truncate impulse response of an IIR filter. Let $h_d(n)$ represent impulse response of a desired low-pass filter [IIR] and $h(n)$ the impulse response of an FIR filter then

$$h(n) = h_d(n)w(n) \rightarrow \textcircled{1}$$

$$w(n) = \begin{cases} 1, & N_1 \leq n \leq N_2 \\ 0, & \text{otherwise} \end{cases}$$

$w(n)$ defined above is called rectangular window. Taking DTFT on b.s of eqn ① we get

$$H(\omega) = H_d(\omega) * W(\omega) \rightarrow \textcircled{2}$$

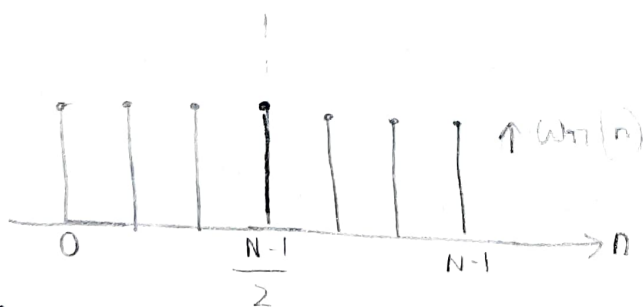
Let us now assume $H_d(\omega)$ represent frequency response of ideal low pass filter $w(\omega)$ frequency response of a causal rectangular window that starts at $n=0$ and ends at $n=N-1$. Then the frequency response of FIR filter is as shown in fig (3).

The frequency response of FIR filter shown in fig (3) is the smeared version of $H_d(\omega)$ shown in fig (1). In general, wider the main lobe $w(\omega)$ more will be the smearing and vice-versa.

In precise, if $w(\omega)$ is an impulse function, then $H(\omega)$ will look exactly like $H_d(\omega)$. Hence, in practice, we make N large enough so that smearing is minimized and yet small enough so that practical implementation becomes possible. Some of the most commonly used windows are described below:

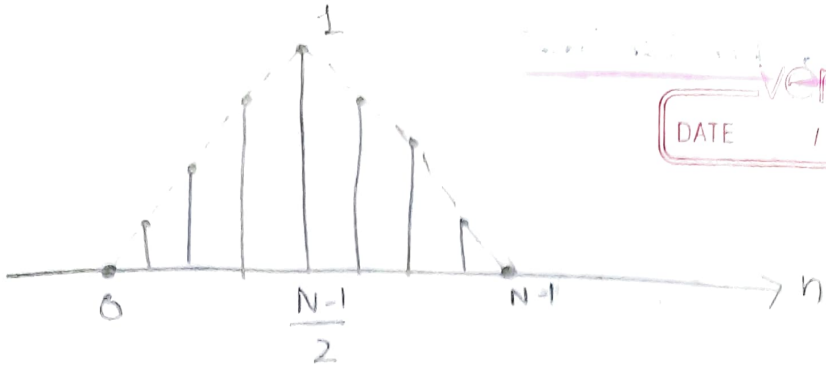
(a) Rectangular window

$$w_R(n) = \begin{cases} 1, & 0 \leq n \leq N-1 \\ 0, & \text{otherwise} \end{cases}$$



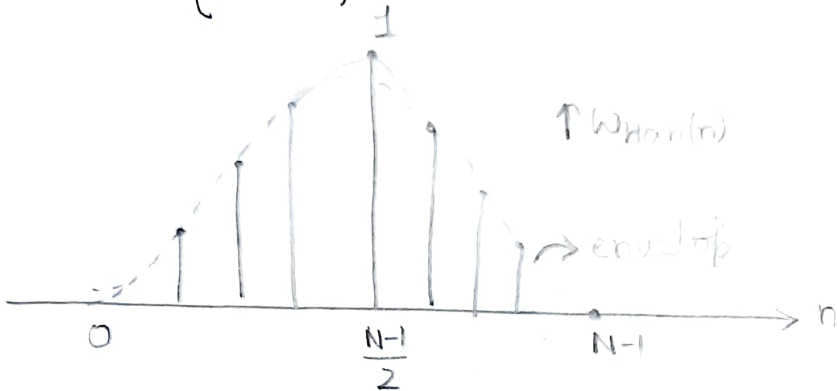
(b) Bartlett window

$$w_R(n) = \begin{cases} 1 - \frac{2 \left(n - \left(\frac{N-1}{2} \right) \right)^2}{N-1} & ; 0 \leq n \leq N-1 \\ 0 & ; \text{o.w} \end{cases}$$



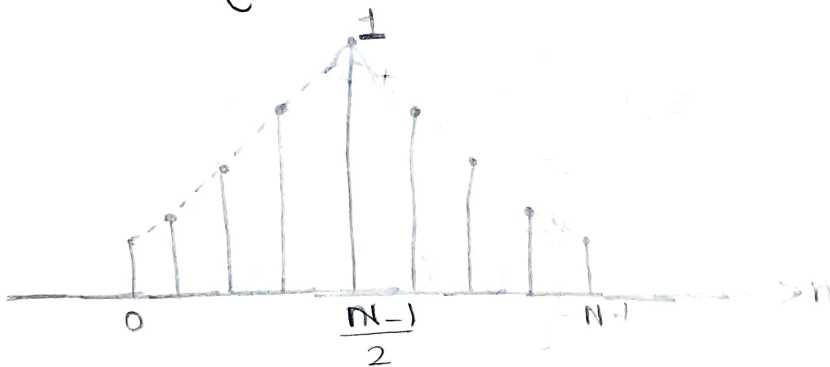
3) Hanning window

$$W_{\text{Han}}(n) = \begin{cases} \frac{1}{2} \left[1 - \cos\left(\frac{2\pi n}{N-1}\right) \right], & 0 \leq n \leq N-1 \\ 0, & \text{otherwise} \end{cases}$$



4) Hamming window

$$W_{\text{Ham}}(n) = \begin{cases} 0.54 - 0.46 \cos\left(\frac{2\pi n}{N-1}\right), & 0 \leq n \leq N-1 \\ 0, & \text{otherwise} \end{cases}$$



5) Blackman window

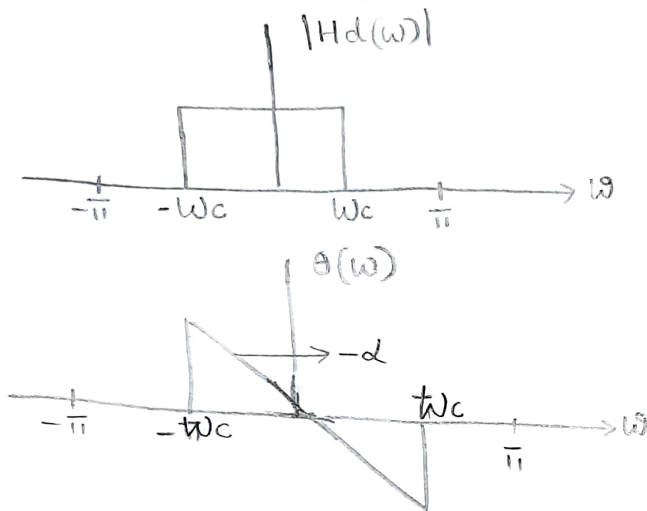
$$W_{\text{Bl}}(n) = \begin{cases} 0.42 - 0.5 \cos\left(\frac{2\pi n}{N-1}\right) + 0.08 \cos\left(\frac{4\pi n}{N-1}\right), & 0 \leq n \leq N-1 \\ 0, & \text{otherwise} \end{cases}$$



DESIGN PROCEDURE:

Let $H_d(\omega)$ represent frequency response of an ideal low pass filter (IIR). Such a filter has the following mathematical description.

$$H_d(\omega) = \begin{cases} e^{-j\omega\alpha} & ; 0 < |\omega| < \omega_c \\ 0 & ; \omega_c < |\omega| < \pi \end{cases}$$



Impulse response, $h_d(n)$ of the ideal low pass filter described above is found as IDFT of $H_d(\omega)$.

$$h_d(n) \triangleq \frac{1}{2\pi} \int_{-\pi}^{\pi} H_d(\omega) e^{+j\omega n} d\omega$$

$$h_d(n) \triangleq \frac{1}{2\pi} \int_{-\omega_c}^{\omega_c} e^{-j\omega\alpha} e^{j\omega n} \cdot d\omega \rightarrow (1)$$

$$h_d(n) = \frac{1}{2\pi} \int_{-\omega_c}^{\omega_c} e^{+j\omega(n-\alpha)} d\omega$$

$$h_d(n) = \frac{1}{2\pi} \left[\frac{e^{j\omega(n-\alpha)}}{j(n-\alpha)} \right]_{-\omega_c}^{\omega_c}$$

$$h_d(n) = \frac{1}{\pi(n-\alpha)} \left[\frac{e^{j\omega_c(n-\alpha)} - e^{-j\omega_c(n-\alpha)}}{2j} \right]$$

$$h_d(n) = \frac{1}{\pi(n-\alpha)} \sin \omega_c(n-\alpha)$$

$$h_d(n) = \frac{\sin \omega_c(n-\alpha)}{\pi(n-\alpha)} ; n \neq \alpha$$

Letting $n=d$ in eqn (1), we get

$$hd(\alpha) = \frac{1}{2\pi} \int_{-\omega_c}^{\omega_c} 1 \cdot d\omega$$

$$e^{\alpha} = 1$$

$$hd(\alpha) = \frac{2\omega_c}{2\pi} = \frac{\omega_c}{\pi}$$

Summarising the results, we get

$$hd(n) = \begin{cases} \frac{\sin \omega_c (n-d)}{\pi (n-d)} & ; n \neq d \\ \frac{\omega_c}{\pi} & ; n = d \end{cases}$$

Impulse response of low pass FIR filter is obtained by multiplying $hd(n)$ by a causal window function $w(n)$ that starts at $n=0$ and ends at $n=N-1$
i.e. $h(n) = hd(n)w(n) ; 0 \leq n \leq N-1$

Since $hd(n)$ is symmetric about $n=d$ and window function $w(n)$ is symmetric about $n=\frac{N-1}{2}$ a linear phase results if

$$\underline{\underline{\alpha = \frac{N-1}{2}}}$$

The condition $\alpha = \frac{N-1}{2}$ ensures $h(n)$ is symmetric about its mid pt.

Following imp points may be noted:

→ The cut-off frequencies ω_c depends upon the type of the window used

→ Transition width of FIR filter is approximately equal to width of main lobe of frequency response of window used

Table gives approximately transition width and minimum stop band attenuation for diff types of windows used in FIR filter table.

Type	transition width ($\Delta\omega$) in rad	min. attenuation in stop band
① Rectangular	$\frac{4\pi}{N}$	21 dB
② Bartlett	$\frac{8\pi}{N}$	25 dB
③ Hanning	$\frac{8\pi}{N}$	44 dB
④ Hamming	$\frac{8\pi}{N}$	53 dB
⑤ Blackman	$\frac{12\pi}{N}$	74 dB

→ Let K_p, W_p and K_s, W_s represent the pass and stop band specifications respectively. Then an FIR filter is designed using the following iterative procedure:

- ① Select the type of window to be the one highest of the list, such that the stop band attenuation exceeds $-K_s$ dB
- ② Size of the window is found using the eqn
 $W_s - W_p \geq K \frac{2\pi}{N}$; where $K=2$ for rectangular
 $K=4$ for Bartlett, Hanning and Hamming window
and $K=6$ for Blackmann window

③ Impulse response of low pass FIR filter is given by
 $h(n) = hd(n)w(n) ; 0 \leq n \leq N-1$
 $\hookrightarrow \textcircled{1}$

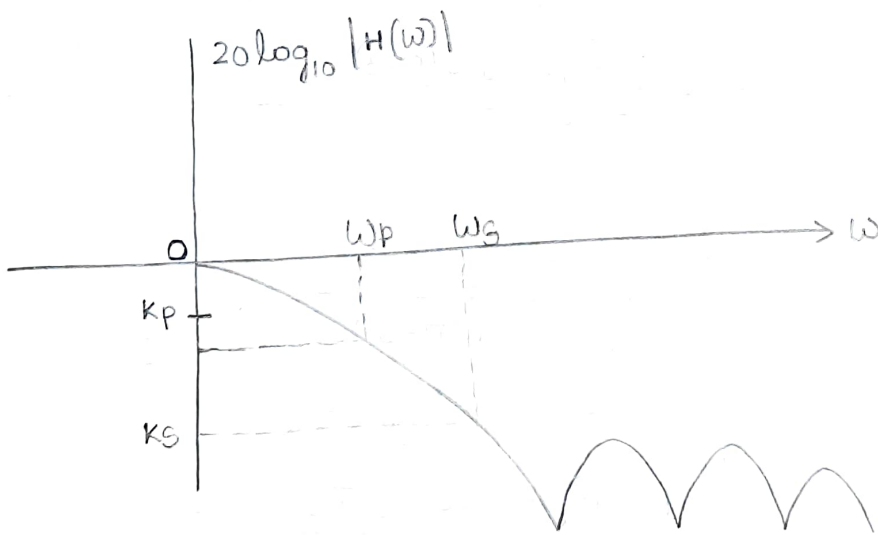
In eqn ①, $hd(n) = \begin{cases} \frac{\sin \omega_c(n-d)}{\pi(n-d)} ; n \neq d \\ \frac{\omega_c}{\pi} ; n = d \end{cases}$

For 1st trial impulse response, we choose $d = \frac{N-1}{2}$ and $\omega_c = \omega_p$. Since N is odd, frequency response of FIR filter is computed using expression given below:

$$H(\omega) = e^{-j\omega \left(\frac{N-1}{2}\right)} \left\{ h\left(\frac{N-1}{2}\right) + \sum_{n=0}^{N-3/2} 2h(n) \cos\left[\omega\left(\frac{N-1}{2} - n\right)\right] \right\}$$

$\hookrightarrow \textcircled{2}$

Using eqn ① and ②, we plot first trial frequency response and it look approximately as shown in fig below:



As expected, at $\omega = \omega_p$, we get a passband attenuation greater than $-K_p$ dB. Hence, we increase ω_c slightly and then using eqn ① and ②, second trial frequency response of FIR filter is plotted. Again check for pass band requirement. If not satisfied. Repeat this procedure. Till pass-band and stop-band requirement are met

→ Once ω_c is fixed, try decreasing N , so that passband and stop band requirements are not disturbed. In this way N and α are optimised.

→ Sub the values of ω_c and optimised α in eqn (1) and compute FIR filter coefficient $h(n)$ for $n=0, 1, \dots, N$.

① Frequency response of desired low pass filter is given by $H_d(\omega) = \begin{cases} e^{-j3\omega} & ; 0 < |\omega| < \frac{\pi}{4} \\ 0 & ; \frac{\pi}{4} < |\omega| < \pi \end{cases}$

Design an FIR filter using hamming window. Also find an expression for frequency response of designed FIR filter?

S: WKT $h_d(n) = \frac{1}{2\pi} \int_{-\pi}^{\pi} H_d(\omega) e^{j\omega n} d\omega$

$$\Rightarrow h_d(n) = \frac{1}{2\pi} \int_{-\pi/4}^{\pi/4} e^{j\omega(n-3)} d\omega$$

$$h_d(n) = \frac{1}{2\pi} \frac{\sin \frac{\pi}{4}(n-3)}{\pi(n-3)} ; n \neq 3$$

Setting $n=3$ in eqn (1) we get

$$h_d(n) = \frac{\omega_c}{\pi} = \frac{\pi}{4 \times \pi} = \frac{1}{4}$$

Summarising the results we get

$$h_d(n) = \begin{cases} \frac{\sin \frac{\pi}{4}(n-3)}{\pi(n-3)} & ; n \neq 3 \\ \frac{1}{4} & ; n = 3 \end{cases}$$

only for low pass filter

From exp for $H_d(\omega)$, we find that $\alpha=3$

$$\therefore \alpha = \frac{N-1}{2} = 3 \Rightarrow \underline{N=7}$$

Impulse response of low pass FIR filter is

$$h(n) = h_d(n) w_{\text{ham}}(n) ; 0 \leq n \leq 6$$

$$w_{\text{ham}}(n) = \begin{cases} 0.54 - 0.46 \cos\left(\frac{3\pi n}{N-1}\right) & ; 0 \leq n \leq N-1 \\ 0 & ; \text{otherwise} \end{cases}$$

n	$h_d(n)$	$w_{\text{ham}}(n)$	$h(n)$
0	0.075	0.08	0.006
1	0.15915	0.31	0.04933
2	0.22508	0.77	0.173816
3	1/4	1	1/4 = 0.25
4	0.22508	0.77	0.17325
5	0.15915	0.31	0.04933
6	0.075	0.08	0.006

Since N is odd, frequency response of low pass FIR filter is calculated using the expression

given below:

$$H(\omega) = e^{-j\omega\left(\frac{N-1}{2}\right)} \left[h\left(\frac{N-1}{2}\right) + \sum_{n=0}^{N-3/2} 2h(n) \cos\left[\omega\left(\frac{N-1}{2}\right) - n\right] \right]$$

$$\Rightarrow H(\omega) = e^{-j3\omega} \left[h(3) + \sum_{n=0}^2 2h(n) \cos\left[\omega\left(\frac{N-1}{2}\right) - n\right] \right]$$

$$H(\omega) = e^{-j3\omega} \left[h(3) + 2h(0) \cos 3\omega + 2h(1) \cos 2\omega + 2h(2) \cos \omega \right]$$

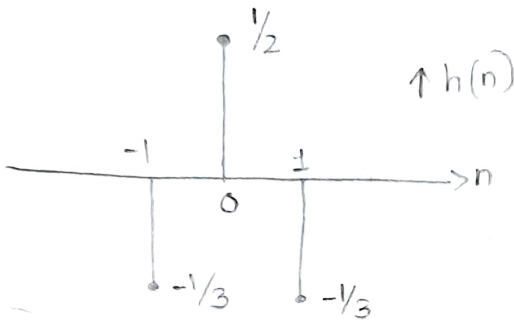
$$H(\omega) = e^{-j3\omega} \left[0.25 + 0.012 \cos 3\omega + 0.09866 \cos 2\omega + 0.3466 \cos \omega \right]$$

2) An FIR filter is specified by the following unit sample response:

$$h(n) = -\frac{1}{3} \delta(n+1) + \frac{1}{2} \delta(n) - \frac{1}{3} \delta(n-1)$$

- (a) Is it a ~~symmetry~~ filter and linear phase filter? Explain
 (b) Is it a causal filter? why or why not?
 (c) Is it a low pass filter? If not, find the type of filter

S:



- (a) Since $h(n)$ exhibits even symmetry about $n=0$ [mid pt], FIR filter will give a linear phase
 (b) Since $h(n) \neq 0$ for $n < 0$, filter is non causal.
 (c) DTFT of $h(n)$ is

$$H(e^{j\omega}) = H(\omega) = \sum_{n=-\infty}^{\infty} h(n) e^{j\omega n}$$

$$\Rightarrow H(\omega) = \sum_{n=-\infty}^{\infty} \left[-\frac{1}{3} \delta(n+1) + \frac{1}{2} \delta(n) - \frac{1}{3} \delta(n-1) \right] e^{j\omega n}$$

By applying sifting rule, we get

$$\Rightarrow H(\omega) = -\frac{1}{3} e^{-j\omega} + \frac{1}{2} e^{j\omega} - \frac{1}{3} e^{j\omega}$$

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

$$H(\omega) = \frac{1}{2} - \frac{1}{3} [2 \cos \omega]$$

$$\Rightarrow |H(\omega)| = \left| \frac{1}{2} - \frac{2}{3} \cos \omega \right|$$

=

13, 27, 32, 43, 58
 ↓ ↓ ↓ ↓ ↓
 Rec best hamming

$H(j\omega)$ gain

21, 25, 44, 53, 74

$$w_R(n) = \begin{cases} 1, & 0 \leq n \leq N-1 \end{cases}$$

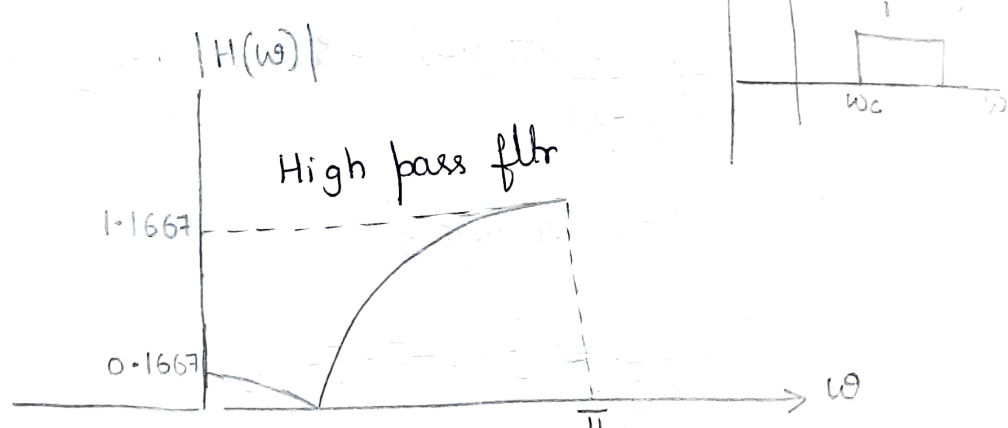
$$w_b(n) = \begin{cases} 1 - 2 \left[\frac{n - (N-1)/2}{N-1} \right]^2, & 0 \leq n \leq N-1 \end{cases}$$

$$w_h(n) = 0.5 - 0.5 \cos \left(\frac{2\pi n}{N} \right)$$

$$w_{hm}(n) = 0.54 - 0.46 \cos \left(\frac{2\pi n}{N} \right)$$

$$w_{blac}(n) = 0.42 + 0.45 \cos \left(\frac{2\pi n}{N} \right) + 0.14 \cos \left(\frac{4\pi n}{N} \right)$$

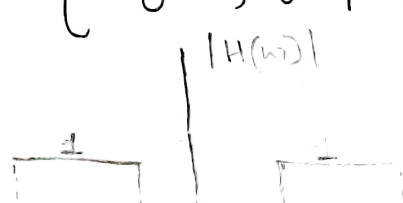
ω	$ H(\omega) $
0	+0.16667
0.1π	+0.13403
0.2π	+0.03934
→ $0.23\pi = 0$	
0.3π	0.10814
0.4π	0.2939
0.5π	0.5
0.6π	0.7060
0.7π	0.8918
0.8π	1.0393
0.9π	1.134
π	1.16667



3) Design a high pass FIR filter using Hamming window. Take $N=7$ and $\omega_c = 2\pi$ rad. Also find an expression for frequency response of designed high pass FIR filter?

S: Frequency response of an ideal high pass filter is

$$H_d(\omega) = \begin{cases} e^{-j\omega d} & ; \omega_c < |\omega| < \pi \\ 0 & ; 0 < |\omega| < \omega_c \end{cases}$$



To find $hd(n)$:

$$hd(n) = \frac{1}{2\pi} \int_{-\pi}^{\pi} Hd(\omega) e^{j\omega n} \cdot d\omega$$

$$hd(n) = \frac{1}{2\pi} \int_{-\pi}^{-\omega_c} e^{j\omega(n-d)} \cdot d\omega + \int_{\omega_c}^{\pi} e^{j\omega(n-d)} \cdot d\omega \rightarrow (1)$$

$$hd(n) = \frac{1}{2\pi} \left[\frac{e^{j\omega(n-d)}}{j(n-d)} \right]_{-\pi}^{-\omega_c} + \frac{1}{2\pi} \left[\frac{e^{j\omega(n-d)}}{j(n-d)} \right]_{\omega_c}^{\pi}$$

$$= \frac{1}{2\pi} \left[\frac{e^{-j\omega_c(n-d)} - e^{-j\pi(n-d)}}{-e^{-j\pi(n-d)} + e^{-j\omega_c(n-d)}} \right]$$

$$= \frac{1}{2\pi(n-d)} \left[\frac{e^{j\pi(n-d)} - e^{-j\pi(n-d)}}{2j} - \frac{e^{j\omega_c(n-d)} - e^{-j\omega_c(n-d)}}{2j} \right]$$

$$= \frac{1}{\pi(n-d)} \left[\sin \pi(n-d) - \sin \omega_c(n-d) \right] ; n \neq d$$

Letting $n=d$ in eqn (1)

$$hd(n) = \frac{1}{2\pi} \times \pi [-\omega_c + \pi + \pi - \omega_c]$$

$$hd(n) = \left[\frac{\pi - \omega_c}{\pi} \right] //$$

Summarising the results, we get

$$hd(n) = \begin{cases} \frac{\sin \pi(n-d) - \sin \omega_c(n-d)}{\pi(n-d)} ; n \neq d \\ \frac{\pi - \omega_c}{\pi} ; n = d \end{cases}$$

Impulse response of high pass FIR filter is

$$h(n) = hd(n) w(n)$$

$$\text{where } w_{ham}(n) = \begin{cases} 0.54 - 0.46 \cos\left(\frac{2\pi n}{N-1}\right) & 0 \leq n \leq N-1 \\ 0 & \text{elsewhere} \end{cases}$$

n	hd(n)	wharm(n)	h(n)
0	0.02964	0.08	0.00231
1	0.12045	0.31	0.0313
2	-0.2894	0.77	-0.2228
3	0.3633	1	0.3633
4	-0.2894	0.77	-0.2228
5	0.12045	0.31	0.0313
6	0.2894	0.08	0.00231

VERIUS
DATE: / / 200
FIR filter

since N is odd, $h(n)$ is even symmetric about its midpoint and $h(n)$ is odd, T frequency response of the ~~low~~ high pass filter \rightarrow FIR is

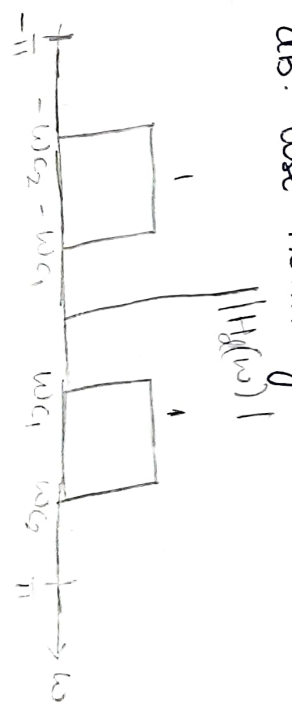
$$H(\omega) = e^{-j\omega \left(\frac{N-1}{2}\right)} \left[\sum_{n=0}^{N-1/2} h(n) \cos(\omega \left(\frac{N-1}{2} - n\right)) + \sum_{n=0}^{N-1/2} 2h(n) \cos(\omega \left(\frac{N-1}{2} - n\right)) \right]$$

$$\Rightarrow H(\omega) = e^{-3\omega j} \left[h(3) + \sum_{n=0}^2 a h(n) \cos(\omega(3-n)) \right]$$

$$\Rightarrow H(\omega) = e^{-j3\omega} \left[h(3) + 2h(0) \cos 3\omega + 2h(1) \cos 2\omega + 2h(2) \cos \omega \right]$$

$$\Rightarrow H(\omega) = e^{-j3\omega} \left[0.3633 + 0.00474 \cos 3\omega + 0.0746 \cos 2\omega - 0.4456 \cos \omega \right]$$

4) Design an band-pass FIR filter for the following specification: $N=7$, $\omega_{c1}=1 \text{ rad}$, $\omega_{c2}=2 \text{ rad}$ also find the magnitude of frequency response at $\omega=1.5 \text{ rad}$ in dB. Use Hanning window?



3:

Frequency response of ideal band pass filter is

$$H_d(\omega) = \begin{cases} e^{-j\omega\alpha} & ; \omega_{c1} < |\omega| < \omega_{c2} < \pi \\ 0 & ; \text{otherwise} \end{cases}$$

WKT

$$h_d(n) \triangleq \frac{1}{2\pi} \int_{-\pi}^{\pi} H_d(\omega) e^{j\omega n} \cdot d\omega$$

$$h_d(n) = \frac{1}{2\pi} \int_{-\omega_{c2}}^{-\omega_{c1}} e^{-j\omega(n-\alpha)} \cdot d\omega + \int_{\omega_{c1}}^{\omega_{c2}} e^{j\omega(n-\alpha)} \rightarrow \text{①}$$

$$h_d(n) = \frac{1}{\pi(n-\alpha)} [\sin \omega_{c2}(n-\alpha) - \sin \omega_{c1}(n-\alpha)] ; n \neq \alpha$$

Letting $n = \alpha$ in eqn ① we get

$$h_d(n) = \frac{1}{2\pi} [\omega_{c1} + \omega_{c2} + \omega_{c2} - \omega_{c1}]$$

$$h_d(n) = \frac{\omega_{c2} - \omega_{c1}}{\pi}$$

Impulse response of the band pass FIR filter is given by

$$h(n) = h_d(n) w_{han}(n) ; 0 \leq n \leq N-1 = 6$$

Here

$$w_{han}(n) = \begin{cases} \frac{1}{2} \left[1 - \cos\left(\frac{2\pi n}{N-1}\right) \right] & ; 0 \leq n \leq N-1 \\ 0 & ; 0 < n \end{cases}$$

$$\alpha = \frac{N-1}{2} = \frac{6-1}{2}$$

n	$h_d(n)$	$w_{han}(n)$	$h(n)$
0	-0.0446	0	0
1	-0.2651	0.25	-0.066275
2	0.02159	0.75	0.0161925
3	0.3183	1	0.3183
4	0.02159	0.75	0.0161925
5	-0.2651	0.25	-0.066275
6	-0.0446	0	0

Since N is odd and $h(n)$ exhibits even symmetry
 Frequency response of the band pass FIR filter is

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$$H(\omega) = e^{-j\omega \frac{(N-1)}{2}} \left[h\left(\frac{N-1}{2}\right) + \sum_{n=0}^{\frac{N-3}{2}} 2h(n) \cos\left(\omega\left(\frac{N-1}{2} - n\right)\right) \right]$$

$$|H(\omega)| = \left| h\left(\frac{N-1}{2}\right) + \sum_{n=0}^{\frac{N-3}{2}} 2h(n) \cos(\omega(\frac{N-1}{2} - n)) \right|$$

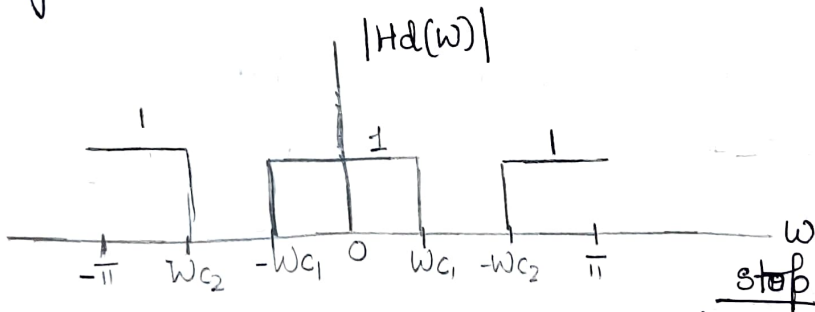
$$|H(\omega)| = \left| h(3) + 2h(0) \cos 3\omega + 2h(1) \cos 2\omega + 2h(2) \cos \omega \right|$$

$$|H(\omega)| = \left| 0.3183 + 0.1326 \cos 2\omega + 0.03238 \cos \omega \right|$$

$$20 \log_{10} |H(\omega)|_{\omega=1.5} = -6.609 \text{ dB}$$

5) Design a band reject FIR filter to meet the following specifications: $\omega_{c1} = 1 \text{ rad}$, $\omega_{c2} = 2 \text{ rad}$, $N = 7$. Use rectangular window?

S:



Frequency response of ideal band reject filter is

$$H_d(\omega) = \begin{cases} e^{-j\omega d} & ; \text{ otherwise} \\ 0 & ; \omega_{c1} < |\omega| < \omega_{c2} \end{cases}$$

WKT

$$h_d(n) \triangleq \frac{1}{2\pi} \int_{-\pi}^{\pi} H_d(\omega) e^{j\omega n} d\omega$$

$$h_d(n) \triangleq \frac{1}{2\pi} \int_{-\pi}^{-\omega_{c2}} e^{j\omega(n-d)} d\omega + \frac{1}{2\pi} \int_{-\omega_{c1}}^{\omega_{c1}} e^{j\omega(n-d)} d\omega + \frac{1}{2\pi} \int_{\omega_{c2}}^{\pi} e^{j\omega(n-d)} d\omega$$

$$h_d(n) = \frac{1}{2\pi} \left[\left[\frac{e^{j\omega(n-d)}}{j(n-d)} \right]_{-\pi}^{-\omega_{c2}} + \left[\frac{e^{j\omega(n-d)}}{j(n-d)} \right]_{-\omega_{c1}}^{\omega_{c1}} + \left[\frac{e^{j\omega(n-d)}}{j(n-d)} \right]_{\omega_{c2}}^{\pi} \right]$$

~~0.03238~~ 0.032

150

$$\Rightarrow hd(n) = \frac{1}{\pi(n-d)} \left[\sin \pi(n-d) + \sin \omega_{c1}(n-d) - \sin \omega_{c2}(n-d) \right] ; n \neq d$$

Letting $n=d$ we get

$$hd(n) = \frac{1}{2\pi} \left[\int_{-\pi}^{-\omega_{c2}} dw + \int_{-\omega_{c1}}^{\omega_{c1}} dw + \int_{\omega_{c2}}^{\pi} dw \right]$$

$$= \frac{-\omega_{c2} + \pi + \omega_{c1} + \omega_{c1} + \pi - \omega_{c2}}{2\pi}$$

$$hd(n) = \frac{\omega_{c1} - \omega_{c2} + \pi}{\pi} \quad n=d$$

Hence, the impulse response of band reject FIR filter is

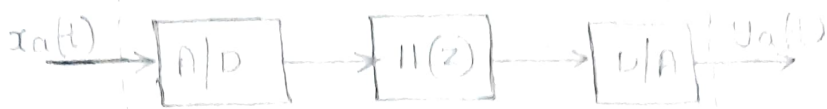
$$h(n) = hd(n) w_n(n) ; 0 \leq n \leq N-1=6$$

$$\text{where } w_n(n) = \begin{cases} 1, & 0 \leq n \leq 6 \\ 0, & \text{o.w} \end{cases}$$

n	hd(n)	w _n (n)	h(n)
0	+0.044	1	+0.044
1	0.265	1	0.265
2	-0.0215	1	-0.0215
3	0.6817	1	0.6817
4	0.265	1	-0.0215
5	0.265	1	0.265
6	+0.044	1	0.044

67) Design a low pass digital filter to be used in A/D - $H(z)$ - D/A structure that will have -3 dB cut-off at 30π rad/s and an attenuation of 50 dB at 45π rad/s. Filter is required to have a linear phase and system uses sampling rate of 100 sample per second.

S:



Equivalent analog filter $H_{eq}(s)$

In the problem, specifications $H_{eq}(s)$ are given and they are as follows:

$$K_p = -3 \text{ dB}, \quad \Omega_p = 30\pi \text{ rad/s}$$

$$K_s = -50 \text{ dB}, \quad \Omega_s = 45\pi \text{ rad/s} \quad \text{and} \quad T = \frac{1}{100} \text{ sec}$$

① Convert band edge analog frequency into digital frequencies using the formula

$$\omega = \Omega T ; \quad T = 1/100 \text{ s}$$

$$\omega_p = \Omega_p T = 0.3\pi ; \quad K_p = -3 \text{ dB}$$

$$\omega_s = \Omega_s T = 0.45\pi ; \quad K_s = -50 \text{ dB}$$

② Prewarp

Type	Transition width ($\Delta\omega$)	min stop-band attenuation
Rectangular	$4\pi/N$	21 dB
Bartlett	$8\pi/N$	25 dB
Hanning	$8\pi/N$	44 dB
Hamming	$8\pi/N$	53 dB
Blackman	$12\pi/N$	74 dB

To meet a stop band attenuation of 50dB, we have two choices in the form of hamming and blackman window. Since blackman has higher transition width compare hamming window, we choose hamming window

③ Size of the window is selected using the relation given below

$$W_s - W_p \geq \frac{2\pi}{N} \cdot K$$

$$\frac{4 \times 2\pi}{K} = \frac{8\pi}{N}$$

For hamming window : $K=4$

hence $0.45\pi - 0.3\pi \geq \frac{8\pi}{N}$

$$N \geq \frac{8}{0.15}$$

$$N \geq 53.33$$

Selecting N to be next higher odd integer
 $N=55$ $\left[d = \frac{N-1}{2} = 27 \right]$ must be an integer

④ Let the frequency response of low pass ideal filter be

$$H_d(\omega) = \begin{cases} e^{-j\omega d} & ; |\omega| < \omega_c \\ 0 & ; \omega_c < |\omega| < \pi \end{cases}$$

$$h_d(n) \triangleq \frac{1}{2\pi} \int_{-\omega_c}^{\omega_c} H_d(\omega) e^{j\omega n} d\omega \quad e^{j\omega d}$$

$$h_d(n) = \frac{1}{2\pi} \int_{-\omega_c}^{\omega_c} e^{j\omega(n-d)} d\omega$$

$$h_d(n) = \frac{\sin \omega_c(n-d)}{\pi(n-d)} ; n \neq d$$

Putting $n=d$ in eqn ①, we get

$$h_d(d) = \frac{1}{2\pi} \int_{-\omega_c}^{\omega_c} 1 \cdot d\omega = \frac{\omega_c}{\pi}$$

Impulse response of low pass FIR filter is

$$h(n) = h_d(n) \text{Wham}(n) ; 0 \leq n \leq N-1 = 54$$

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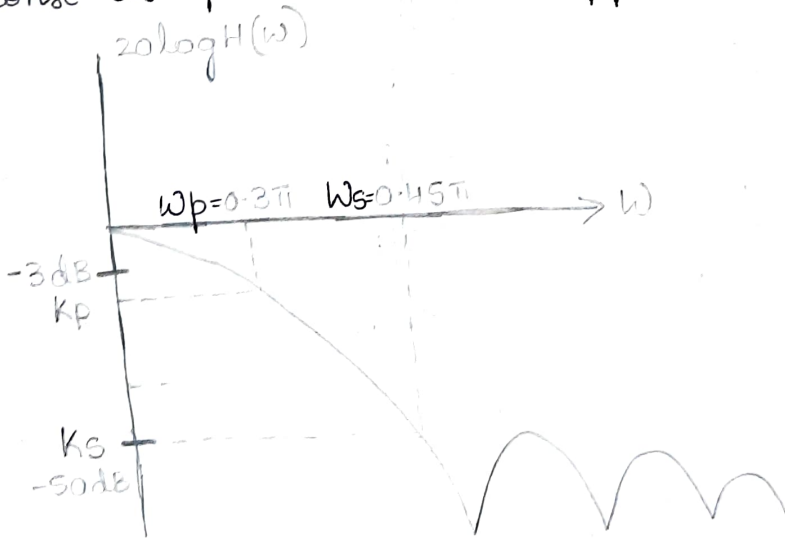
$$\Rightarrow h(n) = \begin{cases} \frac{\sin \omega_c (n-d)}{\pi (n-d)} \times 0.54 - 0.46 \cos\left(\frac{2\pi n}{N-1}\right) ; n \neq d \\ \frac{\omega_c}{\pi} \times 0.5 \pm ; n = d \end{cases} \rightarrow (2)$$

For first trial, impulse response, we choose $\omega_c = \omega_p = 0.3\pi$ and $d = \frac{N-1}{2} = \underline{27}$

Since N is odd, frequency response of low pass FIR filter is computed using the express given below

$$H(\omega) = e^{-j\omega \left(\frac{N-1}{2}\right)} \left\{ h\left(\frac{N-1}{2}\right) + \sum_{n=0}^{N-3/2} 2h(n) \cos \omega \left(\left(\frac{N-1}{2}\right) - n\right) \right\}$$

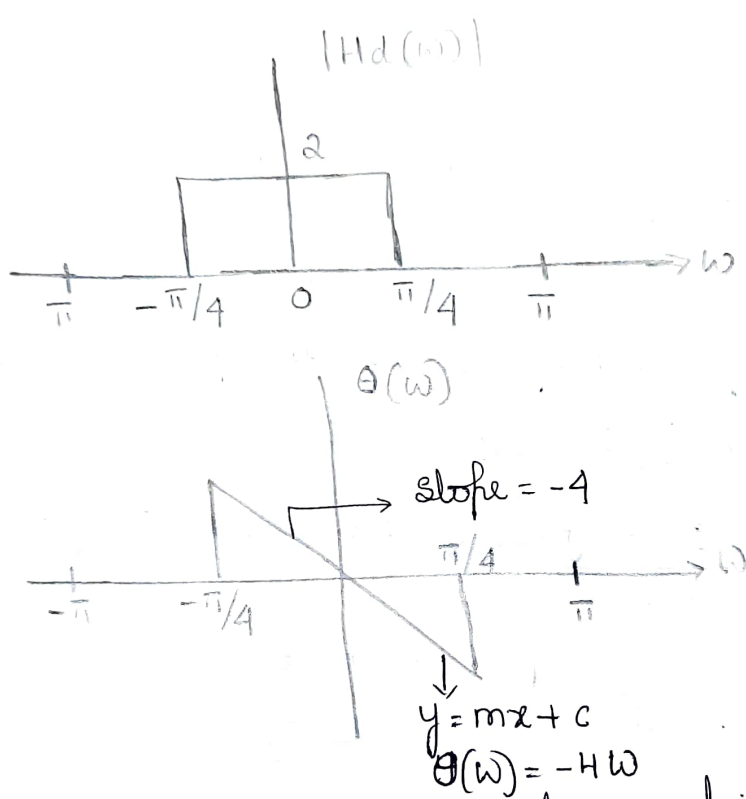
Using eqn (2) and (3), first trial frequency response is plotted and it appears as shown below $\rightarrow (3)$



It is found from above fig that $\omega_p = 0.3\pi$, the pass band attenuatn is greater than -3dB. Hence, slightly increase ω_c and then eqn (2) and (3) plot mag response. check for passband requirement. Continue this procedure till at $\omega_p = \omega_p$ the pass band attenuatn less than or equal -3dB (matlab ans $\omega_c = 0.33\pi$)

7) Fig below shows magnitude and phase response of an ideal low pass filter. Find $hd(n)$

- b) Suppose $hd(n)$ is truncated by a hamming window. What is $h(n)$?
- c) Roughly sketch the frequency response of low pass FIR filter designed in part b. Ensure to indicate the values of cut-off frequency and transition and stop band attenuation



S: In general, $H_d(w)$ is complex and is given by

$$H_d(w) = |H_d(w)| e^{j\theta(w)}$$

$$= \begin{cases} 2 e^{-j4w} & ; -\frac{\pi}{4} \leq w \leq \frac{\pi}{4} \\ 0 & ; \frac{\pi}{4} < |w| < \pi \end{cases}$$

WKT

$$hd(n) \triangleq \frac{1}{2\pi} \int_{-\pi}^{\pi} H_d(w) e^{jwn} dw$$

$$hd(n) = \frac{1}{2\pi} \int_{-\pi/4}^{\pi/4} 2 e^{jw(n-4)} dw \rightarrow \textcircled{1}$$

$$hd(n) = \frac{2 \sin \frac{\pi}{4} (n-4)}{\pi (n-4)} ; n \neq 4$$

Letting $n=4$ in eqn ① we get

$$hd(n) = \frac{1}{2\pi} \left[\frac{2\pi}{4} \right] = \frac{1}{2} \times 1 = \underline{\underline{\frac{1}{2}}}$$

⑥ WKT
 $h(n)$

Impulse response of low pass FIR filter is
 $h(n) = hd(n) w_{\text{hamm}}(n) ; 0 \leq n \leq N-1 = 8$

[NOTE : $\alpha=4 \Rightarrow \frac{N-1}{2} = \alpha \Rightarrow N = \underline{\underline{9}}$]

where, $w_{\text{Ham}}(n) = \begin{cases} 0.54 - 0.46 \cos\left(\frac{2\pi n}{N-1}\right) ; 0 \leq n \leq N-1 \\ 0 ; \text{otherwise} \end{cases}$
 $= 8$

n	hd(n)	$w_{\text{ham}}(n)$	$h(n)$
0	0	0.08	0
1	0.15	0.2147	0.0322
2	0.318	0.54	0.17172
3	0.45	0.8653	0.3893
4	1/2	1	0.5
5	0.45	0.8653	0.3893
6	0.318	0.54	0.17172
7	0.15	0.2147	0.0322
8	0	0.08	<u>0</u>

⑦ Since N is odd and $h(n)$ has even sym

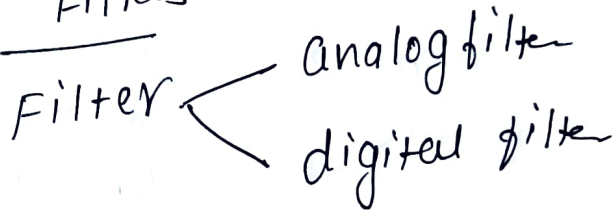
IIR Filter design.

Characteristics of commonly used analog filters.
- Butterworth & Chebyshev filters. Analog to analog freq transformations.

Introduction:

A Filter is one which rejects unwanted freq's from the i/p signals & allows the desired freq's to obtain the required (Phase) o/p signal

Types of Filters



Analog filter - i/p & o/p are continuous time signals
Digital filter - " " " " discrete-time signals

* A digital filter is generally a discrete LTI system which approximates a freq response i.e. desired with its i/p being digital samples & o/p also digital samples

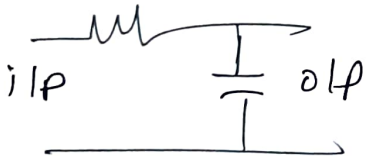
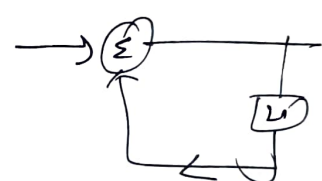
* Types of Digital Filter

IIR
[Infinite Impulse Response]
all recursive type where present o/p depends on present i/p, past i/p & o/p samples

FIR
[Finite Impulse Response]
all non recursive type where present o/p depends upon present i/p & past i/p samples

Classification of ^{Analog} Filter in accordance with the freq selective characteristics as
 Low pass filter, Band pass
 High pass filter & band stop filter

Comparison bet' Analog & Digital filter

Analog	Digital
(1) I/O's are continuous time signals	(1) I/O's are discrete time signals
(2) Implementation of these filters is carried out using passive components	(2) These are implemented on a digital computer or HLB using DSP elements such as adder, x^k & delay
	
(3) Analog filter theory is based on linear differential eqn	(3) based on linear difference eqn
(4) Laplace transforms are used for analysis in 's' - plane	(4) z - transforms are used for analysis in z - plane
(5) <u>disadv</u> higher noise sensitivity non linearity lack of flexibility	(5) Digital filter requires additional ADC/DAC converters
(6) Environmental, Parameters interference, affects the performance & noise	(6) freq range is restricted to half the sampling rate
	(7) negligible effect of environmental parameters & interference noise & others

* In this chapter the design of IIR filter that are realizable & stable are all discussed in detail.

* The impulse response $h(n)$ for a realizable filter is

$$h(n) = 0 ; n \leq 0 \rightarrow \textcircled{1}$$

& for stability it must satisfy the condition

$$\sum_{n=0}^{\infty} |h(n)| < \infty \rightarrow \textcircled{2}$$

IIR digital filters have transfer fun of the form

$$H(z) = \frac{\sum_{n=0}^{\infty} h(n) z^{-n}}{1 + \sum_{k=1}^N a_k z^{-k}} = \frac{\sum_{k=0}^M b_k z^{-k}}{1 + \sum_{k=1}^N a_k z^{-k}} \rightarrow \textcircled{3}$$

* The design of an IIR filter for given specifications is finding filter coefficients a_k & b_k of eq $\textcircled{3}$.

PS Types of Analog Filter :-

A filter is one which rejects unwanted freq from the i/p signal & allows the desired freq's to obtain the required shape of o/p signal

Pass band \rightarrow The range of freq's of signal that are passed thro' the filter

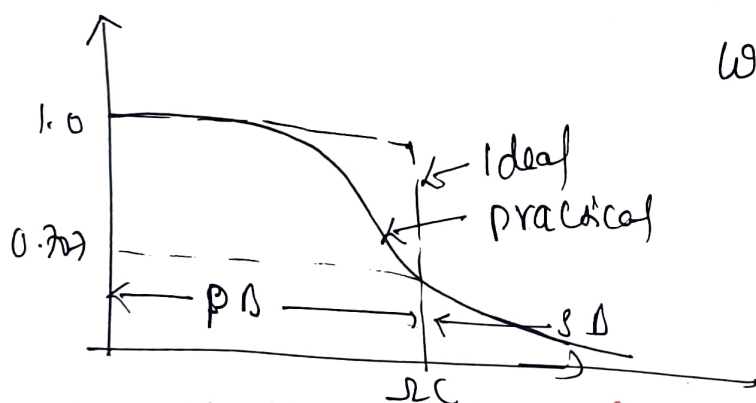
Stop band \rightarrow those freq's that are blocked

Analog

* The ~ filters are of different types based on

- (1) their magnitude response - LP, HP, BP, RS
- (2) their cut off freq
- (3) shape of their amplitude response
- (4) shape of their phase response
- (5) nature of device used $\left\{ \begin{array}{l} \text{Passive} \\ \text{Active} \end{array} \right.$

- (1) Low pass filter - allows low freq's to pass thro' it while attenuates high freq



Design of digital filters from analog filter

The most common technique used for designing IIR digital filters is known as indirect method - which involves first designing an analog prototype filter & then transforming the prototype to a digital filter

* for given specifications of digital filter, the derivation of digital filter transfer fun requires 3 steps

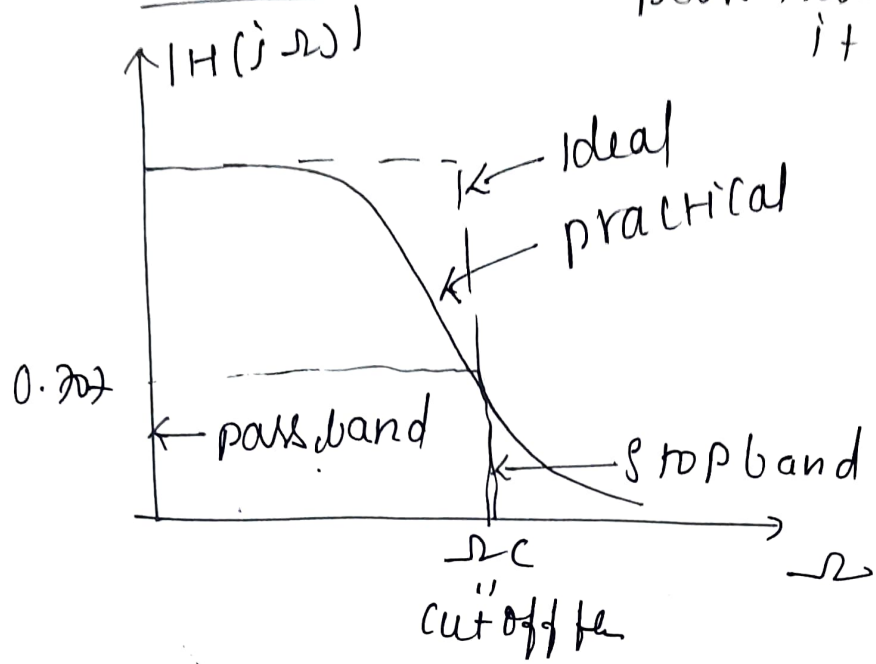
- (1) map the desired digital filter specifications into equivalent analog filter
- (2) derive the analog TF for the analog prototype
- (3) transform the TF of analog prototype into an equivalent digital filter transfer fun

Ideal & practical response of LP, HP, BP & BS filter

①

Low pass / -

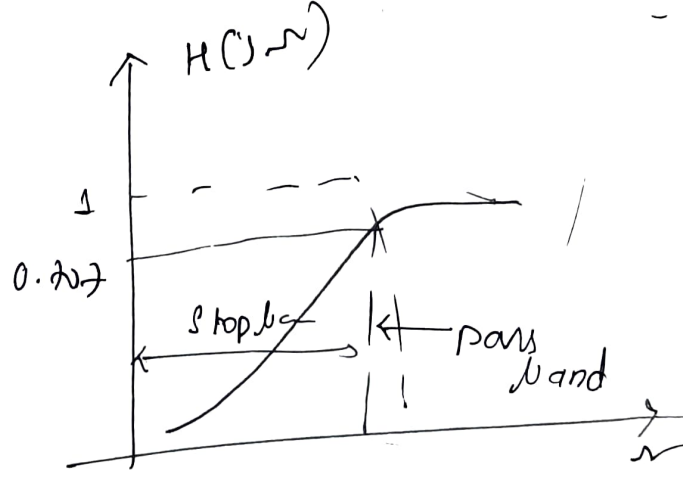
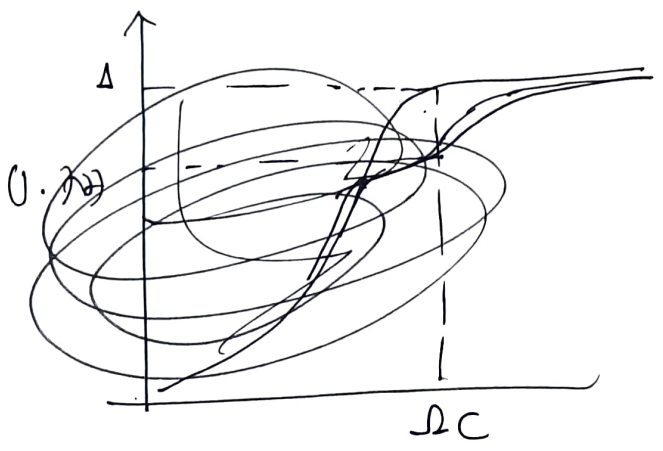
allows Low freq's to pass thro' it while it attenuates high freq



Cut off freq:
defined as freq where the gain has changed by some specific amount of ~~let~~ ^{attenuation} to mean mid band gain

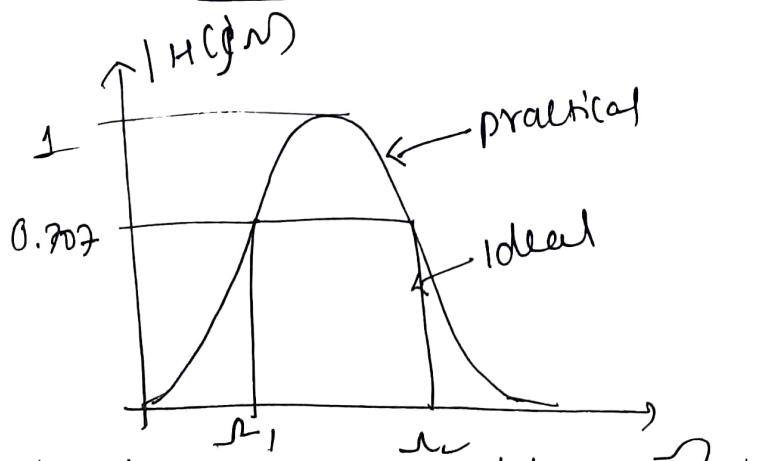
②

High pass



③

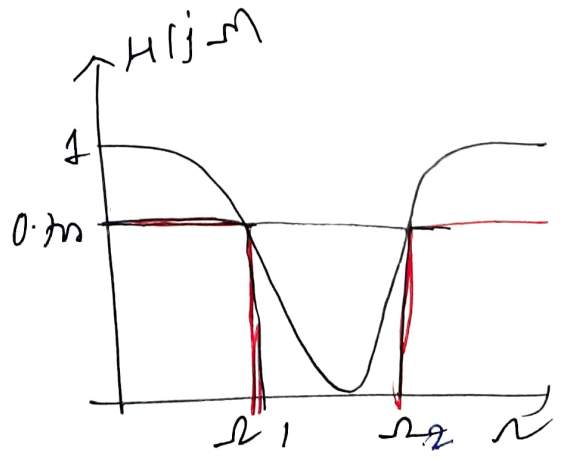
Band pass



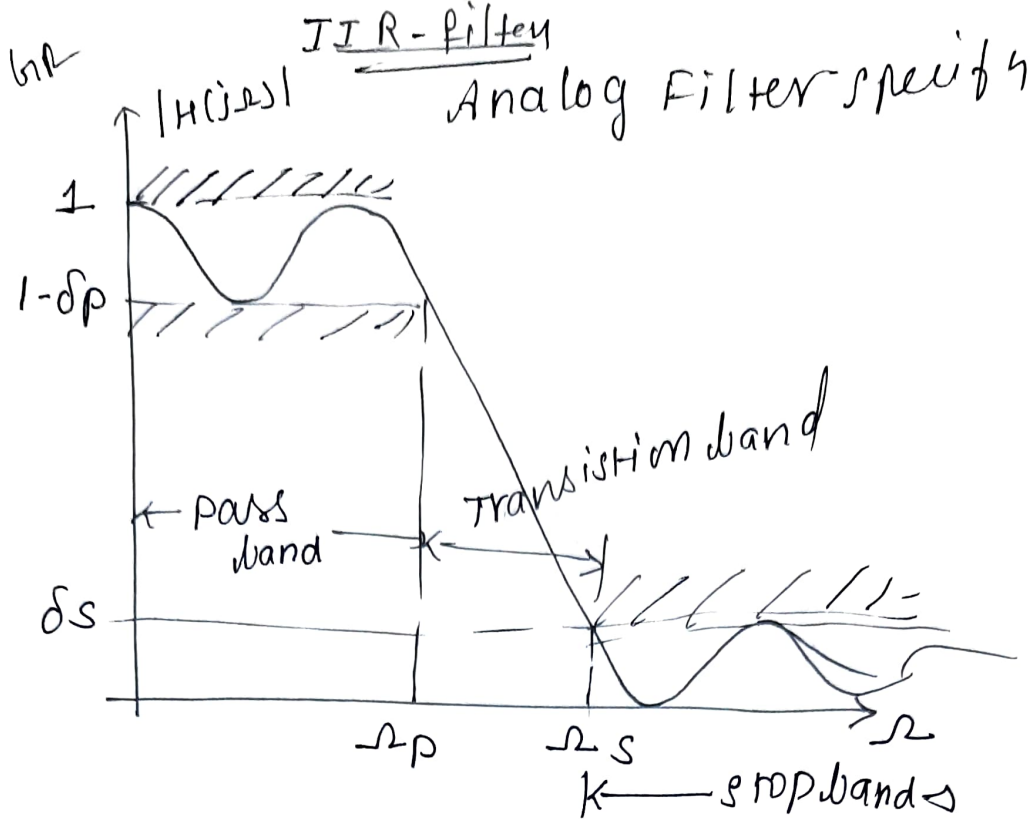
It allows only a band of freqs ω_1 to ω_2 to pass & stops all other freq

④

Band-Reject



It rejects all freq's bet' ω_1 & ω_2 & allows remaining freq



Specification of LPF

- * An important step in the design of an analog filter is the definition of the freq response specifications that should be satisfied by the filter freq response
- * These specifications describe how the filter reacts in the steady-state to sinusoidal i/p
- * fig above shows a typical magnitude freq response to LPF

pass band → The range of freq's of signal that are passed thro' the filter

stop band → those freq's that are blocked

$\omega_p \rightarrow$ pass band edge freq
 $\omega_s \rightarrow$ stop band edge freq.

(4)

* The freq range bet' ω_p & ω_s is called transition band where no specification is provided.

* The hatched area indicates forbidden magnitude values in these bands.

* In transition band, the magnitude rises monotonically in this band

* mathematical description of the freq response

$$1 - \delta_p \leq |H(j\omega)| \leq 1, \text{ for } 0 \leq \omega \leq \omega_p$$
$$0 \leq |H(j\omega)| \leq \delta_s \text{ for } |\omega| \geq \omega_s$$

* $\delta_p \rightarrow$ tolerance of magnitude response in pass band.

* The desired magnitude response in passband is 1.

~~$\delta_p \rightarrow$ pass band tolerance~~

* $\delta_s \rightarrow$ tolerance of magnitude response in stopband.

* The desired magnitude response in the stopband is 0.

* we define $\delta_p \rightarrow$ ~~pass band atten.~~

$$\textcircled{1} \quad A_p = -20 \log(1 - \delta_p)$$

as passband ripple in dB.

$$\textcircled{2} \quad K_p = -A_p = 20 \log(1 - \delta_p)$$

is defined as passband gain at $\omega = \omega_p$.

$\delta_s \rightarrow$ stopband attenuation.

$$\textcircled{3} \quad A_s = -20 \log \delta_s$$

is defined as stopband attenuation or ripple in dB

$$\textcircled{4} \quad K_s = -A_s = 20 \log_{10} \delta_s \quad \omega = \omega_s$$

is defined as stopband gain at $\omega = \omega_s$.

NOTE:-
 $\delta_p \rightarrow$ pass band tolerance/ripple
 $\delta_s \rightarrow$ stopband "
 $A_p \rightarrow$ pass band attenuation
 $A_s \rightarrow$ stopband "

* The main classes of analog filter

(5)

Butterworth filter

Chebyshev filter

* In this section the properties & design procedures for analog filters (Butterworth & Chebyshev) are presented along with the procedures & transform required to convert them into LPF, HPF, BPF, BSF.

[All pole approximation / Butterworth app] monotonic pass band & monotonic stop band

Butterworth Filters

They have smooth pass band with a relatively wide transition region. where as Chebyshev filter have sharp transition region & a not so smooth pass band.

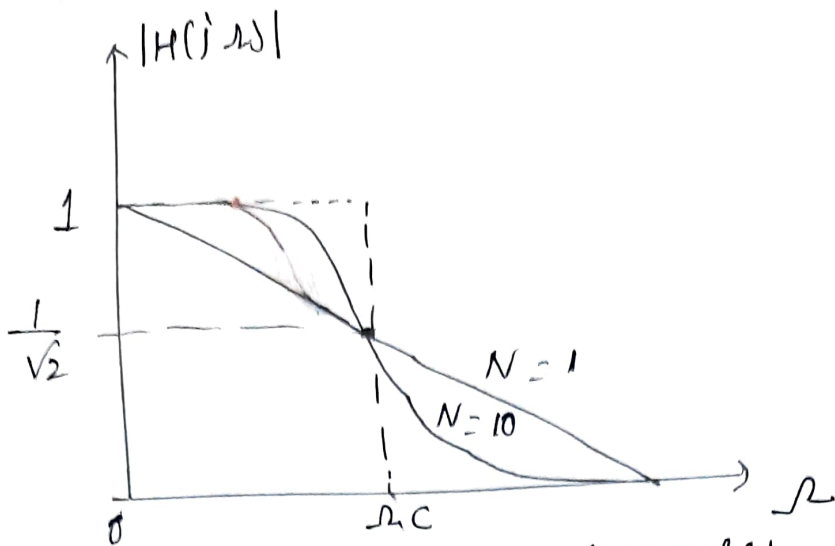
* A Butterworth filter is characterized by its magnitude freq response

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

where

$N \rightarrow$ order of the filter

$\omega_c \rightarrow$ cutoff freq where the filter magnitude is $\frac{1}{\sqrt{2}}$ times the dc gain ($\omega \rightarrow 0$)



Typical magnitude responses
of Butterworth filter

The following observations are made from
the above fig.:

1) $|H(j\Omega)| = 1$ for all N $\Omega = 0$

2) $|H(j\Omega)| = \frac{1}{\sqrt{2}}$ at $\Omega = \Omega_c$ for
all finite N .

This means that $20 \log_{10} |H(j\Omega_c)| = -3.01 \text{ dB}$

3) $|H(j\Omega)| \rightarrow 0$ as $\Omega \rightarrow \infty$

4) The magnitude characteristics is said to
be maximally flat 0°

$$\frac{d^n |H(j\Omega)|}{d\Omega^n} \bigg|_{\Omega=0} = 0 \quad \text{for } n=1, 2, \dots, 2N-1$$

5) $|H(j\Omega)|$ is a monotonically decreasing
fun of freq (Ω)

i.e., $|H(j\Omega_2)| < |H(j\Omega_1)|$ for any

values of Ω_1 & Ω_2 such that $0 \leq \Omega_1 < \Omega_2$

* The magnitude-square freq response of the normalized ($\Omega_c=1$) LP Butterworth filter is (6)

$$\boxed{|H_N(j\Omega)|^2 = \frac{1}{1 + \Omega^{2N}}} \quad \text{--- (1)}$$

$|A| = A \times A^*$

$$|H_N(j\Omega) H_N(-j\Omega)| = \frac{1}{1 + \Omega^{2N}} \rightarrow \text{(2)}$$

Replacing $j\Omega$ by s & hence $\Omega = \frac{s}{j}$ in eq (2) we get

$$H_N(s) H_N(-s) = \frac{1}{1 + \left(\frac{s}{j}\right)^{2N}} \rightarrow \text{(3)}$$

* The transfer fun $H_N(s) \cdot H_N(-s)$ has no finite zeros. ~~& $H_N(s)$ itself has no bi.~~

* The poles of the product $H_N(s) H_N(-s)$ are determined by equating the denominator to zero

$$\text{i.e., } 1 + \left(\frac{s}{j}\right)^{2N} = 0$$

$$\left(\frac{s}{j}\right)^{2N} = -1$$

$$\left(\frac{s}{j}\right) = (-1)^{1/2N}$$

$$s = (-1)^{1/N}$$

$$s = (-1)^{1/2N}, j \rightarrow (4)$$

WKT

$$-1 = e^{j\pi(2k+1)} \quad k = 0, 1, \dots, 2N-1$$

$$j = e^{j\pi/2}$$

The poles are given by

$$s_k = e^{j\frac{\pi(2k+1)}{2N}} \cdot e^{j\pi/2}$$

$$s_k = e^{j\left(\frac{k\pi}{N} + \frac{\pi}{2N} + \frac{\pi}{2}\right)} \rightarrow (5)$$

$$= 1 e^{j\theta_k}$$

$$= 1 \angle \theta_k$$

The poles of $H_N(s) H_N(-s)$ are the roots of the characteristic eqⁿ that lie on the circle of unit radius & are placed with at angles

$$\theta_k = \frac{k\pi}{N} + \frac{\pi}{2N} + \frac{\pi}{2}$$

$$0 \leq k \leq (2N-1)$$

(6)

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* The above analysis implies that the roots of the characteristic eqn are the poles of $H_N(s)H_N(-s)$ which lies on a unit circle.

* The transfer fun of normalized LP Butterworth filter is

$$H_N(s) = \frac{1}{\prod_{LHP} (s - s_k)} = \frac{1}{B_N(s)}$$

LHP
 ↓
 $s_k = \text{left hand poles}$

$B_N(s) = \text{Butterworth polynomial of order } N.$

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Table below shows first 5 Butterworth polynomial in a self factored form

N	$B_N(s)$
1	$s + 1$
2	$s^2 + \sqrt{2}s + 1$
3	$(s^2 + s + 1)(s + 1)$
4	$(s^2 + 0.76536s + 1)(s^2 + 1.84776s + 1)$
5	$(s + 1)(s^2 + 0.6180s + 1)(s^2 + 1.6180s + 1)$

(7)
 NO
 H1
 2
 1
 lo
 e
 no

1) $N=1$ [First order filter]

WKT

$$S_k = 1 \angle \theta_k = e^{j\theta_k}$$

$$\theta_k = \frac{k\pi}{N} + \frac{\pi}{2N} + \frac{\pi}{2}$$

$$k = 0, 1, \dots, 2N-1$$

for $N=1$, $k=0, 1$

$$\theta_k = k\pi + \frac{\pi}{2} + \frac{\pi}{2}$$

$$\boxed{\theta_k = k\pi + \pi}$$

(i) $k=0$ $S_0 = 1 \angle \theta_0 =$

$$S_0 = 1 \angle \theta_0$$

$$\theta_0 = (0)\pi + \pi = \pi$$

$$S_0 = 1 \angle \theta_0 = 1 \angle \pi = 1e^{j\pi} = \cos\pi + j\sin\pi$$
$$-1 + j0$$

$$\boxed{S_0 = -1}$$

(ii) $k=1$

$$S_1 = 1 \angle \theta_1$$

$$\theta_1 = \pi + \pi = 2\pi$$

$$S_1 = 1 \angle 2\pi = 1e^{j2\pi} = \cos 2\pi + j\sin 2\pi$$

$$1 + j0$$

$$\boxed{S_1 = 1}$$

$$B_N(s) = \pi (s - s_k)$$

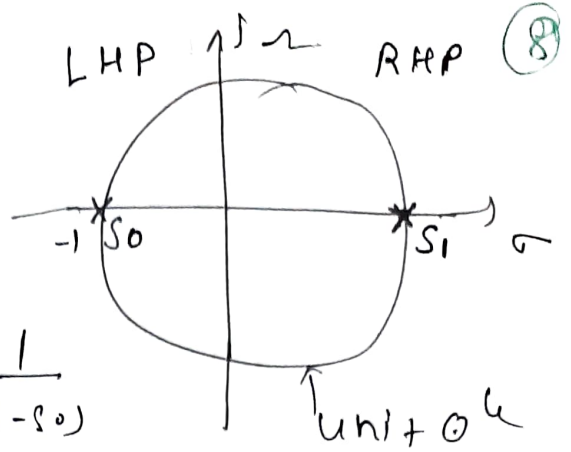
$$= \pi (s - s_0)$$

$$B_1(s) = s - (-1)$$

$$B_1(s) = (s + 1)$$

$$H_1(s) = \frac{1}{(s - s_0)}$$

$$= \frac{1}{s + 1}$$



2) $N = 2$ (2nd order filter)

for $N = 2$ $K = 0, 1, 2, 3$

$$\therefore \theta_K = \frac{K\pi}{N} + \frac{\pi}{2N} + \frac{\pi}{2}$$

$$= \frac{K\pi}{2} + \frac{\pi}{4} + \frac{\pi}{2}$$

$$\theta_K = \frac{K\pi}{2} + \frac{3\pi}{4} \quad K = 0, 1, 2, 3$$

cross check

$$H_1(j\omega) = \frac{1}{j\omega + 1}$$

$$|H_1(j\omega)| = \frac{1}{\sqrt{\omega^2 + 1}}$$

$\omega = 1$

$$|H_1(j\omega)|_{\omega=1} = \frac{1}{\sqrt{2}}$$

(i) $K = 0$ $s_0 = 1 \angle \theta_0$

$$\theta_0 = 0 \cdot \frac{\pi}{2} + \frac{3\pi}{4} = \frac{3\pi}{4}$$

$$s_0 = 1 \angle \frac{3\pi}{4} = \cos 135^\circ + j \sin 135^\circ$$

$$s_0 = -\frac{1}{\sqrt{2}} + j \frac{1}{\sqrt{2}}$$

(ii) $K = 1$ $s_1 = 1 \angle \theta_1 = 1 \angle \left[\frac{\pi}{2} + \frac{3\pi}{4} \right] = 1 \angle \frac{5\pi}{4}$

$$\frac{5\pi}{4} = 225^\circ$$

$$= \cos \frac{5\pi}{4} + j \sin \frac{5\pi}{4}$$

$$s_1 = -\frac{1}{\sqrt{2}} - j \frac{1}{\sqrt{2}}$$

$$(ii) S_2 = 1 \angle \theta_2 \quad \theta_2 = \frac{2\pi}{2} + \frac{3\pi}{4} = \frac{7\pi}{4} = 315^\circ$$

$$S_2 = 1 \angle \frac{7\pi}{4} = \cos \frac{7\pi}{4} + j \sin \frac{7\pi}{4}$$

$$\boxed{S_2 = \frac{1}{\sqrt{2}} - j \frac{1}{\sqrt{2}}}$$

$$(iv) \underline{K=3}$$

$$S_3 = 1 \angle \theta_3$$

$$\theta_3 = \frac{3\pi}{2} + \frac{3\pi}{4} = \frac{9\pi}{4} = 405^\circ$$

$$S_3 = 1 \angle \frac{9\pi}{4} = \cos \frac{9\pi}{4} + j \sin \frac{9\pi}{4}$$

$$\boxed{S_3 = \frac{1}{\sqrt{2}} + j \frac{1}{\sqrt{2}}}$$

$$H_2(s) = \frac{1}{(s-s_0)(s-s_1)}$$

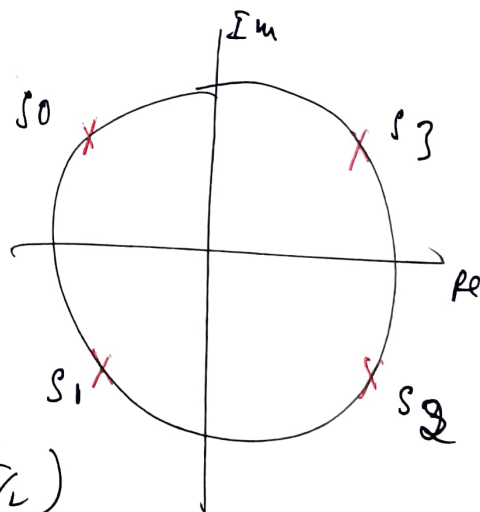
$$B_2(s) = (s-s_0)(s-s_1)$$

$$= \left(s + \frac{1}{\sqrt{2}} - j \frac{1}{\sqrt{2}}\right) \left(s + \frac{1}{\sqrt{2}} + j \frac{1}{\sqrt{2}}\right)$$

$$= s^2 + \frac{s}{\sqrt{2}} + j \frac{s}{\sqrt{2}} + \frac{s}{\sqrt{2}} + \frac{1}{2} + j \frac{1}{2} - j \frac{s}{\sqrt{2}} - j \frac{1}{2} + \frac{1}{2}$$

$$= s^2 + \frac{2s}{\sqrt{2}} + 1$$

$$\boxed{B_2(s) = s^2 + \sqrt{2}s + 1}$$



$$H_2(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$$

(ii) N = 3 [Third order filter]

wkt $s_k = 1 \angle \theta_k$

$$\theta_k = \frac{k\pi}{N} + \frac{\pi}{2N} + \frac{\pi}{2}; \quad k = 0, 1, \dots, 2N-1$$

N = 3 we get

$$\theta_k = \frac{k\pi}{3} + \frac{\pi}{6} + \frac{\pi}{2}; \quad k = 0, 1, 2, 3, 4, 5$$

$$= \frac{k\pi}{3} + \frac{2\pi}{3}$$

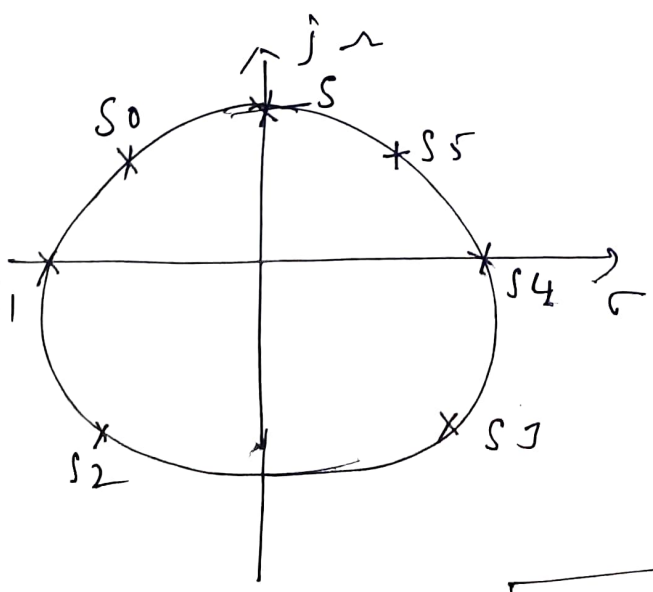
$$s_0 = 1 \angle \frac{2\pi}{3} = -0.5 + j0.866 \quad (\sqrt{3}/2)$$

$$s_1 = 1 \angle \pi = -1$$

$$s_2 = 1 \angle \frac{4\pi}{3} = -0.5 - j0.866$$

$$s_3 = 1 \angle \frac{4\pi}{3} = -0.5 - j0.866$$

$$s_4 = 1 \angle 2\pi = 1, \quad s_5 = 1 \angle \frac{7\pi}{3} = 0.5 + j0.866$$



$$H_3(s) = \frac{1}{(s-s_0)(s-s_1)(s-s_2)}$$

$$B_3(s) = (s + 0.5 - j0.866)(s + 1)(s + 0.5 + j0.866)$$

$$= (s+1) [(s+0.5)^2 + (0.866)^2]$$

$$B_3(s) = (s+1)(s^2 + s + 1)$$

Frequency Transformation or Spectral Transformation or Analog to Analog Transformation

* A LPF is used as a prototype or standard for all types of filter design

* Let $H(s)$ be the transfer fun of normalised LPF

* Let $H'(s)$ is a transfer fun of a new filter which is obtained by replacing s by $\frac{s}{\Omega u}$

$$\text{i.e., } H'(s) = H(s) \Big|_{s \rightarrow \left(\frac{s}{\Omega u}\right)}$$

$$= H\left(\frac{s}{\Omega u}\right)$$

$$\text{Let } s = j\Omega$$

$$H'(j\Omega) = H\left(\frac{j\Omega}{\Omega u}\right)$$

$$\text{Let } \Omega = \Omega u$$

$$H'(j\Omega u) = H(j)$$

* The above eqn means that the freq response (10) of a new filter evaluated at $\Omega = \Omega_u$ is equal to the value of the normalized filter at $\Omega = 1 \text{ rad/sec}$

* The cut off freq has moved from $\Omega_c = 1 \text{ rad/sec}$ to $\Omega_u \text{ rad/sec}$

* Similarly transformation can be defined for normalized LPF to HPF, BPF & BSF as shown below

1) normalized Low pass to Low pass transformation

$$s \rightarrow \frac{s}{\Omega_u}$$

2) Low pass to high pass transformation

$$s \rightarrow \frac{\Omega_u}{s}$$

3) Lowpass to band pass transformation
[combination of LP & HP]

$$s \rightarrow \frac{s^2 + \Omega_u \Omega_l}{s(\Omega_u - \Omega_l)}$$

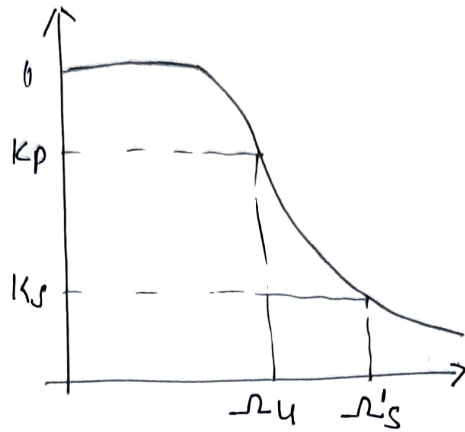
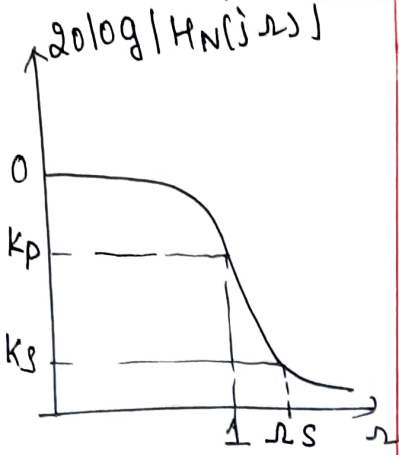
4) Lowpass to band stop transformation

$$s \rightarrow \frac{s(\Omega_u - \Omega_l)}{s^2 + \Omega_u \Omega_l}$$

Prototype filter response

Transformed filter response

Actual filter design eqn

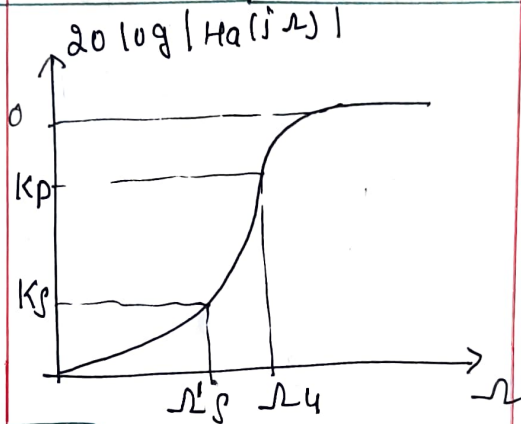


$$\Omega_s = \frac{\Omega'_s}{\Omega_u}$$

LPF $H_n(s)$ $\boxed{s \rightarrow \frac{s}{\Omega_u}}$

LPF $H_a(s)$

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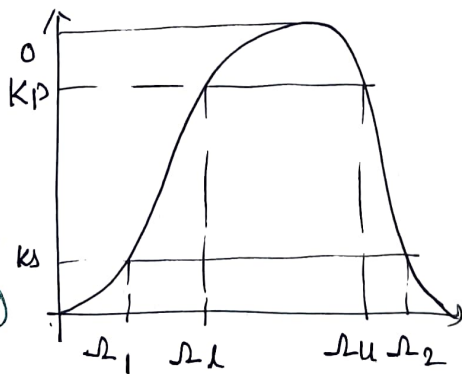
$$\bar{\Omega}_s = \frac{\Omega_u}{\Omega'_s}$$

$\boxed{s \rightarrow \frac{\Omega_u}{s}}$

HPF

— " —

$$s \rightarrow \frac{s^2 + \Omega_u \Omega_d}{s(\Omega_u - \Omega_d)}$$



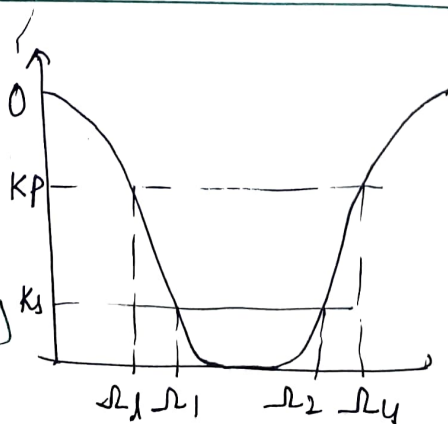
$$\Omega_s = \min\{|\Omega_1|, |\Omega_2|\}$$

$$A = \frac{-\Omega_1^2 + \Omega_1 \Omega_u}{\Omega_1 (\Omega_u - \Omega_d)}$$

$$B = \frac{\Omega_2^2 - \Omega_2 \Omega_u}{\Omega_2 (\Omega_u - \Omega_d)}$$

— " —

$$s \rightarrow \frac{s(\Omega_u - \Omega_d)}{s^2 + \Omega_u \Omega_d}$$



$$\Omega_s = \min\{|\Omega_1|, |\Omega_2|\}$$

$$A = \frac{\Omega_1 (\Omega_u - \Omega_d)}{-\Omega_1^2 + \Omega_1 \Omega_u}$$

$$B = \frac{\Omega_2 (\Omega_u - \Omega_d)}{\Omega_2^2 - \Omega_u \Omega_d}$$

Design of

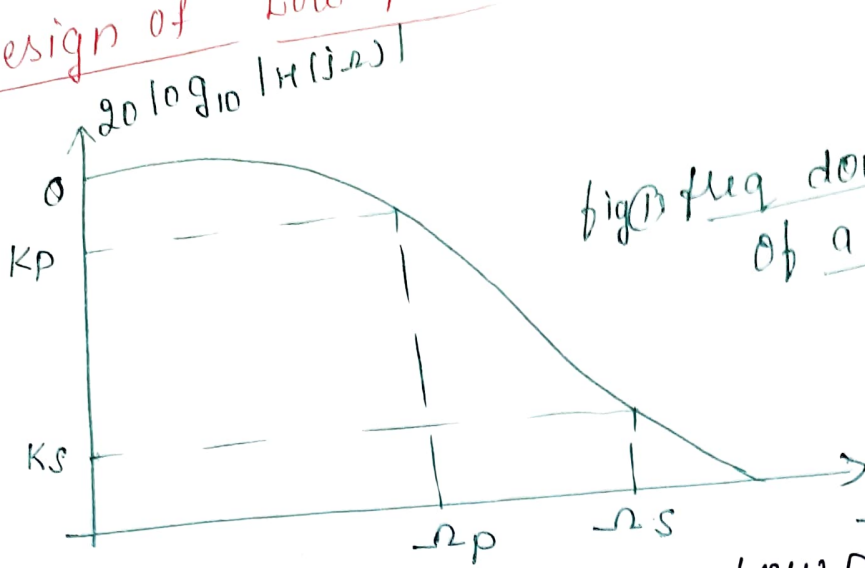


fig 1) freq domain specification of a LP Butterworth filter

We are required to design a low pass Butterworth filter given the following specifications.

pass band gain - K_p , stop band gain - K_s
 pass band freq - Ω_p & stop band freq - Ω_s

$$K_p \leq 20 \log |H(j\Omega)| \leq 0 \quad \text{for all } \Omega \leq \Omega_p$$

$$20 \log |H(j\Omega)| \leq K_s \quad \text{for all } \Omega \geq \Omega_s$$

The magnitude response of a LP Butterworth filter is given by

$$|H(j\Omega)| = \frac{1}{\left[1 + \left(\frac{\Omega}{\Omega_c}\right)^{2N}\right]^{1/2}} \quad \text{--- (1)}$$

to find N:-

taking $20 \log_{10}$ on both sides we get

$$20 \log_{10} |H(j\Omega)| = -20 \log_{10} \left[1 + \left(\frac{\Omega}{\Omega_c}\right)^{2N}\right]^{1/2}$$

$$= -10 \log_{10} \left[1 + \left(\frac{\Omega}{\Omega_c}\right)^{2N}\right] \quad \text{--- (2)}$$

from fig (1) we find that
at $\omega = \omega_p$, $20 \log_{10} |H(j\omega)| = K_p$

\therefore eq (2) becomes

$$K_p = -10 \log_{10} \left[1 + \left(\frac{\omega_p}{\omega_c} \right)^{2N} \right]$$

$$\therefore \left[\left(\frac{\omega_p}{\omega_c} \right)^{2N} = 10^{-\frac{K_p}{10}} - 1 \right] \rightarrow (3)$$

||| at $\omega = \omega_s$, $20 \log_{10} |H(j\omega)| = K_s$

eq (2)

$$K_s = -10 \log_{10} \left[1 + \left(\frac{\omega_s}{\omega_c} \right)^{2N} \right]$$

$$\left[\left(\frac{\omega_s}{\omega_c} \right)^{2N} = 10^{-\frac{K_s}{10}} - 1 \right] \rightarrow (4)$$

dividing eq (3) by (4)

$$\left(\frac{\omega_p}{\omega_s} \right)^{2N} = \frac{10^{-K_p/10} - 1}{10^{-K_s/10} - 1}$$

Taking logarithm on both side

$$2N \log_{10} \left(\frac{\omega_p}{\omega_s} \right) = \log_{10} \left[\frac{(10^{-K_p/10} - 1)}{(10^{-K_s/10} - 1)} \right]$$

$$N = \frac{\log_{10} \left[\left(10^{-K_P/10} - 1 \right) / \left(10^{-K_S/10} - 1 \right) \right]}{2 \log_{10} \left(\frac{\Omega_P}{\Omega_S} \right)}$$

If N is an integer, we take that value otherwise N is rounded off to next larger integer \rightarrow (5)

Once the order of 'N' is decided, the procedure for finding the cut off freq ω_c is as follows:

1) If we desire to meet the passband requirement exactly - the cut off freq is selected from eq (3)

$$\left(\frac{\Omega_P}{\Omega_C} \right)^{2N} = 10^{-K_P/10} - 1$$

$$\therefore \Omega_{C1} = \frac{\Omega_P}{\left[10^{-K_P/10} - 1 \right]^{1/2N}} \rightarrow (6)$$

2) If we wish to meet our requirements at stopband then

$$\Omega_{C2} = \frac{\Omega_S}{\left[10^{-K_S/10} - 1 \right]^{1/2N}} \rightarrow (7)$$

3) If one desires to do better in both the bands select Ω_C - cut off freq as the arithmetic mean of the 2 cut off freq's found above

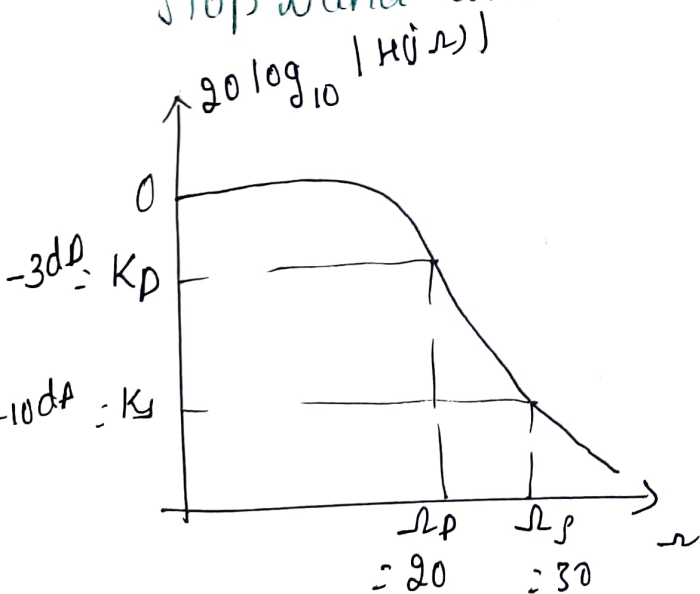
$$\Omega_C = \text{(avg) mean} (\Omega_{C1}, \Omega_{C2})$$

Design steps

- 1) From the given specifications, find the order of the filter 'N'
- 2) Find the cut off freq ω_c to meet the PB/SB both requirements
- 3) to determine BUTTERWORTH polynomials & Transfer function for a normalized LP filter.
- 4) to find transfer fun of the filter given

$$H_a(s) = H_N(s) \Big|_{s \rightarrow \text{(depending on filter)}}$$

① Design an Analog Butterworth filter that has a -3dB pass band attenuation at a freq of 20 rad/sec and at least -10dB stop band attenuation at 30 rad/sec



Given:

$$K_p = -3 \text{ dB} \quad \omega_p = 20 \text{ rad/sec}$$

$$K_s = -10 \text{ dB} \quad \omega_s = 30 \text{ rad/sec}$$

$$\left(\frac{\omega_p}{\omega_s} \right) = \frac{2}{3}$$

$$1) \text{ WKT } N = \frac{\log_{10} \left[\left(10^{-K_p/10} - 1 \right) \left(10^{-K_s/10} - 1 \right) \right]}{2 \log_{10} \left(\frac{\omega_p}{\omega_s} \right)}$$

$$= \log_{10} \left[\left(10^{-\left(\frac{-3}{10}\right)} - 1 \right) \left| \left(10^{-\left(\frac{-10}{10}\right)} - 1 \right) \right. \right]$$

$$2 \log_{10} \left(\frac{20}{30} \right)$$

$$N = 2.7153 = \text{Rounding off to the next larger integer}$$

$$\boxed{N=3}$$

(i) Find normalised Butterworth polynomial & transfer funⁿ derive it

$$B_3(s) = (s+1)(s^2+s+1)$$

$$\text{TF: } H_3(s) = \frac{1}{(s+1)(s^2+s+1)} = \frac{1}{B_3(s)}$$

(ii) Let us determine the cut off the freq ω_c to meet the P.B

$$\omega_c = \frac{\omega_p}{\left[10^{-K_p/10} - 1 \right]^{1/2N}} = \frac{20}{\left[10^{-\left(\frac{-3}{10}\right)} - 1 \right]^{1/2 \times 3}}$$

$$\boxed{\omega_c = 20 \text{ rad/sec}}$$

to find T.F for a cut off freq 20 rad/sec

$$(iii) H_a(s) = H_3(s) \Big|_{s \rightarrow \frac{s}{\omega_c}} = H_3(s) \Big|_{s \rightarrow \frac{s}{20}}$$

$$H_a(s) = \frac{1}{\left[\frac{s}{20} + 1\right] \left[\left(\frac{s}{20}\right)^2 + \left(\frac{s}{20}\right) + 1\right]}$$

$$= \frac{1}{\frac{s+20}{20} \left[\frac{s^2}{400} + \frac{20s}{400} + \frac{400}{400}\right]}$$

$$= \frac{20 \times 400}{(s+20)(s^2 + 20s + 400)}$$

$$H_a(s) = \frac{8 \times 10^3}{(s+20)(s^2 + 20s + 400)}$$

Verification of the design

Let $s = j\omega$ in $H_a(s)$

$$H_a(j\omega) = \frac{8 \times 10^3}{(j\omega + 20)(-\omega^2 + j20\omega + 400)}$$

$$|H_a(j\omega)| = \frac{8 \times 10^3}{\sqrt{\omega^2 + 400} \sqrt{(400 - \omega^2)^2 + (20\omega)^2}}$$

$$20 \log_{10} |H_a(s=j\omega)|_{\omega=20}$$

$$= 20 \log_{10} \left[\frac{8 \times 10^3}{\sqrt{20^2 + 400} \sqrt{(400 - 400)^2 + (20 \times 20)^2}} \right]$$

$$= -3.01 \text{ dB}$$

$$\approx -3 \text{ dB}$$

$$20 \log_{10} |H_a(s=j\omega)|_{\omega=30} = 20 \log_{10} (\quad)$$

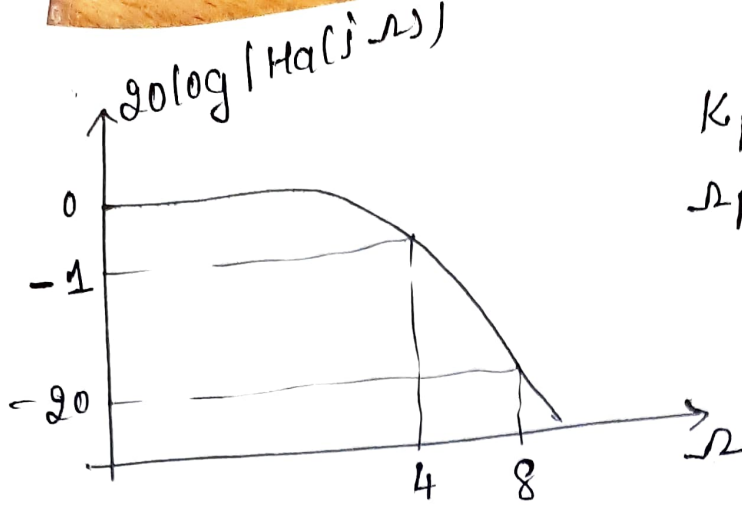
$$= -10.93 \text{ dB}$$

② A Butterworth LPF has to meet the following specifications

(a) passband gain $K_p = -1 \text{ dB}$ at $\omega_p = 4 \text{ rad/sec}$

(b) stopband gain is greater than or equal to 20 dB at $\omega_s = 8 \text{ rad/sec}$

determine the T.F. $H_a(s)$ of the Butterworth filter to meet the above specifications.



$$K_p = -10 \text{ dB} \quad K_s \geq -20 \text{ dB}$$

$$\Delta p = 4 \text{ rad/sec} \quad \Delta s = 8 \text{ rad/sec}$$

(i) $N = 4 \cdot 289 \approx 5$

ii) to find 5th order normalised butterworth polynomial $B_5(s)$

Let $S_K = 1 \angle \theta_K$

$$\theta_K = \frac{\pi K}{N} + \frac{\pi}{2N} + \frac{\pi}{2} =$$

$$N=5$$

$$S_0 = 1 \angle \frac{6\pi}{10} = -0.309 + j0.951$$

$$S_1 = 1 \angle \frac{8\pi}{10} = -0.809 + j0.588$$

$$S_2 = 1 \angle \pi = -1$$

$$S_3 = 1 \angle \frac{12\pi}{10} = -0.809 - j0.588$$

$$S_4 = 1 \angle \frac{14\pi}{10} = -0.309 - j0.951$$

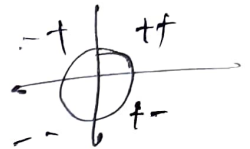
$$S_5 = 1 \angle \frac{16\pi}{10} = 0.309 - j0.951$$

$$S_6 = 1 \angle \frac{18\pi}{10} = 0.809 - j0.588$$

$$S_7 = 1 \angle 2\pi = 1$$

$$S_8 = 1 \angle \frac{22\pi}{10} = 0.809 + j0.588$$

$$S_9 = 1 \angle \frac{24\pi}{10} = 0.309 + j0.951$$



$$H_S(s) = \frac{1}{(s-s_0)(s-s_1)(s-s_2)(s-s_3)(s-s_4)} \quad \text{⑮}$$

$$= \frac{1}{(s+1)(s^2+0.6180s+1)(s^2+1.6180s+1)}$$

$$= \frac{1}{s^5 + 3.236s^4 + 5.236s^3 + 5.236s^2 + 3.236s + 1}$$

(iii) to find cut off freq

$$\omega_c = \frac{\omega_p}{(10^{-Kp/10} - 1)^{1/2N}}$$

$$\Rightarrow \frac{4.5787}{4.5787}$$

$$\omega_c = 5.05$$

$$\omega_c = 4.815$$

(iv) to find H_a(s)

$$H_a(s) = H_S(s) \Big|_{s \rightarrow \frac{s}{\omega_c}}$$

$$= \frac{1}{\left(\frac{s}{4.5787}\right)^5 + 3.236 \left(\frac{s}{4.5787}\right)^4}$$

$$+ 5.236 \left(\frac{s}{4.5787}\right)^3 + 5.236 \left(\frac{s}{4.5787}\right)^2$$

$$+ 3.236 \left(\frac{s}{4.5787}\right) + 1$$

$$2012.4$$

$$= \frac{1}{s^5 + 14.82s^4 + 109.8s^3 + 502.6s^2 + 1422.3s}$$

$$+ 2012.411$$

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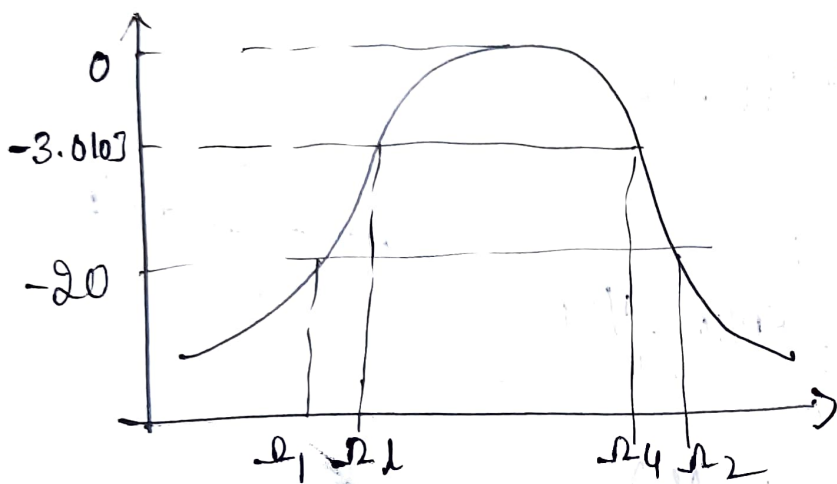
1

③ Design a analog band pass filter to meet the foll freq. domain specifications

(a) -3.0103 dB upper & lower cut off freq of 50 Hz & 20 kHz

(b) a stopband attenuation of at least 20 dB at 20 Hz & 45 kHz

& (c) a monotonic freq response.



NOTE: Ω values must be in radians

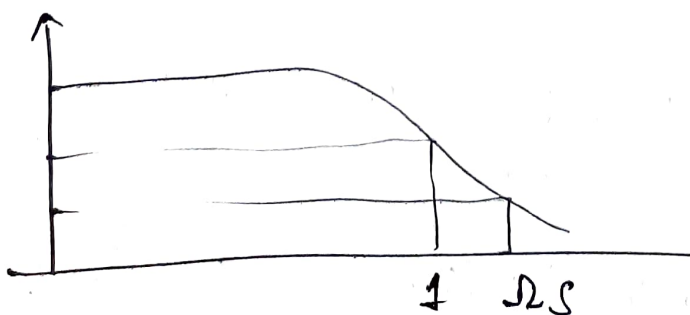
$$\Omega_1 = 2\pi \times 20 = 125.663 \text{ rad/sec}$$

$$\Omega_2 = 45 \times 10^3 \times 2\pi = 2.827 \times 10^5 \text{ rad/sec}$$

$$\Omega_3 = 2\pi \times 20 \times 10^3 = 1.257 \times 10^5 \text{ rad/sec}$$

$$\Omega_4 = 2\pi \times 50 = 314.159 \text{ rad/sec}$$

↳ let us use the backward design



$$A = \frac{-\Omega_1^2 + \Omega_2 \Omega_4}{\Omega_1 (\Omega_4 - \Omega_2)} = 2.51$$

$$B = \frac{\Omega_2^2 - \Omega_1 \Omega_4}{\Omega_2 (\Omega_4 - \Omega_1)} = 2.25$$

$$\Omega_s = \min [|A|, |B|] = \underline{\underline{2.25}}$$

∴ Order of normalised LP Butterworth filter

$$N = \frac{\log \left[\frac{[10^{-\frac{K_p}{10}} - 1]}{[10^{-\frac{K_s}{10}} - 1]} \right]}{2 \log \left(\frac{1}{\Omega_s} \right)}$$

$$= 2.83$$

$$N \approx \underline{\underline{3}}$$

$$(ii) H_3(s) = \frac{1}{s^3 + 2s^2 + 2s + 1}$$

$$(iv) H_a(s) = H_3(s) \Big|_s \rightarrow \frac{s^2 + \Omega_u \Omega_l}{s(\Omega_u - \Omega_l)} = \frac{s^2 + 3.949 \times 10^7}{(1.2538 \times 10^5)s}$$

$$\Omega_u \Omega_l = 3.949 \times 10^7$$

$$\Omega_u - \Omega_l = 1.2538 \times 10^5$$

$$1.9695 \times 10^{15} s^3$$

$$H_a(s) = \frac{s^6 + 2.51 \times 10^5 s^5 + 3.154 \times 10^5 s^4 + 1.989 \times 10^{15} s^3 + 1.2453 \times 10^{18} s^2 + 3.9073 \times 10^{20} s + 6.1529 \times 10^{22}}$$

(4) Let $H(s) = \frac{1}{s^2 + s + 1}$ represent the T.F. of a LPF with pass band freq of 1 rad/sec
 use freq transformations to find the T.F.s of the foll analog filter.

- (a) A LPF with a pass band of 10 rad/sec
- (b) A HPF with a cut off freq of 10 rad/sec
- (c) A BPF with a pass band of 10 rad/sec & center freq of 100 rad/sec
- (d) A band stop filter with a stopband of 2 rad/sec & a center freq of 10 rad/sec

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

(a) $s \rightarrow \frac{s}{10}$

$$H_a(s) = H(s) / s \rightarrow \frac{s}{10}$$

$$H_a(s) = \frac{100}{s^2 + 10s + 100}$$

(b) $s \rightarrow \frac{10}{s}$

$$H_a(s) = H(s) / s \rightarrow \frac{10}{s}$$

$$H_a(s) = \frac{s^2}{s^2 + 10s + 100}$$

(c) $s \rightarrow$

LPF \rightarrow BPF

$$s \rightarrow \frac{s^2 + \Omega_u \Omega_l}{s(\Omega_u - \Omega_l)}$$

center freq of bandpass

$$\Omega_0 = \sqrt{\Omega_u \Omega_l}$$

$$B_0 = \text{width of passband} = \Omega_u - \Omega_l$$

$$s \rightarrow \frac{s^2 + \Omega_0^2}{s B_0} = \frac{s^2 + (100)^2}{10 \times s}$$

$$= \frac{s^2 + 10^4}{10s}$$

$$H(s) = \frac{100 s^2}{s^4 + 10s^3 + 20100s^2 + 10^5 s + 10^8}$$

LPF \rightarrow BSF

$$s \rightarrow \frac{s(\Omega_u - \Omega_l)}{s^2 + \Omega_u \Omega_l} = \frac{s B_0}{s^2 + \Omega_0^2}$$

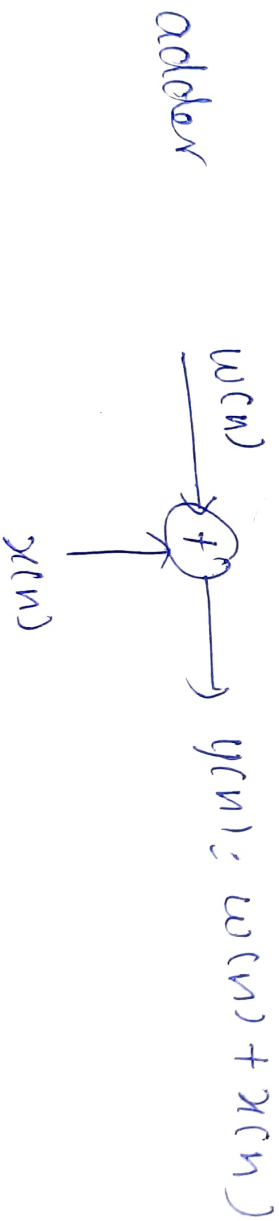
$$s \rightarrow \frac{2s}{s^2 + 100}$$

$$H(s) = \frac{(s^2 + 100)^2}{s^4 + 2s^3 + 204s^2 + 200s + 10^4}$$

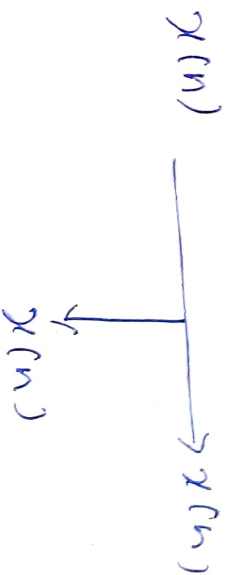
* A Digital Filters can be realized using various configurations or structures.

* A structure can be represented using either block diagram or signal flow graph

* The block diagram representation of a digital filter consists unit delays, multipliers, adders
pick-off nodes are shown below:



Pick-off nodes



Basic IIR Filter Structures

~~Consider~~ Causal IIR filter is characterized by

$$H(z) = \frac{Y(z)}{X(z)} = \frac{\sum_{k=0}^M b_k z^{-k}}{1 + \sum_{k=1}^N a_k z^{-k}} \rightarrow \textcircled{1}$$

$$\frac{Y(z)}{X(z)} = \frac{b_0 + b_1 z^{-1} + b_2 z^{-2} + \dots + b_M z^{-M}}{1 + a_1 z^{-1} + a_2 z^{-2} + a_3 z^{-3} + \dots + a_N z^{-N}}$$

Cross multiply, & IZT $\rightarrow \textcircled{2}$

$$y(n) = - \sum_{k=1}^N a_k z^{-k} + \sum_{k=0}^M b_k z^{-k} \rightarrow \textcircled{3}$$

1. Direct-form structure

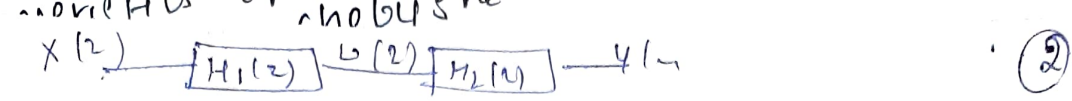
from eq $\textcircled{1}$

$$H(z) = \frac{\sum_{k=0}^M b_k z^{-k}}{1 + \sum_{k=1}^N a_k z^{-k}} \rightarrow \textcircled{4}$$

we can represent $H(z)$ as cascade of 2 systems ~~for~~ with system funⁿ $H_1(z)$ & $H_2(z)$

$$H(z) = H_1(z) \cdot H_2(z) \rightarrow \textcircled{2}$$

- we represent IIR filter in several form
- (i) direct form I
 - (ii) cascade
 - (iii) parallel
 - (iv) lattice
 - (v) lattice ladder



where

$$H_1(z) = \frac{W(z)}{X(z)} = \sum_{k=0}^M b_k z^{-k} \rightarrow (3)$$

$$H_2(z) = \frac{Y(z)}{W(z)} = \frac{1}{1 + \sum_{k=1}^N a_k z^{-k}} \rightarrow (4)$$

~~ex~~ consider a third order (N=3) filter

$$H(z) = \frac{Y(z)}{X(z)} = \frac{b_0 + b_1 z^{-1} + b_2 z^{-2} + b_3 z^{-3}}{1 + a_1 z^{-1} + a_2 z^{-2} + a_3 z^{-3}} \rightarrow (5)$$

eq (5) can be written as

$$H(z) = H_1(z) \cdot H_2(z)$$

where

$$H_1(z) = \frac{W(z)}{X(z)} = b_0 + b_1 z^{-1} + b_2 z^{-2} + b_3 z^{-3} \rightarrow (6)$$

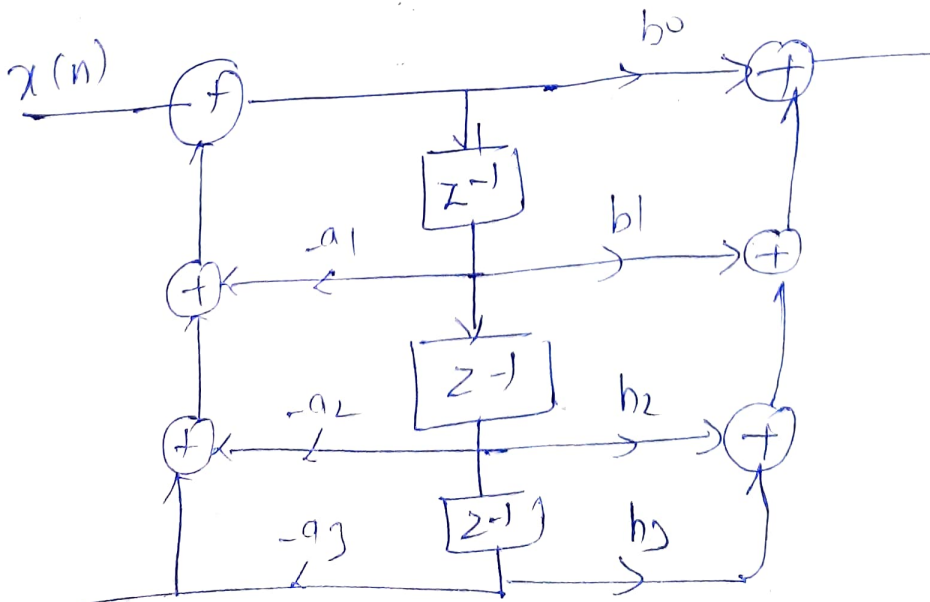
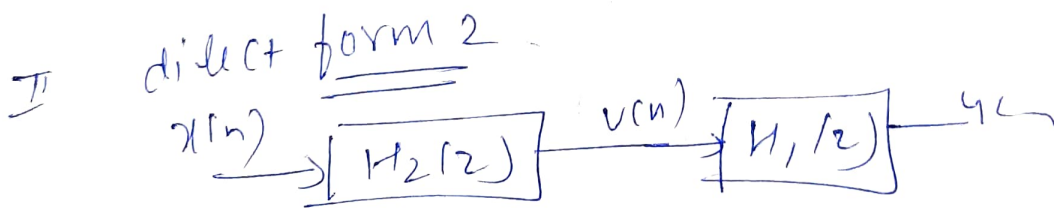
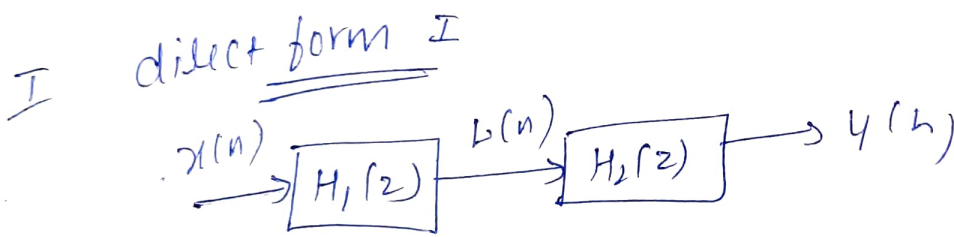
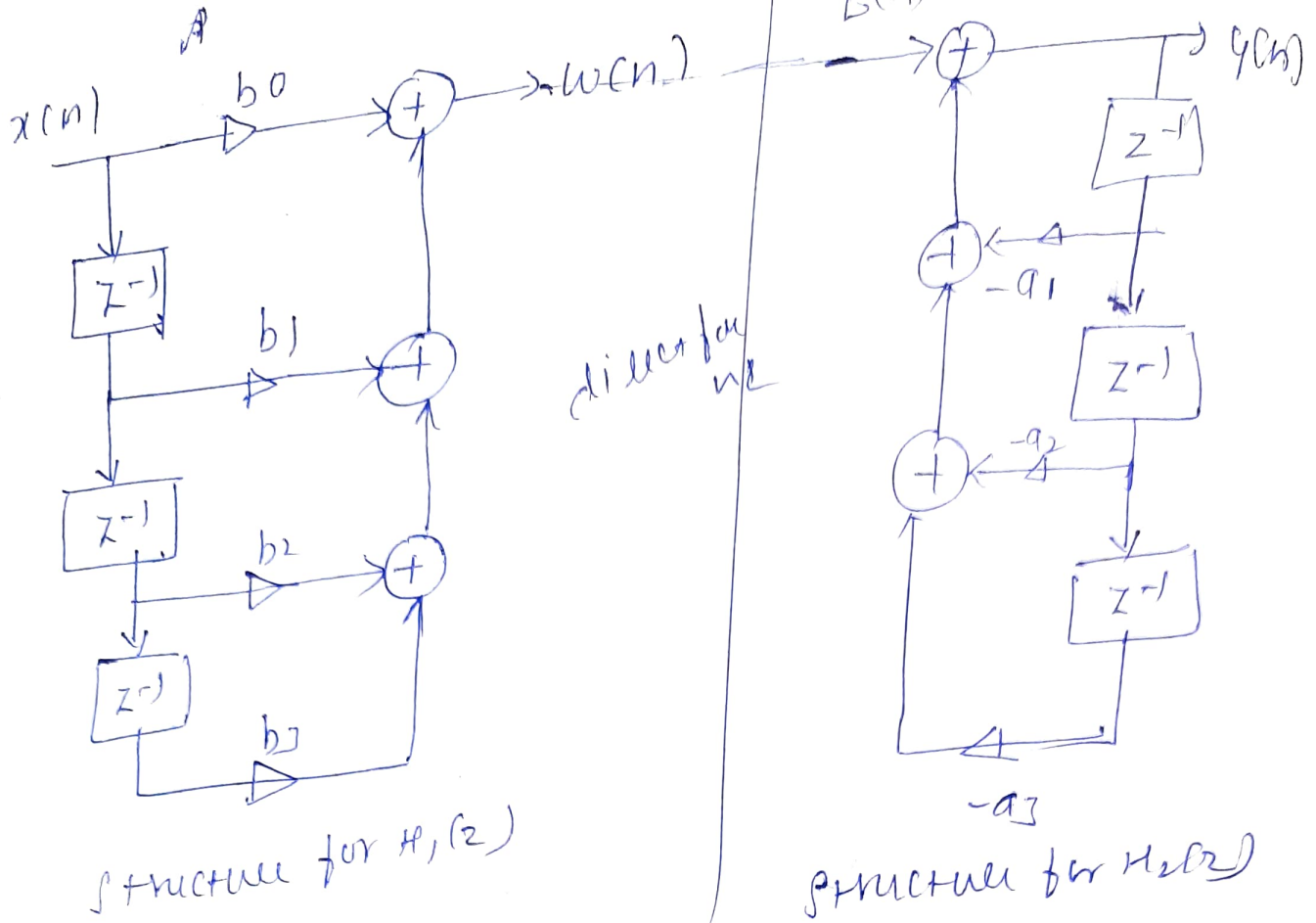
$$H_2(z) = \frac{Y(z)}{W(z)} = \frac{1}{1 + a_1 z^{-1} + a_2 z^{-2} + a_3 z^{-3}} \rightarrow (7)$$

taking inverse ZT on eq (6)

$$W(n) = b_0 x(n) + b_1 x(n-1) + b_2 x(n-2) + b_3 x(n-3) \rightarrow (8)$$

taking inverse ZT on eq (7)

$$Y(n) = W(n) - a_1 y(n-1) - a_2 y(n-2) - a_3 y(n-3) \rightarrow (9)$$



I obtain direct form - I & II realizations (3)
 for a digital IIR filter described by the system fun

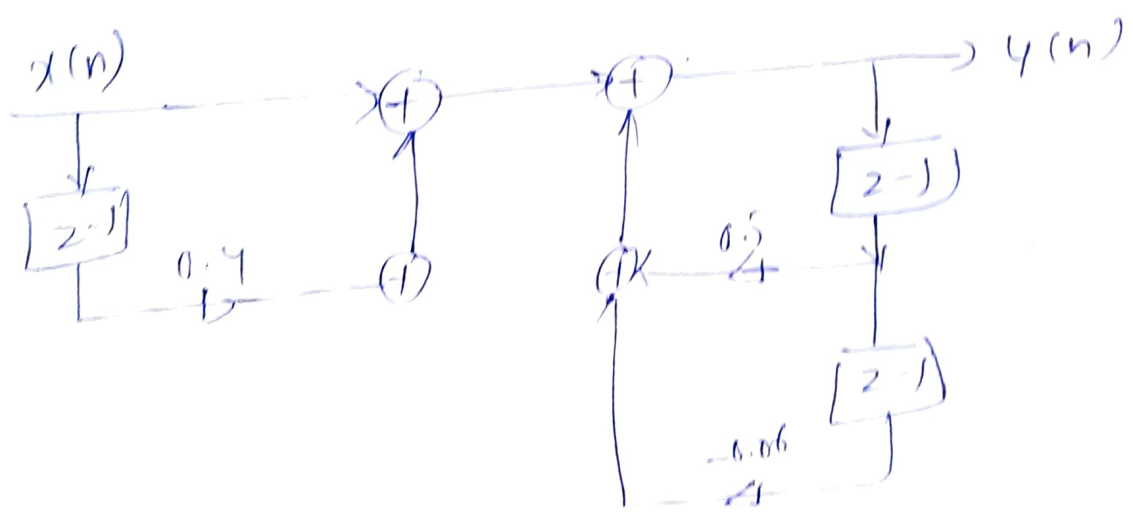
$$H(z) = \frac{1 + 0.4z^{-1}}{1 - 0.5z^{-1} + 0.06z^{-2}}$$

$$\frac{Y(z)}{X(z)} = \frac{1 + 0.4z^{-1}}{1 - 0.5z^{-1} + 0.06z^{-2}}$$

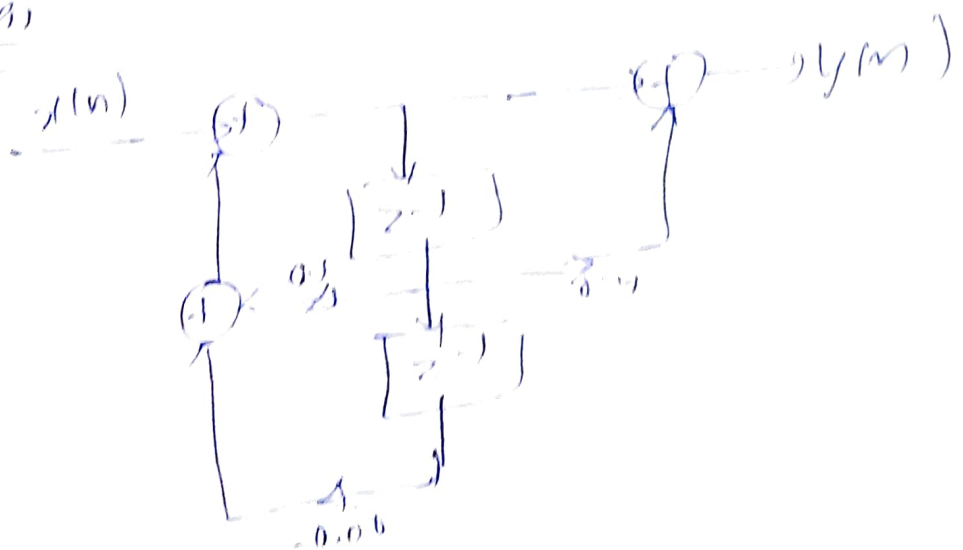
$$Y(z) - 0.5z^{-1}Y(z) + 0.06z^{-2}Y(z) = X(z) + 0.4z^{-1}X(z)$$

IIR

$$y(n) = x(n) + 0.4x(n-1) + 0.5y(n-1) - 0.06y(n-2)$$



direct form (2)



$$(2) \quad H(z) = \frac{z^{-1} - 3z^{-2}}{(10 - z^{-1})(1 + 0.5z^{-1} + 0.5z^{-2})}$$

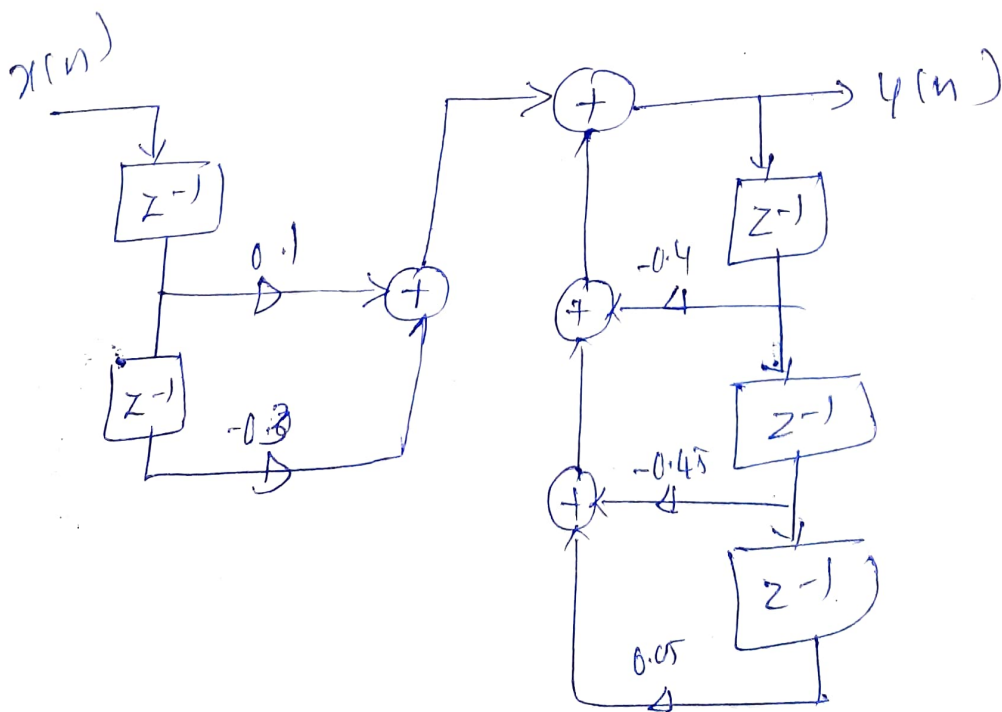
$$H(z) = \frac{z^{-1} - 3z^{-2}}{10 + 4z^{-1} + 4.5z^{-2} + 0.5z^{-3}}$$

$$\frac{Y(z)}{X(z)} = \frac{0.1z^{-1} - 0.3z^{-2}}{1 + 0.4z^{-1} + 0.45z^{-2} - 0.05z^{-3}}$$

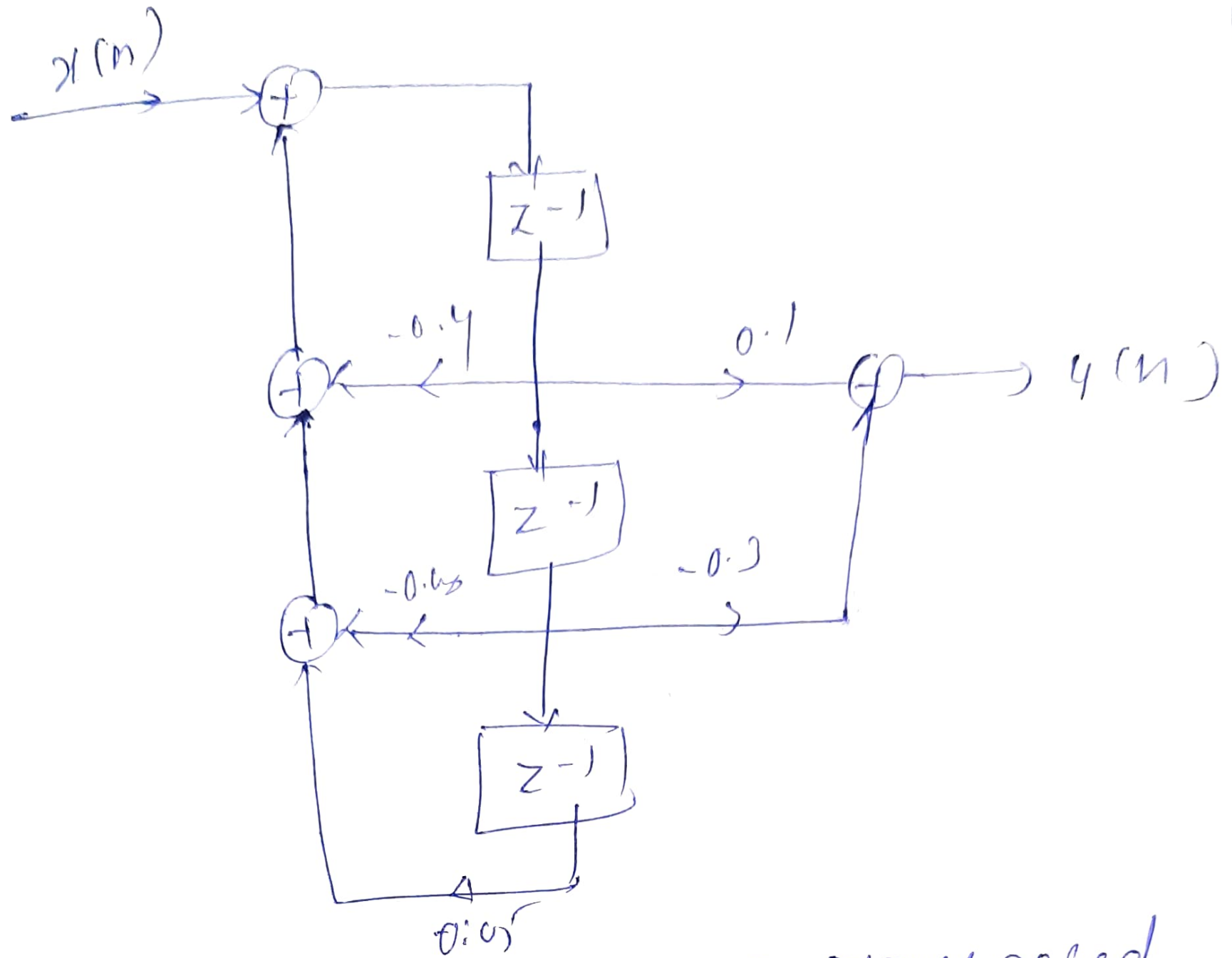
$$Y(z) = 0.1z^{-1}X(z) - 0.3z^{-2}X(z) - 0.4z^{-1}Y(z) - 0.45z^{-2}Y(z) + 0.05z^{-3}Y(z)$$

∫ z T

$$y(n) = 0.1x(n-1) - 0.3x(n-2) - 0.4y(n-1) - 0.45y(n-2) + 0.05y(n-3)$$



11



Transposed

f2)

* To simulate an analog filter, the digital filter $H(z)$ must be always be used in A/D - $H(z)$ - D/A structure as shown in above fig

* different techniques are used for designing $H(z)$ which are listed below

(i) Bilinear transformation

(ii) Impulse invariant transform

(iii) Backward difference method

(iv) matched z-transform method

(i) Bilinear transformation

Let us consider the 1st order differential equation given by

$$a_1 y_a'(t) + a_0 y_a(t) = b_0 x_a(t) \rightarrow (1)$$

where a_1 , a_0 & b_0 are arbitrary constants
Taking LT on both sides with all initial condns equal to zero

$$a_1 s Y_a(s) + a_0 Y_a(s) = b_0 X_a(s)$$

$$Y_a(s) [a_1 s + a_0] = b_0 X_a(s)$$

$$H_a(s) \triangleq \frac{Y_a(s)}{X_a(s)} = \frac{b_0}{a_1 s + a_0} \rightarrow (2)$$

②

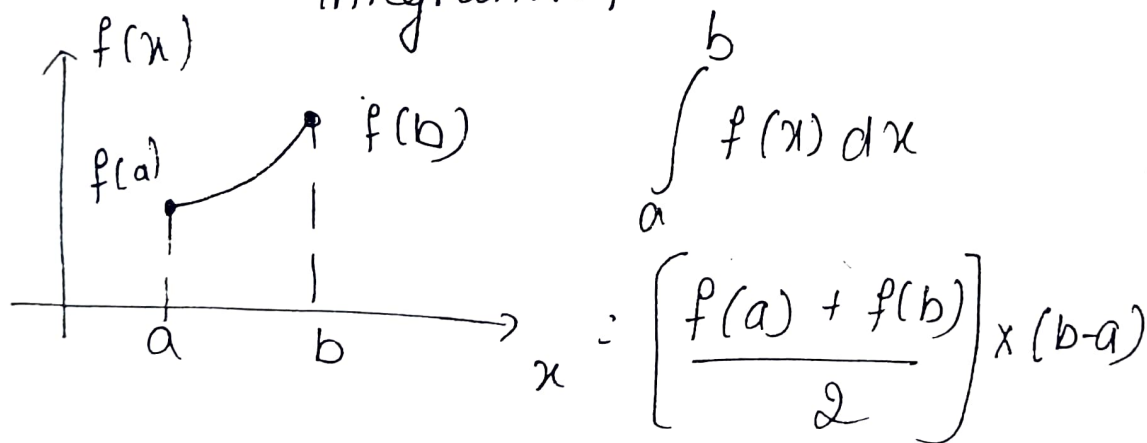
the fundamental theorem of integral calculus allows us to write

$$y_a(t) = \int_{t_0}^t y_a'(t) dt + y_a(t_0) \quad (3)$$

replace $t = nT$ & $t_0 = (n-1)T$ in eq (3) we get

$$y_a(nT) = \int_{(n-1)T}^{nT} y_a'(t) dt + y_a((n-1)T) \quad (4)$$

Recall - The trapezoidal rule of integration /



$$\int_{(n-1)T}^{nT} y_a'(t) dt = \left[y_a'(nT) + y_a'((n-1)T) \right]$$

$$\frac{(b-a)}{nT - (nT - T)} = \frac{T}{2} \left[y_a'(nT) + y_a'((n-1)T) \right]$$

eq (4) become

~~$y_a(t)$~~

$$y_a(nT) = \frac{T}{2} [y_a'(nT) + y_a'(n-1)T] + y_a(n-1)T$$

→ (5)

from eq (1) we have

$$y_a'(t) = -\frac{a_0}{a_1} y_a(t) + \frac{b_0}{a_1} x_a(t)$$

replacing t by nT & by $(n-1)T$

we get

$$y_a'(nT) = -\frac{a_0}{a_1} y_a(nT) + \frac{b_0}{a_1} x_a(nT) \rightarrow (6)$$

$$y_a'((n-1)T) = -\frac{a_0}{a_1} y_a((n-1)T) + \frac{b_0}{a_1} x_a((n-1)T)$$

→ (7)

sub eqs (6) & (7) in (5)

$$y_a(nT) = \frac{T}{2} \left[-\frac{a_0}{a_1} y_a(nT) + \frac{b_0}{a_1} x_a(nT) \right.$$

$$\left. -\frac{a_0}{a_1} y_a((n-1)T) + \frac{b_0}{a_1} x_a((n-1)T) \right]$$

$$+ y_a(n-1)T$$

denoting $y_a(nT)$ by $y(n)$

& $x_a(nT)$ by $x(n)$

$$y(n) = \frac{T}{2} \left\{ \begin{aligned} &-\frac{a_0}{a_1} y(n) + \frac{b_0}{a_1} x(n) \\ &-\frac{a_0}{a_1} y(n-1) + \frac{b_0}{a_1} x(n-1) \end{aligned} \right\} + y(n-1)$$

taking Z-transforms on both sides, we get

$$Y(z) = \frac{T}{2} \left\{ \begin{aligned} &-\frac{a_0}{a_1} Y(z) + \frac{b_0}{a_1} X(z) - \frac{a_0}{a_1} z^{-1} Y(z) \\ &+ \frac{b_0}{a_1} z^{-1} X(z) \end{aligned} \right\} + z^{-1} Y(z)$$

$$= -\frac{a_0}{a_1} \frac{T}{2} Y(z) + \frac{b_0}{a_1} \frac{T}{2} X(z) - \frac{a_0}{a_1} \frac{T}{2} z^{-1} Y(z) + \frac{b_0}{a_1} z^{-1} \frac{T}{2} X(z) + z^{-1} Y(z)$$

~~$$= X(z)$$~~

~~$$= \frac{b_0}{a_1} \frac{T}{2} X(z) \{1 + z^{-1}\}$$~~

~~$$= X(z) \left\{ \frac{b_0 T}{a_1} + \frac{b_0 T}{a_1} z^{-1} \right\} + Y(z) \left\{ \begin{aligned} &-\frac{a_0}{a_1} \frac{T}{2} - \frac{a_0}{a_1} \frac{T}{2} z^{-1} \\ &+ \frac{b_0}{a_1} z^{-1} \frac{T}{2} + z^{-1} \end{aligned} \right\}$$~~

~~$$= X(z) \left\{ \frac{b_0 T + b_0 T z^{-1}}{2 a_1} \right\} + Y(z)$$~~

$$\frac{-a_0 T - a_0 T z^{-1} + b_0 z^{-1}}{2 a_1}$$

(3)

$$Y(z) = \frac{b_0}{2a_1} T \{ X(z) + z^{-1} X(z) \}$$

$$- \frac{a_0}{2a_1} T \{ Y(z) + z^{-1} Y(z) \}$$

$$+ z^{-1} Y(z)$$

$$Y(z) \left[\frac{a_0}{2a_1} T (1 + z^{-1}) - z^{-1} + 1 \right]$$

$$= X(z) \left[\frac{b_0}{2a_1} T (1 + z^{-1}) \right]$$

$$\frac{Y(z)}{X(z)} = H(z) = \frac{\frac{b_0}{2a_1} T (1 + z^{-1})}{\frac{a_0}{2a_1} T (1 + z^{-1}) - z^{-1} + 1}$$

$\div (1+z^{-1})$ \div $\cancel{NR} \& \cancel{DR}$ of RHS by $(1+z^{-1})$

$$= \frac{\frac{b_0}{2a_1} T (1) \left(\frac{1+z^{-1}}{1+z^{-1}} \right)}{\frac{a_0}{2a_1} T + \frac{(1-z^{-1})}{(1+z^{-1})}}$$

$$= \frac{\frac{b_0}{2a_1} T + \frac{(1-z^{-1})}{(1+z^{-1})}}{\frac{a_0}{2a_1} T + \frac{(1-z^{-1})}{(1+z^{-1})}}$$

$$= \frac{\cancel{\frac{T}{2a_1}} b_0}{\cancel{\frac{T}{2a_1}} \left[a_0 + \left(\frac{1-z^{-1}}{1+z^{-1}} \right) \frac{2a_1}{T} \right]}$$

$$= \frac{b_0}{a_0 + \left\{ \frac{1-z^{-1}}{1+z^{-1}} \right\} \cdot \frac{2a_1}{T}}$$

$$= \frac{b_0}{a_0 + \left\{ \frac{1-z^{-1}}{1+z^{-1}} \right\} \cdot \frac{2a_1}{T}}$$

$$= \frac{b_0}{a_0 + \left\{ \frac{1-z^{-1}}{1+z^{-1}} \right\} \cdot \frac{2a_1}{T}}$$

$$H(z) = \frac{Y(z)}{X(z)} = \frac{b_0}{a_0 + a_1 x \frac{2}{T} \left\{ \frac{1-z^{-1}}{1+z^{-1}} \right\}} \rightarrow (8)$$

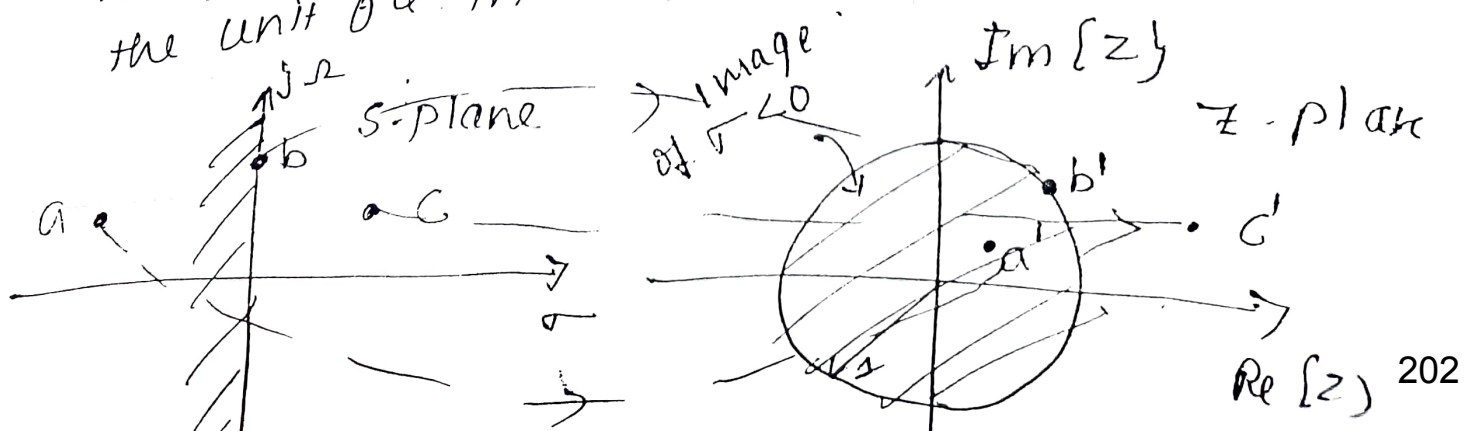
Comparing eq (2) & (8) we get

$$H(z) = H(s) \Big|_{s \rightarrow \frac{2}{T} \left[\frac{1-z^{-1}}{1+z^{-1}} \right]}$$

$$s = \frac{2}{T} \left[\frac{1-z^{-1}}{1+z^{-1}} \right] = \frac{2}{T} \left[\frac{z-1}{z+1} \right]$$

The above eqn is called as bilinear transformation.

- Properties of Bilinear transformation:
- ① bilinear transformation maps $j\omega$ axis of the s -plane onto the unit circle in z -plane. Also the LHS of the s -plane is mapped inside the unit circle while the RHS of the s -plane is mapped outside the unit circle in z -plane.

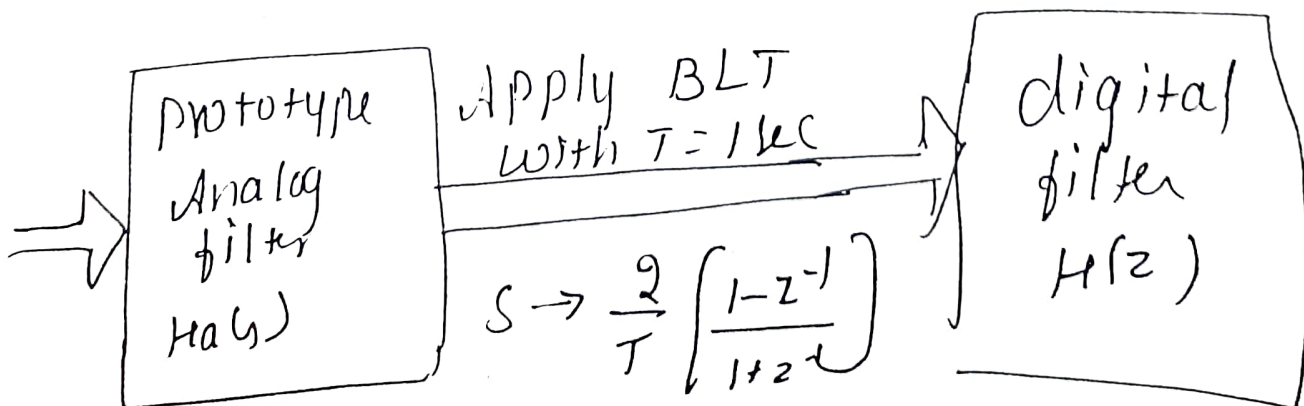
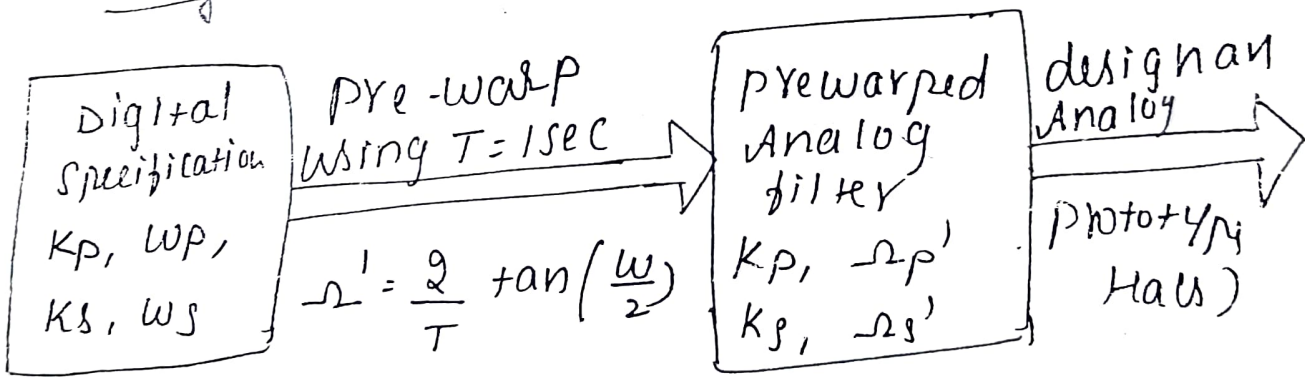


* Bilinear Transformation (BLT) does not preserve the phase response, when the phase response is mapped from s-plane to z-plane

* As a consequence of this, an analog filter having a linear phase in s-domain will no longer have a linear phase in z-domain when it is mapped using BLT.

↓ This is a serious limitation of BLT.

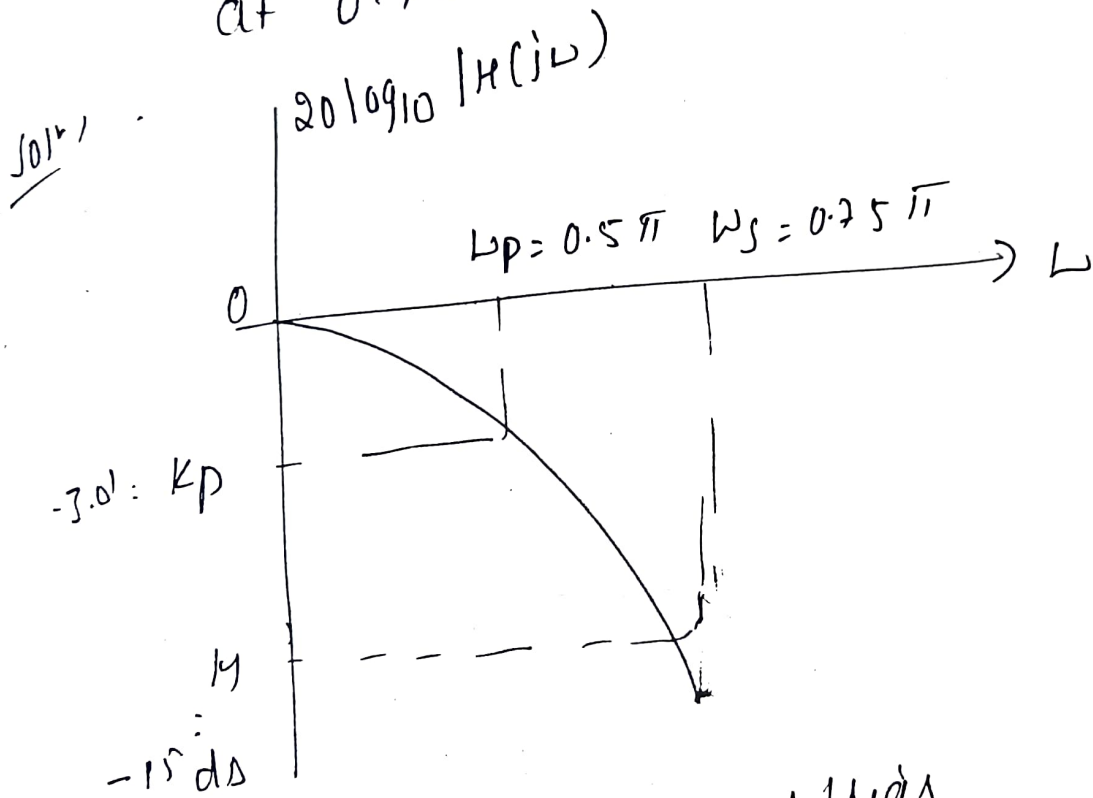
Design of digital filters using BLT



- (33)
- NOTE: (i) monotonic pass & stop bands \rightarrow Butterworth
 (ii) monotonic outside pass band \rightarrow Chebyshev

Q) Design & realize a LPF using B L T [difference eqn] for satisfying the following specifications

- (i) monotonic pass & stop band
- (ii) -3.01 dB cut off freq at 0.5π rad
- (iii) magnitude down at least 15 dB at 0.75π rad



Step 1 - prewarp the digital freqs

$$\omega'_p = \frac{2}{T} \tan\left(\frac{\omega_p}{2}\right) = \frac{2}{1} \tan\left(\frac{0.5\pi}{2}\right)$$

$$= 2 \text{ rad/sec}$$

$$\omega'_s = \frac{2}{T} \tan\left(\frac{\omega_s}{2}\right) = 2 \tan\left[\frac{0.75\pi}{2}\right]$$

2)

$$N = 1.941$$

$$N = 2$$

3)

$$H_2(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$$

4)

$$\omega_c = \frac{\omega_p}{[10^{-K_p/10} - 1]^{1/LN}}$$

$$= 2 \text{ rad/sec}$$

5)

$$H_a(s) = H_2(s) \Big|_{s \rightarrow \frac{s}{\omega_c}} = \frac{s}{2}$$

$$= \frac{4}{s^2 + 2\sqrt{2}s + 4}$$

6)

$$H(z) = H_a(s) \Big|_{s \rightarrow \frac{2}{T} \left[\frac{1-z^{-1}}{1+z^{-1}} \right]}$$

$$T = 1 \text{ sec}$$

$$\frac{4}{\left[\frac{1-z^{-1}}{1+z^{-1}} \right]^2 + 2\sqrt{2} \times 2 \times \left[\frac{1-z^{-1}}{1+z^{-1}} \right] + 4}$$

$$= \frac{4}{(1+z^{-1})^2}$$

$$3.14142 + 0.5858 z^{-2}$$

T.F in the form of difference eqn

$$H(z) = \frac{1 + 2z^{-1} + z^{-2}}{3.4142 + 0.5858z^{-2}} = \frac{Y(z)}{X(z)}$$

$$X(z) + 2z^{-1}X(z) + z^{-2}X(z) = 3.4142Y(z) + 0.5858z^{-2}Y(z)$$

($z = e^{j\omega}$)

taking IZT

$$3.4142 y(n) + 0.5858 y(n-2) = x(n) + 2x(n-1) + x(n-2)$$

$$\therefore y(n) = -0.1716 y(n-2) + 0.2928 x(n) + 0.5857 x(n-1) + 0.2928 x(n-2)$$

design a digital LPF to meet the foll specifications

- (i) pass band ripple = 3 dB
 - (ii) pass band edge freq = 0.5π rad
 - (iii) min stop band atten = 15 dB
 - (iv) stop band atten = ~~15~~ 0.75π rad
- magnitude response is monotonic outside the pass band

I $K_p = -3 \text{ dB}$ $K_s = -15 \text{ dB}$
 $\omega_p = 0.5\pi \text{ rad}$ $\omega_s = 0.75\pi \text{ rad}$

$$\Omega_p' = \frac{2}{T} \tan\left[\frac{\omega_p}{2}\right] = 2 \text{ rad/sec}$$

$$\Omega_s' = 4.8284 \text{ rad/sec}$$

II $K_p = -3 \text{ dB} = 20 \log\left(\frac{1}{\sqrt{1+\epsilon^2}}\right)$

$$\epsilon = 0.99763$$

$$\delta_p = 1 - \frac{1}{\sqrt{1+\epsilon^2}} = 0.29205$$

$$K_s = -15 \text{ dB} = 20 \log \delta_s$$

$$\therefore \delta_s = 0.17783$$

~~order of~~ order of the filter

$$N \geq \frac{\cosh^{-1}(1/d)}{\cosh^{-1}(1/k)}$$

$$d = \sqrt{\frac{(1-\delta_p)^2 - 1}{(\delta_s^2 - 1)}} = 0.180$$

$$k = \frac{\Omega_p'}{\Omega_s'} = 0.41422 \quad N \geq 1.58$$

$$N = 9$$

$$a = \frac{1}{2} \left[\frac{1 + \sqrt{1 + \epsilon^2}}{\epsilon} \right]^{1/N} - \frac{1}{2} \left[\frac{1 + \sqrt{1 + \epsilon^2}}{\epsilon} \right]^{-1/N} \quad (37)$$

$$a = 0.4560$$

$$b = \frac{1}{2} \left[\frac{1 + \sqrt{1 + \epsilon^2}}{\epsilon} \right]^{1/N} + \frac{1}{2} \left[\frac{1 + \sqrt{1 + \epsilon^2}}{\epsilon} \right]^{-1/N}$$

$$b = 1.09906$$

$$v_k = -a \sin \left[(2k-1) \frac{\pi}{2N} \right]$$

$$w_k = b \cos \left[(2k-1) \frac{\pi}{2N} \right]$$

$$k: 1, \dots, 2N$$

$$k_2: 1, \dots, 4$$

k	w_k	$-v_k$
-----	-------	--------

1	-0.32244	0.77715
---	----------	---------

2	-0.32244	-0.77715
---	----------	----------

$$V_N(s) = V_2(s) = (s - s_1)(s - s_2)$$

$$(s + 0.32244 - j0.77715)(s + 0.32244 + j0.77715)$$

$$V_2(s) = s^2 + \underbrace{0.6449s}_{b_1} + \underbrace{0.70800}_{b_0}$$

$$K_{\text{res}} = K_2 = \frac{b_0}{\sqrt{1+\epsilon^2}} = 0.50123$$

$$H_2(s) = \frac{0.50123}{s^2 + 0.6449s + 0.7080}$$

$$H_a(s) = H_2(s) \Big|_{s \rightarrow \frac{s}{\Omega_p}}$$

$$= \frac{0.50123}{\left(\frac{s}{2}\right)^2 + 0.6449\left(\frac{s}{2}\right) + 0.7080}$$

$$H_a(s) = \frac{2.00492}{s^2 + 1.2898s + 2.83204}$$

IV

$$H(z); H_a(s) \Big|_{s \rightarrow \frac{2}{T} \frac{1-z^{-1}}{1+z^{-1}}}$$

$$H(z) = \frac{2.00492 (1+z^{-1})^2}{2(1-z^{-1})^2 + 1.2898(1-z^{-2}) + 2.83204}$$

$$2(1-z^{-1})^2 + 1.2898(1-z^{-2}) + 2.83204 \Big|_{1+z^{-1}}$$



Module -5

DIGITAL SIGNALPROCESSORS

Basic Architectural Features

- ✓ PDSP's should provide instructions similar to microprocessors.
- ✓ Basic computational capabilities provided by the way of instructions should include the following:
 1. Arithmetic operations
 2. logical operations
 3. MAC operations
 4. Signal scaling operations
- ✓ To perform all these operations a dedicated high speed H/W must be provided.
- ✓ The architecture should include the following H/W features also:

1. on chip registers – storage of intermediate results
2. on chip memories – signal samples (RAM)
3. on chip pgm memory – pgms & fixed data such as filter coefficients (ROM).

1. Investigate the basic features that should be provided in DSP architecture to be used to implement the following Nth order FIR filter.

$$Y(n) = \sum_{i=0}^{N-1} h(i) x(n-i) ; \quad n = 0,1,2,\dots$$

Where $x(n)$ denotes the i/p samples

$y(n)$ the o/p samples

$x(n-i)$ is the i/p sample I samples earlier than $x(n)$

and $h(i)$ the i^{th} filter coefficients


DSP Computational Building Blocks

○ The basic building blocks that are essential to carry out DSP computations are as follows :

1. multiplier
2. shifter
3. MAC unit
4. ALU

MULTIPLIER

- ✓ Earlier multiplication schemes relied either on **S/W** or **Micro coded controllers**
- ✓ Both these options require several processor cycles to complete the multiplication
- ✓ The advances made in VLSI technology in speed & size made possible the H/W implementation of parallel multipliers

- 
- ✓ Before designing an actual multiplier, the specifications such as speed , accuracy and dynamic range must be clear.
 - ✓ accuracy and dynamic range – is decided based on the number of bits used to represent the multiplication operands and whether they are represented in fixed point or floating point format.
 - ✓ Speed – is decided by the architecture employed.

Parallel Multiplier

Let us consider the multiplication of 2 unsigned numbers A & B

Let

A --- represents m bits multiplicand $[A_{m-1}, A_{m-2} \dots A_0]$

B --- represents n bits multiplier $[B_{n-1}, B_{n-2} \dots B_0]$

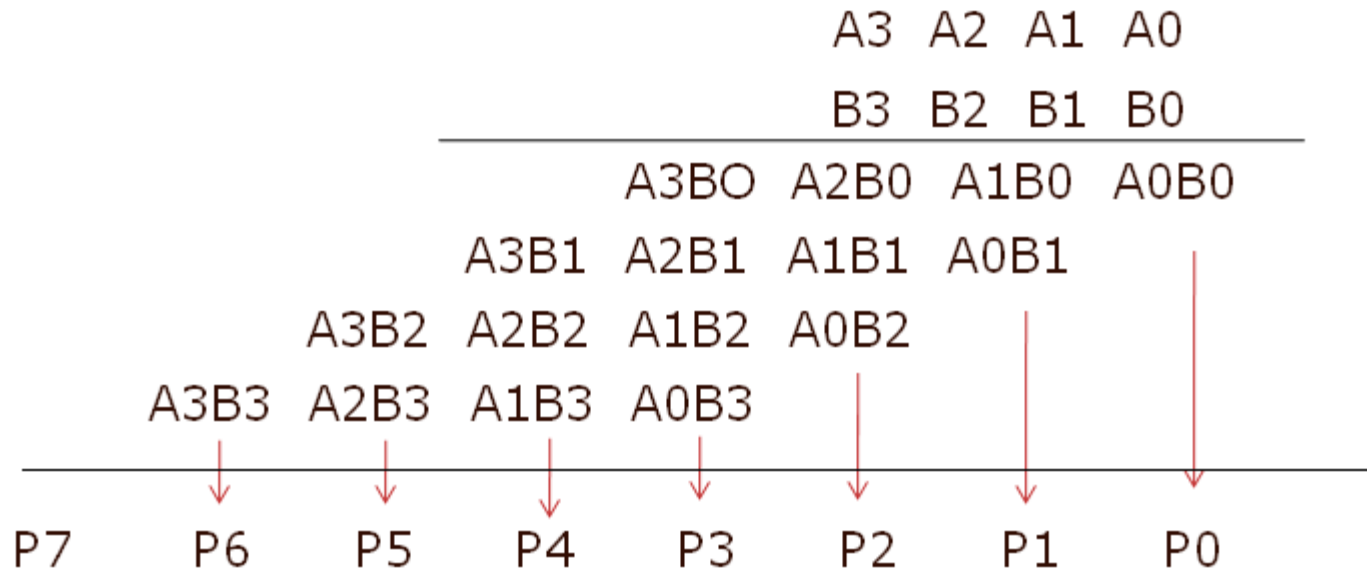
P --- product of A & B . Max (m+n) bits

$$A = \sum_{i=0}^{m-1} A_i 2^i \quad \text{-----1}$$

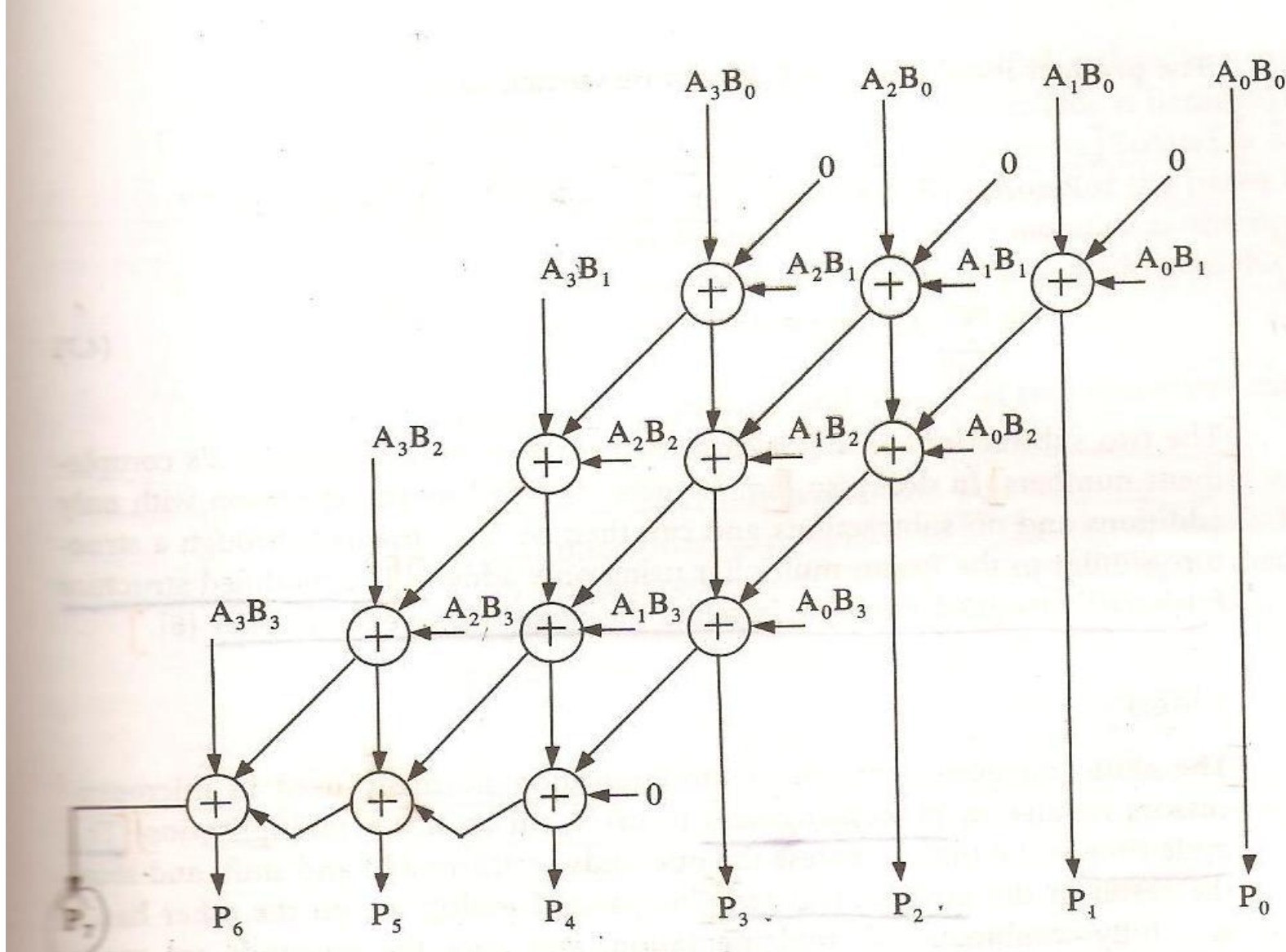
$$B = \sum_{j=0}^{n-1} B_j 2^j \quad \text{-----2}$$

$$P = \sum_{j=0}^{n-1} \left[\sum_{i=0}^{m-1} A_i B_j 2^{i+j} \right] \quad \text{-----3}$$

The multiplication operations using 4 bit s for A & B are shown below.



The fig below shows the H/W structure of the multiplier for this example and is called as Braun multiplier.



The structure of 4 × 4 Braun multiplier

- ✓ For a $n \times n$ multiplier we require $n(n-1)$ adders.
- ✓ The Structure requires : 12 [3 i/p & 2 o/p]
- ✓ Braun's multiplier does not take in account the signs of the numbers that are being multiplied.
- ✓ Additional H/W is required before & after the multiplication when signed numbers represented in 2's complement are used
- ✓ Let us consider A & B represented in 2's complement format
A & B having m & n bits respectively

$$A = -A_{m-1}2^{m-1} + \sum_{i=0}^{m-2} A_i 2^i \quad \text{-----1}$$

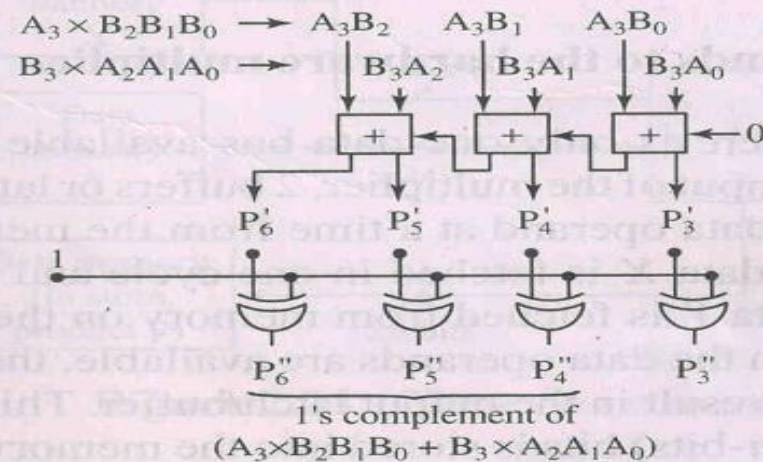
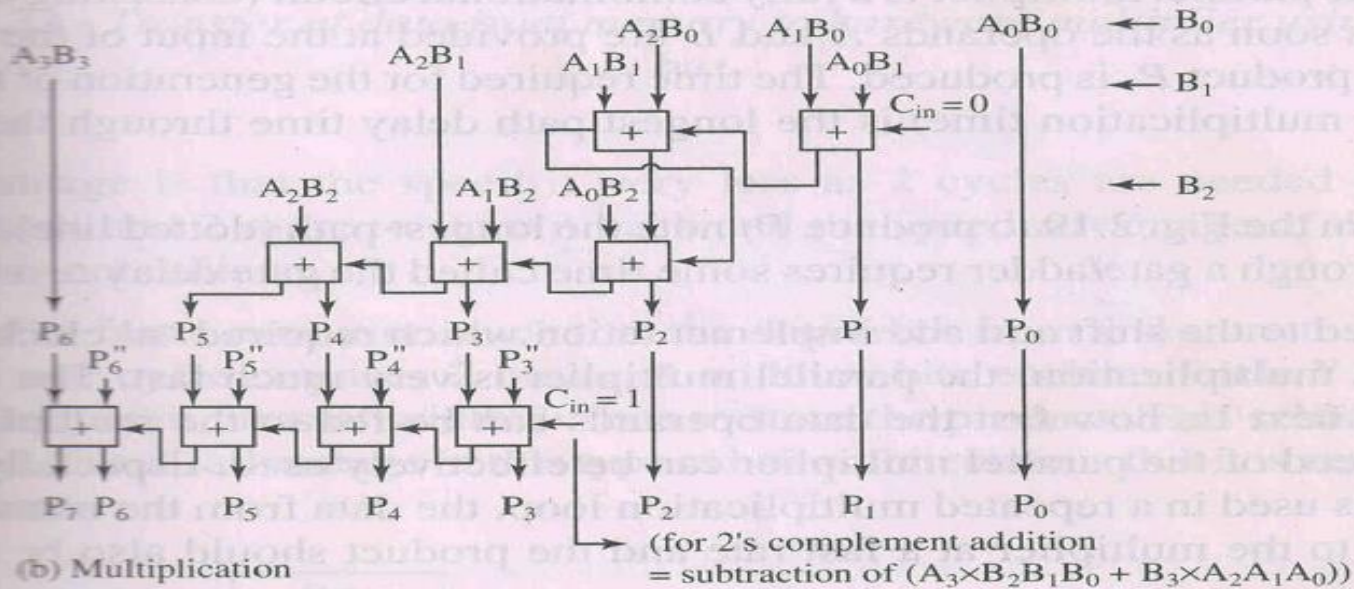
$$B = -B_{n-1}2^{n-1} + \sum_{j=0}^{n-2} B_j 2^j \quad \text{-----2}$$

The product $p = p_{m+n-1}$ is written as

$$p = A_{m-1}B_{n-1}2^{m+n-2} + \sum_{i=0}^{m-2} \sum_{j=0}^{n-2} A_i B_j 2^{i+j} \quad \text{-----3}$$

- ✓ the 2 subtractions in eqn 3 can be expressed as addition of 2's complement number.
- ✓ This can be implemented through a structure similar to the Braun multipliers using only adders.
- ✓ This modified structure is called as Baugh Wooley multiplier.

Baugh wooley multiplier is shown in Fig. 2.22.



(b) Generation of $P_3'' - P_6''$


2.22 Structure of Baugh wooley multiplier for signed number multiplication of two 4-bit signed numbers.

shifters

- ❖ Is an essential component of any DSP architecture.
- ❖ Required to scale down or scale up the operands & results to avoid errors resulting from overflows & underflows during computation.

Let us consider the following cases.

1. Let us consider the sum of 'N' numbers each represented by n bits.
 - As accumulated sum grows the number of bits required to represent it also increases
 - The max no of bits to which the sum can grow is $(n + \log_2 N)$ bits

- 
- If each of the N no is scaled down by $\log_2 N$ bits prior to addition, the loss of the result due to overflow can be avoided.
 - - The accumulator will then hold the sum scaled down by $\log_2 N$ bits.
 - Accuracy of sum is lost but the summation would be completed with out overflow error.
 - The actual sum can be obtained by scaling up the result by $\log_2 N$ bits when required.

2. When 2 no's each of 'n' bits are multiplied, the product can have a max of '2n' bits

- when this product is saved in memory which is 'n' bit wide, the lower order 'n' bits are generally discarded resulting in loss of accuracy
- in case of multiplication of 2 signed no's, the accuracy can be slightly improved
- By shifting the product by one bit position to the left before saving 'n' higher order bits.
- The accuracy improves because instead of discarding n bits we now discard (n-1) bits

3. When carrying out floating point additions

- ➤ the operands should be normalized to have same exponent.
- This is done by shifting one of the operands by required no of bit positions

Problems:

1. It is required to find the sum of 64 no's each represented by 16 bits. How many bits should the accumulator have so that the sum can be computed with out the occurrence of overflow error or loss of accuracy?

Ans:- sum grows by a max of $\log_2 64 = 6$ bits

to avoid over flow the number of bits the accumulator should have is $16+6 = 22$

2. If for problem 1 it is decided to have an accumulator with only 16 bits but shift the numbers before addition to prevent overflow. By how many bits should each number be shifted?

By 6 bits to right since the sum grows by 6 bits

3. If all the numbers in problem 2 are fixed point integers, what is the actual sum of the numbers

The actual sum = (contents of accumulator) $\times 2^6$

4. What is the error in computation of the sum in problem 3

Since 6 lower bits are lost the sum could be off by as much as $2^6 - 1 = 63$

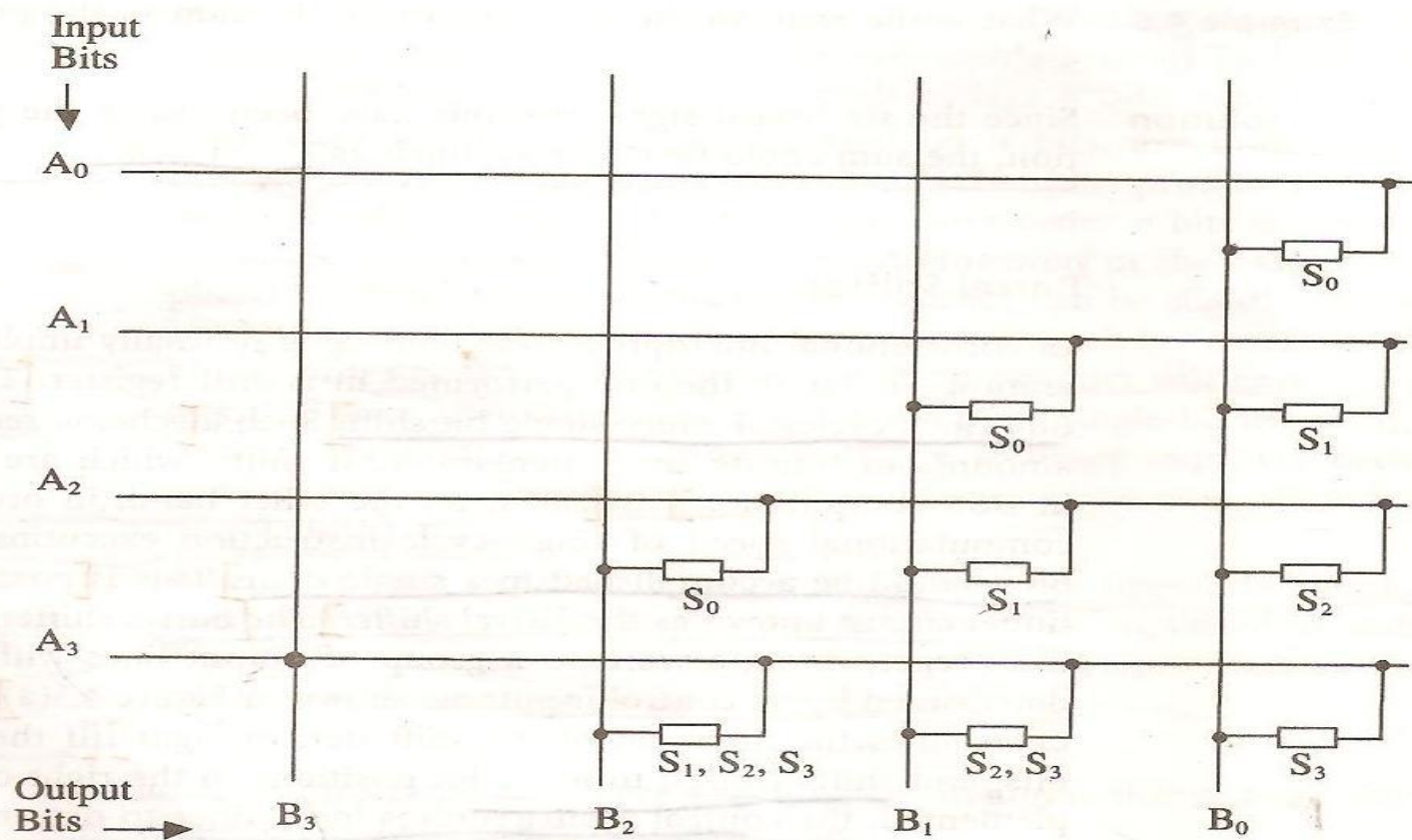
Barrel Shifter

- In μ ps shifting is implemented by a operation similar to one performed in a shift register .
- The operation takes one clock cycle for every single bit shift.
- In DSP's shifting of several bits in a single cycle is possible by a combinational ckt known as Barrel Shifter.
- Barrel shifter connects the i/p lines representing a word to a group of output lines with the required shift which is determined by its control inputs as shown below
- Control i/ps also determines the direction of the shift (left or right)



B.D of barrel shifter

- If the input word has 'n' bits, shift is from 0 to n-1 bits
- The control i/p requires $\log_2 n$ lines to determine the no. of bits to be shifted.
- One more line is required to indicate the direction of the shift.
- **Left shift:-** bits shifted out of the i/p word are discarded and new bits positions are filled with zeros.
- **Right shift:-** the new bit positions are replicated with the MSB to maintain the sign of the shifted result.



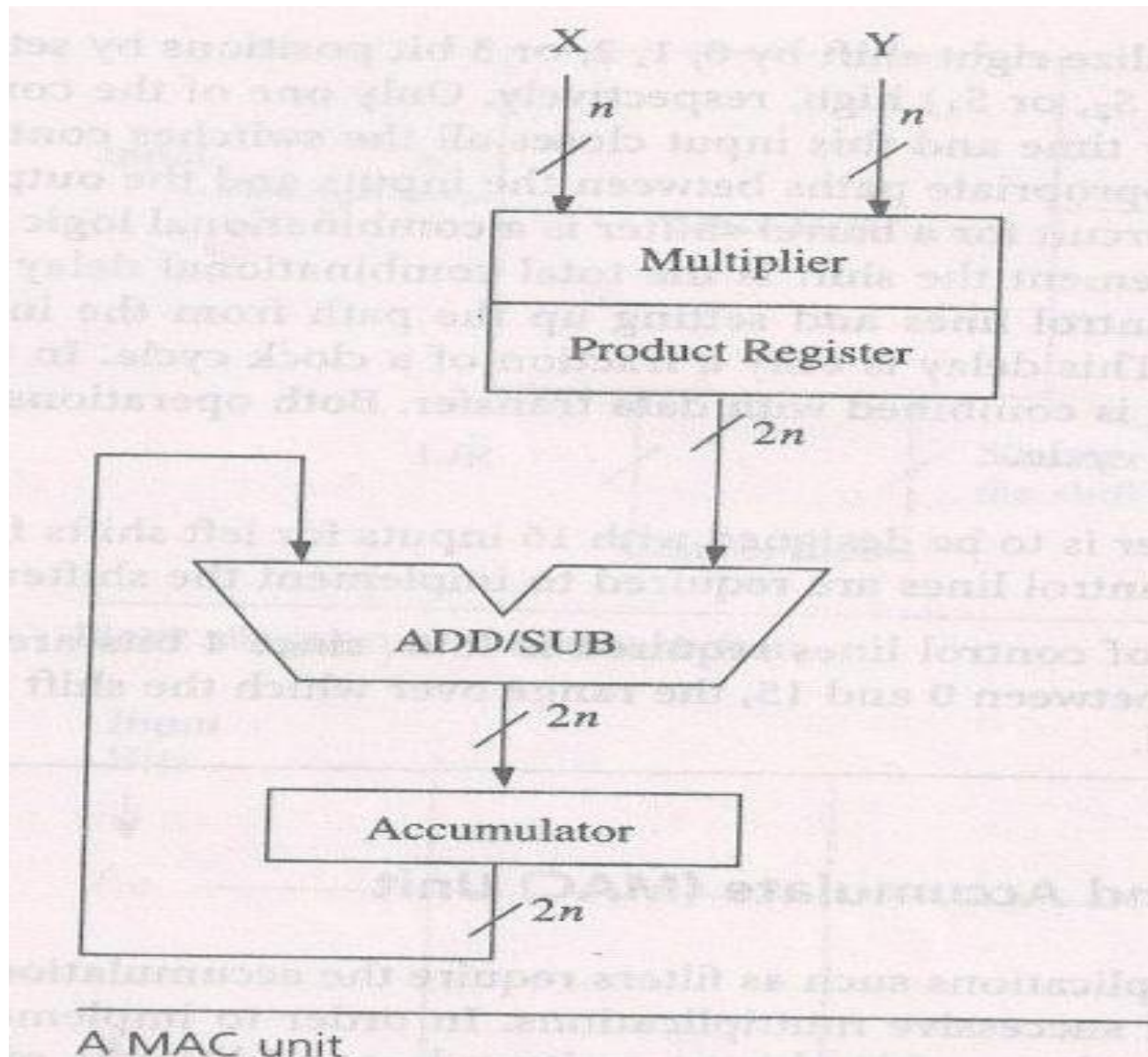
Input	Shift (Switch)	Output ($B_3B_2B_1B_0$)
$A_3A_2A_1A_0$	0 (S_0)	$A_3A_2A_1A_0$
$A_3A_2A_1A_0$	1 (S_1)	$A_3A_3A_2A_1$
$A_3A_2A_1A_0$	2 (S_2)	$A_3A_3A_3A_2$
$A_3A_2A_1A_0$	3 (S_3)	$A_3A_3A_3A_3$

1. A barrel shifter is to designed with 16 inputs for left shift form 0 to 15 bits. How many control lines are required to implement the shifter .

ANS: 4 control lines are required.

Multiply & Accumulate (MAC) Unit

- The configuration of a multiply & accumulate unit is commonly known as MAC unit as shown below.
- The MAC unit is used to implement functions of the type $A + BC$.
- Multiplication & accumulation are 2 distinct operations, each normally require separate instruction execution cycle.
- But they can work in parallel.



Pipelined operation of MAC

- The pipelined operation makes use of the fact that the multiplier and adder can work separately and simultaneously.
- Consider an example

$$y(n) = x_1h_1 + x_2h_2 + x_3h_3$$

Cycle time number	Multiplier	Adder
1	x_1h_1	-
2	x_2h_2	$A+x_1h_1 \rightarrow A$
3	x_3h_3	$A+x_2h_2 \rightarrow A$
4	-	$A+x_3h_3 \rightarrow A$

- At a time when the multiplier is computing the product , the accumulator accumulates the product of the previous multiplication.
- if N products are to be accumulated, N-1 multiplies overlap with accumulations.
- During every first multiply, the accumulator is idle since there is nothing to accumulate.
- Like wise during the very last accumulation, the multiplier is idle since all the 'N' products have been completed.
- It takes a total number of 'N+1' instruction execution cycles to complete the sum of products of 'N' multiplications.
- If N is large the pipelined operation of multiplier & accumulator work in parallel to execute a MAC operation per cycle.

1. If a sum of 256 products is to be computed using a pipelined MAC unit, & if the MAC execution time of the unit is 100 nsec, what will be the total time required to complete the operations?

ANS: total time required = $257 \times 100 \times 10^{-9}$ sec = 25.7 μ sec.

Over flow & under flow

When designing a MAC attention has to be paid for

1. word size at the i/p of the multiplier
2. the sizes of the add/ subtract unit &
3. The accumulator

as over flow & under flow conditions can be encountered .

- Techniques to prevent over flow & under flow conditions are
 1. Barrel shifters at i/p & o/ps of MAC unit
 2. Guard bits in the accumulator
 3. Saturation logic

1. Barrel shifters

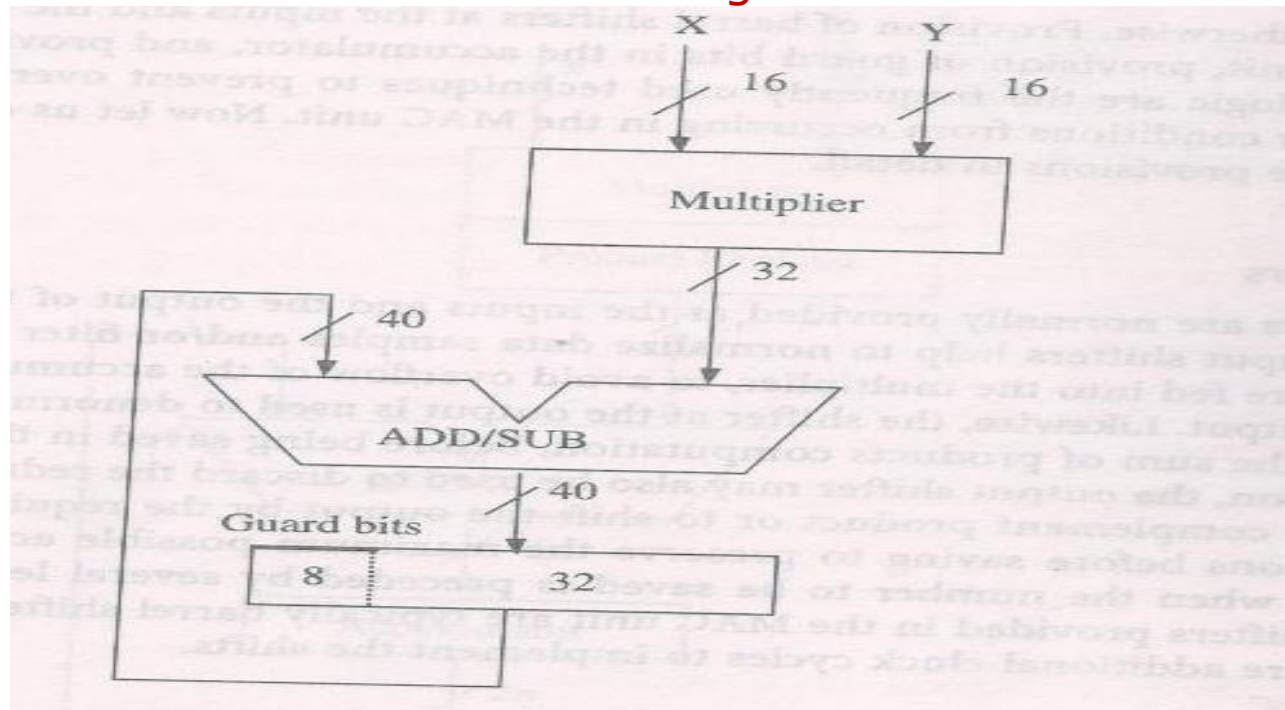
- ❖ shifters are normally provided at i/p & o/ps of MAC unit.
- ❖ the i/p shifters helps to normalize the data samples as they are fed to the multiplier.
- ❖ The shifter at the output are used to denormalize the result after the sum of products before storing in memory.
- ❖ o/p shifters can also be used to discard the redundant sign bits or to shift the o/p by required no. of positions.

2. Guard bits:

- ❖ If accuracy is preserved, the i/ps are not normalized.
- ❖ When repetitive MAC operations are performed the accumulator sum grows with each MAC operation.
- ❖ This increases the no. of bits required to present the result w/o loss of accuracy.
- ❖ So extra bits must be provided in accumulator called as guard bits or extension bits.
- ❖ After completion of computation, the required sum of product , the extension bits may be saved as a separate word if required.
- ❖ Or. The sum along with the guard bits may be shifted by required amount & be saved as a single word.

2. consider a MAC units whose inputs are 16 bits nos. if 256 products are to be summed up in this MAC , how many guard bits should be provided for the accumulator to prevent overflow condition from occurring.

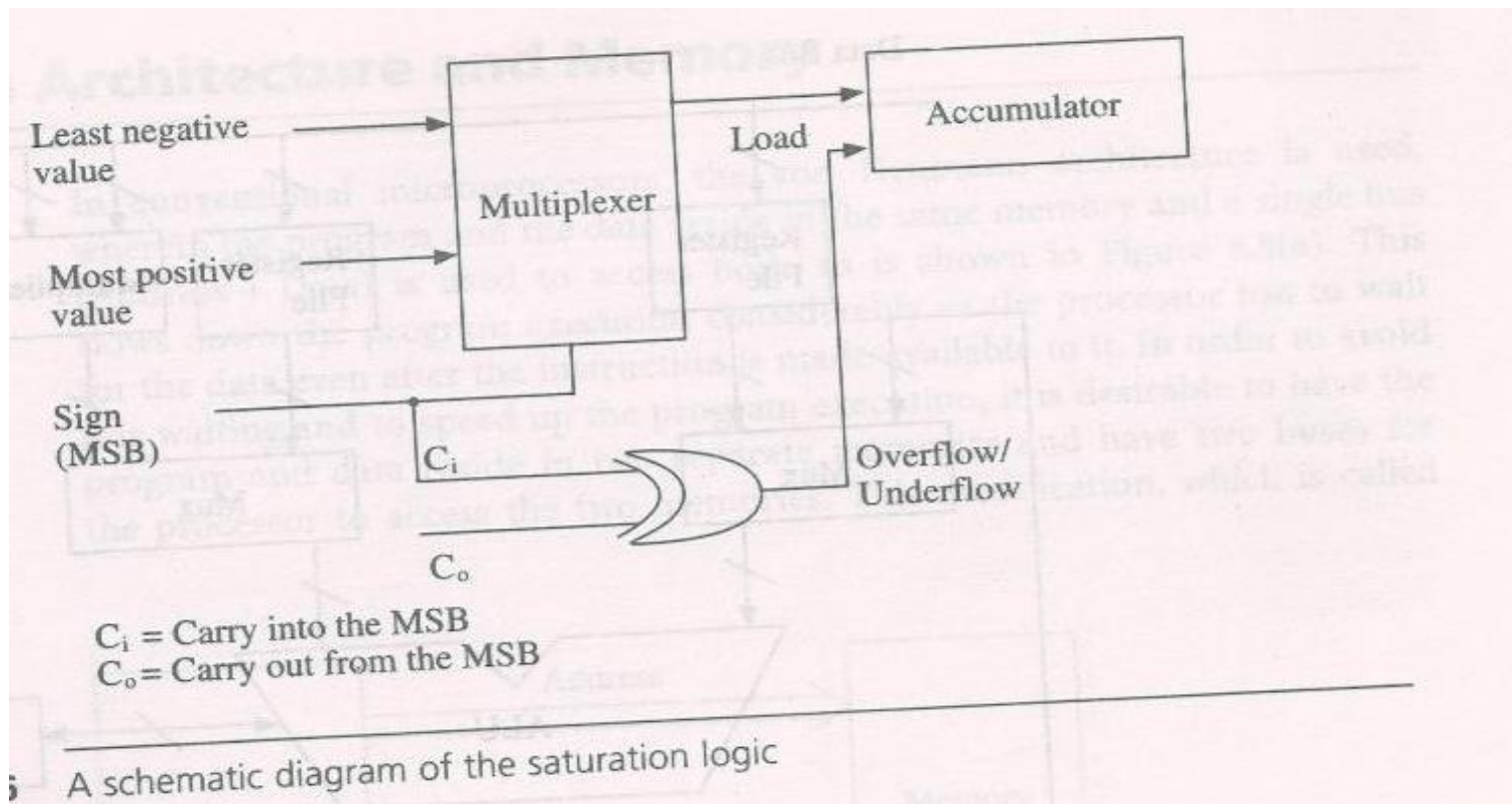
Ans: 16×16 multiplication has 32 bits. Since 256 such products are summed, the sum can grow by max of $\log_2 256 = 8$ bits guard bits should be provided for the accumulator to prevent overflow condition from occurring = 8 bits.



3. Saturation logic: -

- ❖ With or w/o guard bits ,an over flow condition occurs when the accumulated result becomes larger than the largest no. it can hold.
- ❖ When handling a -ve no. an under flow will occur if the accumulator becomes smaller than the smallest no. it can hold.
- ❖ So it is better to limit the accumulator contents to most +ve (most -ve) to avoid an error known as wrap around error.
- ❖ Limiting the accumulator contents to its saturation limit is achieved with a simple logic ckt called as saturation logic as shown below.
- ❖ This ckt detects the overflow / underflow condition & accordingly loads the accumulator with to most +ve or most -ve value

- the overflow / underflow condition is detected by monitoring the carry into the MSB & the carry out of MSB.
- If carry-in \neq to carry out , the overflow / underflow occurs
- The selection bet' the most +ve or most -ve value is made based on the sign bit of the no.



Module-1

COMPUTATIONAL ACCURACY IN DSP APPLICATIONS

- Number formats for signals & coefficients in DSP systems:-

Conditions / parameters to represent Numbers:-

- Range
- precision of signals
- coefficients to be represented
- Hardware complexity
- speed Requirements

FORMATS

(i) Fixed-point format

Problem: What is the range of numbers that can be represented in a fixed-point format using 16 bits if the numbers are treated as (a) signed integers, (b) signed fractions?

(A)

$n = 16$

(a) Range:- -2^{n-1} to $2^{n-1} - 1$
 -2^{15} to $2^{15} - 1$

= -32768 to 32767

(b) Range = -1 to $(1 - 2^{-(n-1)})$

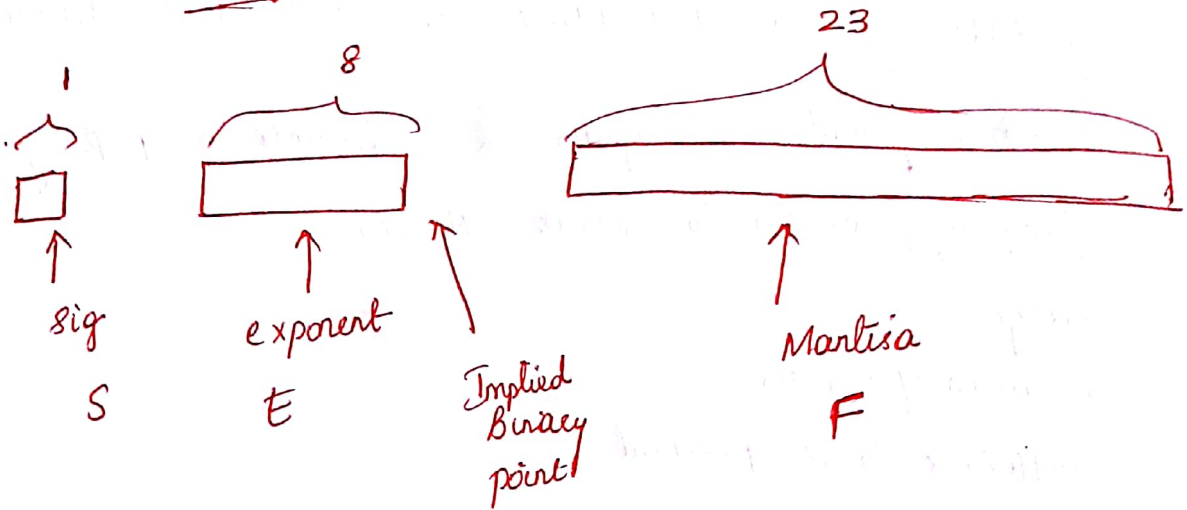
-1 to $(1 - 2^{-15})$

-1 to $+0.999969482$

(ii) Double precision fixed-point format

2

a) Floating-point format



Let M_x - mantissa

E_x - Exponent

value of $x = M_x \cdot 2^{E_x}$

Consider 2 floating point numbers x & y

then product

$$xy = M_x M_y \cdot 2^{E_x + E_y}$$

The value represented by the data format

$$x = (-1)^S \cdot 2^{(E - \text{bias})} \cdot 1.F$$

F - magnitude fraction of the mantissa

E - integer

S - signed bit

bias - depends upon the bits reserved for the exponent.

$$\text{Range} = - (2 - 2^{-n}) \text{ to } + (2 - 2^{-n})$$

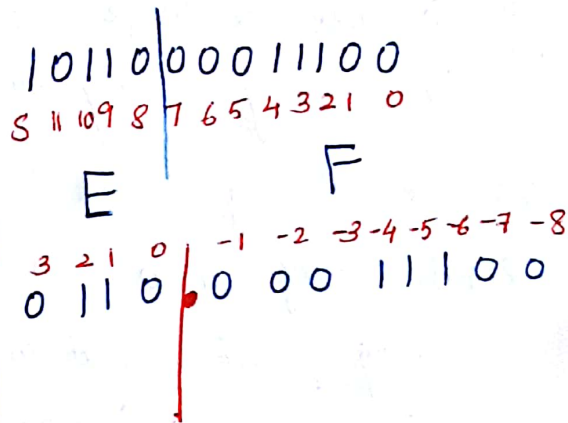
Ex) Find the decimal equivalent of the floating-point binary number 1011000011100. Assume a format similar to IEEE-754 in which the MSB is the sign bit followed by 4 exponent

bits followed by 8 bits for the fractional part.

(3)

(A) Since MSB = 1
The Number is Negative

F =



$$F = 2^{-4} + 2^{-5} + 2^{-6}$$

$$= .109375$$

$$E = 2^1 + 2^2 = 6$$

∴ Value of Number is

$$x = (-1)^S \times 2^{(E - \text{bias})} * 1.F$$

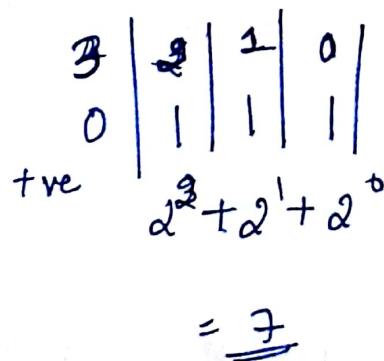
$$x = (-1)^1 * 2^{(6 - 7)} * 1.109375$$

$$x = \underline{\underline{-0.5546875}}$$

Bias - Max positive
Number in

E bits

$$E = 4 \text{ bits}$$



(Q). Using 16 bits for the mantissa and 8 bits for the exponent, what is the range of numbers that can be represented using the floating point format similar to IEEE-754?

(A) The most -ve number will have its mantissa = $-2 + 2^{-16} \therefore (-2 + 2^{-n})$

$$\text{Exponent} = (2^n - 1) - (2^{n-1} - 1)$$

$$n = 8$$

$$= 255 - 127$$

$$= 128 \quad ; \quad \text{-ve No} = -1.999984741 \times 2^{128}$$

The most +ve number is $+1.999984741 \times 2^{128}$

BLOCK FLOATING POINT FORMAT

increases the range & precision of a given fixed point format by retaining as many lower-order bits as is possible.

Q) The following 12-bit Binary fractions are to be stored in an 8-bit memory - show how they can be represented in Block Floating point format so as to improve accuracy

- 000001110011
- 000011110000
- 000000111111
- 000010101010

(A) Using 8-bit fixed point format

$$\boxed{0000}0111 / 0011$$

$$\boxed{0000}1111 / 0000$$

$$\boxed{0000}0011 / 1111$$

$$\boxed{0000}1010 / 1010$$

Last 4 bits will be discarded.

But in all 4 numbers, there are 4 leading zeros

so

$$01110011 \times 2^{-4}$$

$$11110000 \times 2^{-4}$$

$$00111111 \times 2^{-4}$$

$$10101010 \times 2^{-4}$$

DYNAMIC RANGE & PRECISION

The Dynamic range of a signal is the ratio of the maximum value to the minimum value that the signal can take in the given number representation scheme.

→ Dynamic Range & No. of Bits used to represent

→ Increases by 6dB for every additional bit.

Resolution is the minimum value that can be represented using a number representation format.

Consider N bits

$$\text{Resolution} = 1/2^N \text{ for large } N$$

Precision is an issue related to the speed of the DSP implementation.

Q) Calculate the dynamic range and precision of each of the following Number representation formats

a) 24 bit, single precision, fixed point format

Since each bit gives a dynamic range of 6 dB, total dynamic range is $24 \times 6 = 144$ dB

$$\text{Percentage resolution is } 1/2^{24} \times 100 = \underline{6 \times 10^{-6}} \%$$

b) 48-bit, double-precision, fixed point format

Each bit gives dynamic range of 6 dB

$$\text{total dynamic range} = 48 \times 6 = \underline{288} \text{ dB}$$

$$\text{Percentage resolution is } 1/2^{48} \times 100 = 4 \times 10^{-13} \%$$

c) a floating point format with a 16-bit mantissa & 8-bit exponent.

For 8 bit exponent bits, dynamic range is $(2^8 - 1) \times 6 = \underline{1530}$ dB

% resolution depends on 16 bit mantissa

$$\underline{\left(1/2^{16}\right) \times 100 = 1.5 \times 10^{-3} \%}$$