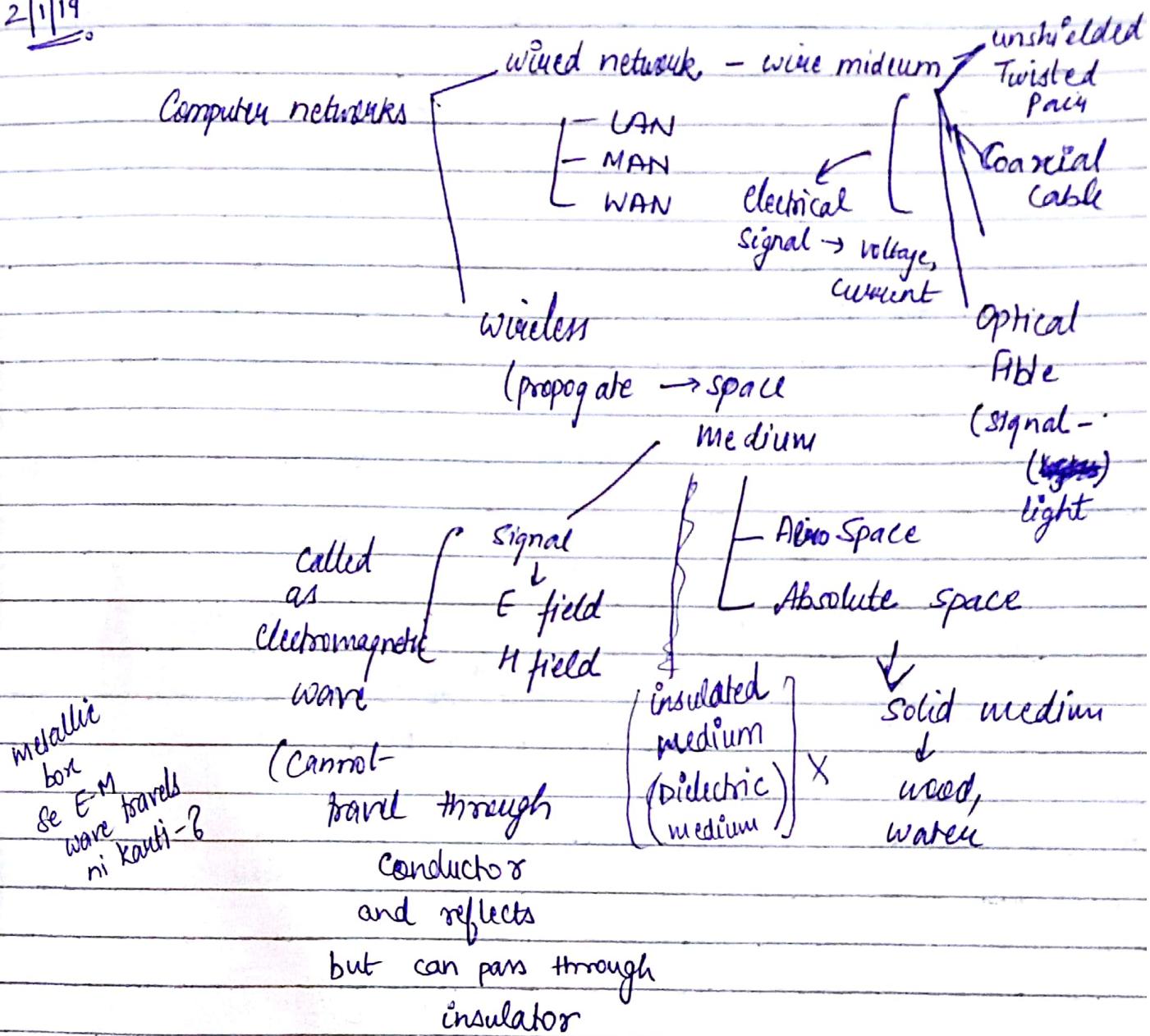


2/1/19



8/1/19

Insulator  
(Dielectric Medium)

Width of a road is the bandwidth wood, water, air space

↳ for jitha broad utne channels

Franenbaum

- wireless signal generation, propagation and reception
- modulation Techniques
- Multiple access schemes

↓ data

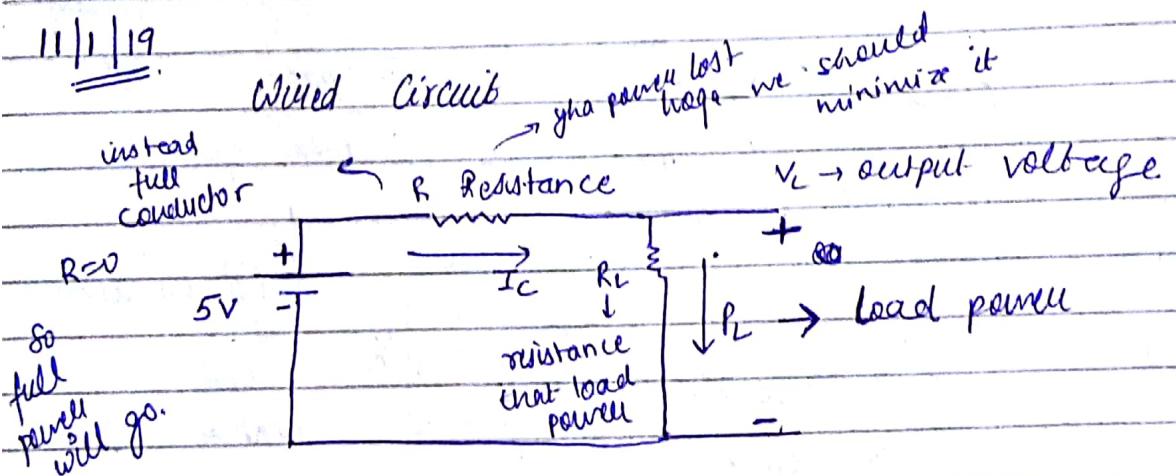
→ Mobile wireless communication →

- 1G →
- 2G → (GSM) + GPRS & EDGE
- 3G →
- 4G →

→ wireless LAN - WiFi

a, b, g, h, n  
Platforms.

11/1/19



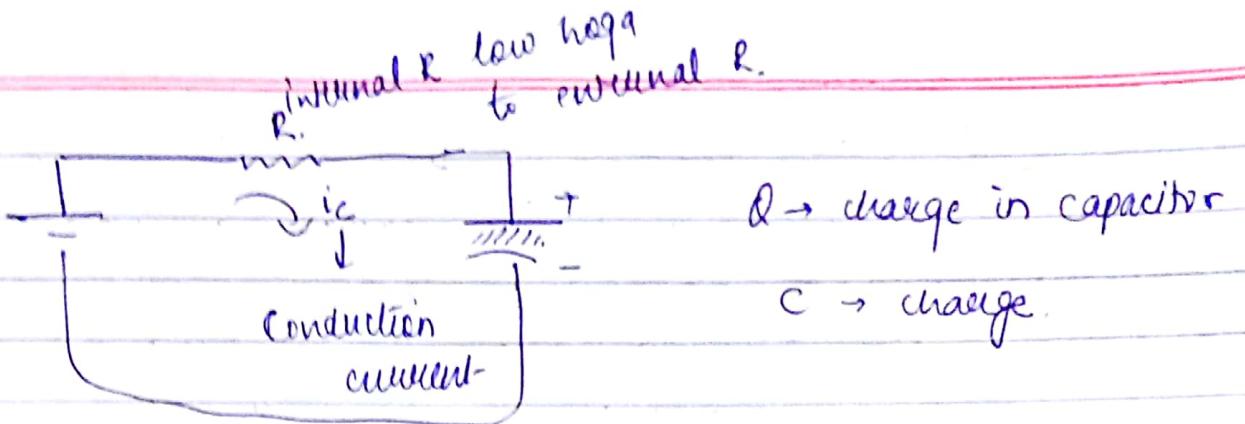
$$I_C = \frac{V}{R + R_L}$$

$$V_L = I_C \times R_L$$

$$P_L = V_L \times I_C = I_C^2 \times R_L$$



dielectric medium.



$$V_C = \frac{Q}{C} = \frac{1}{C} \int_0^t i_c dt$$

capacitor cannot be charged continuously.

when  $t = 0$  switch on kia to capacity or  
pe charge nhi hogta.

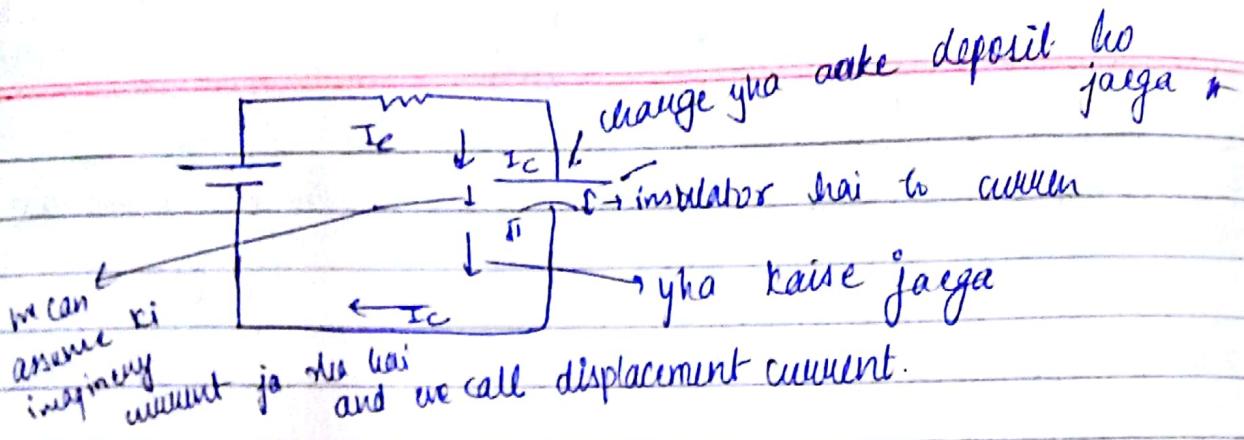
when  $t=0$ ,  $I_C(0+) = \frac{V}{R}$

when  $t$  is something then time changing with respect to capacitor charge hoga.

$$I_C = \frac{V - V_C(t)}{R}$$

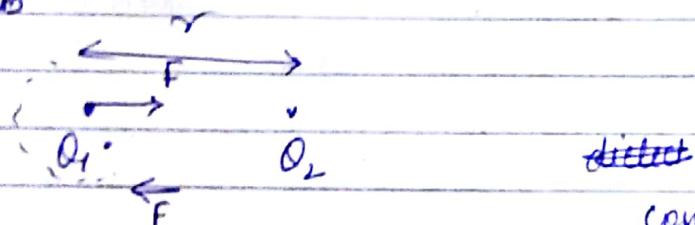
After some time capacitor me poora charge aa jaega  
to ~~resistance~~ resistance across charge o ho jaega

current  $\rightarrow$  to  
electrons  
 $\leftarrow$  direction me flow hota hai



\* to jo plates neutral hai use free charge aa jaenge to electric field set ho jaegi aur dusri plate ke charge ko repel karega use emf wave karega

currents



$$F \propto \frac{Q_1 \times Q_2}{r^2}$$

conductor - conductivity  
insulator  $\rightarrow$  permittivity

$$\vec{F} = \frac{1}{c} \frac{Q_1 \times Q_2}{r^2}$$

permittivity of the medium

~~die~~

I C experiences force from Q

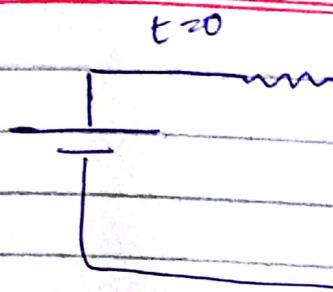
$\vec{F} = \frac{1}{c} \frac{Q \times I}{R^2}$  is an electric field  
na kah ke E hoga

$$\vec{E} = \frac{1}{c} \frac{Q \times I}{R^2}$$

$$W = F \times d$$

$$= E_x d$$

The electric field  
having voltage  
bega.



voltage

how much work to  
be done  
to move a  
unit charge

Voltage  
do points  
be bleach hai  
kai

Electric  
field  
ata  
point  
kai  
hai

When same t.

$$I_d(t) = I_C(t)$$

$I_C(t)$  = hala jaega

(v) alternating  
voltage

Voltage at a point  
is infinity se  
unit charge  
ko long  
is plate ke

unit charge  
ko leke  
ayenge

$\downarrow v_r$

$v_2 - v_1$

the cap  
voltage

Opposite use

se to E change

to get

charging  
voltage

bega to

a point  
from plate I  
to plate II

integration karenge

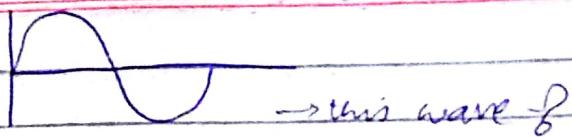
$I_C \rightarrow$  changing current

$\vec{d} \rightarrow$  changing displacement

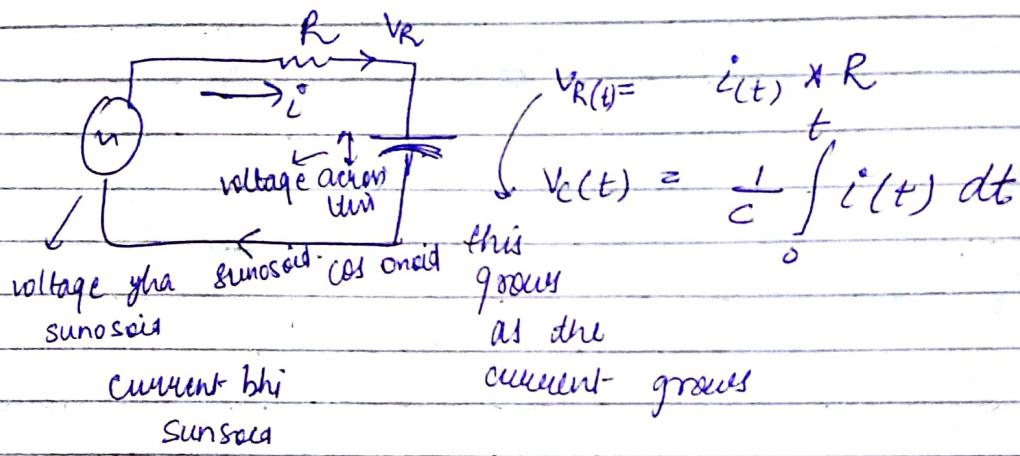
$I_C$



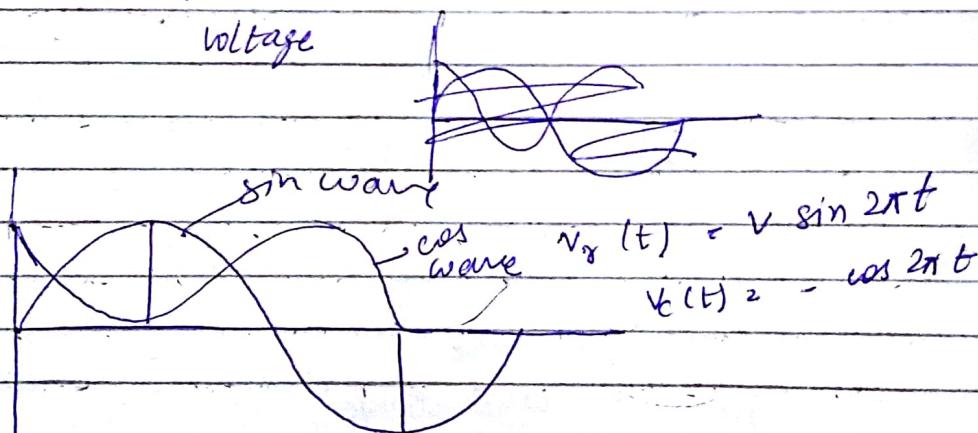
## sineoid - b



When I have a sinusoidal voltage  
RC-Circuit

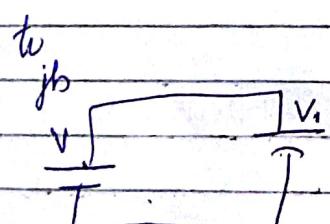


16/1/19



$$\cos \phi = \sin(\theta + \pi/2)$$

if static voltage has



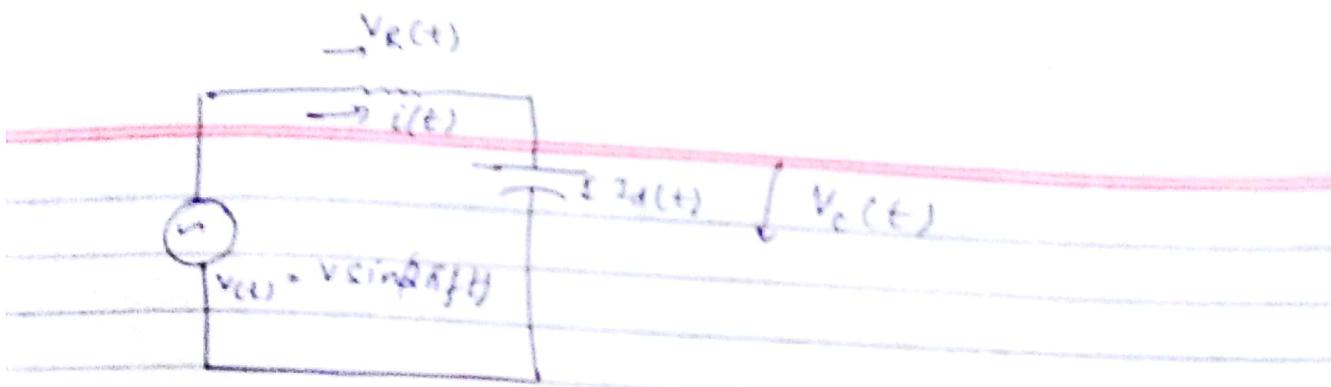
$$V = V_0$$

no go to  
current flow  
how bnd  
no jiega

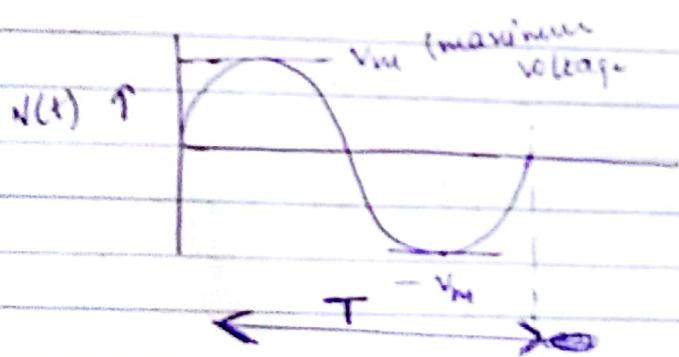
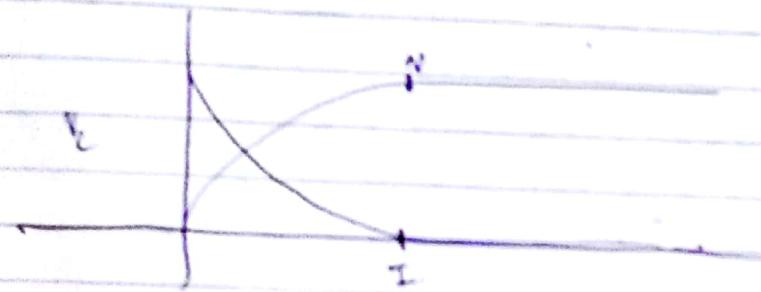
but in case

of alternating voltage

After



16/1/19.



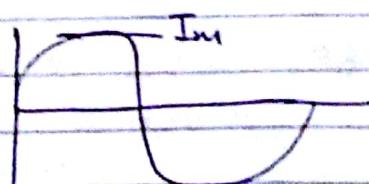
$$V(t) = V_m \sin 2\pi f t$$

$$= V_m \sin 2\pi \frac{1}{T} t$$

Alternating voltage

$$i(t) = I_m \sin 2\pi f t$$

$$V_R(t) = i(t) \times R = V_R \sin 2\pi f t$$



$$V_R = I_m \times R$$

$$V_c(t) = \frac{Q(t)}{C}$$

~~then~~

voltage in capacitor is change

Since change is continuously changing  
so integration

$$V_c(t) = \frac{1}{C} \int i_d(t) dt$$

~~Current across the capacitor  
differential of  $i_d(t)$~~

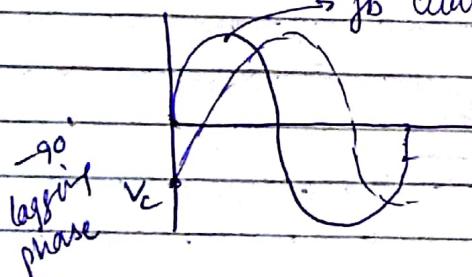
$$i(t) = i_d(t)$$

$$V_c(t) = \frac{1}{C} \int I_m \sin 2\pi ft dt$$

$$= - \frac{I_m}{C} \cos 2\pi ft$$

$$V_c(t) = - V_c \cos 2\pi ft$$

→ if current shows how voltage across capacitor is -ve

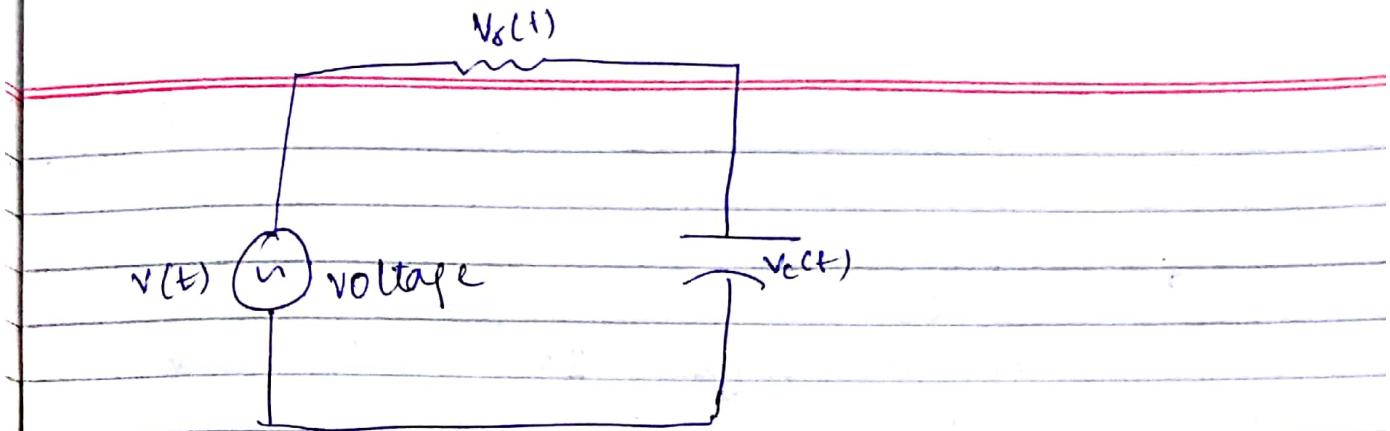


voltage across capacitor is sinusoidal but not same sinusoidal as current.

with different phase.

voltage and current across resistor is sinusoidal  
voltage across capacitor is

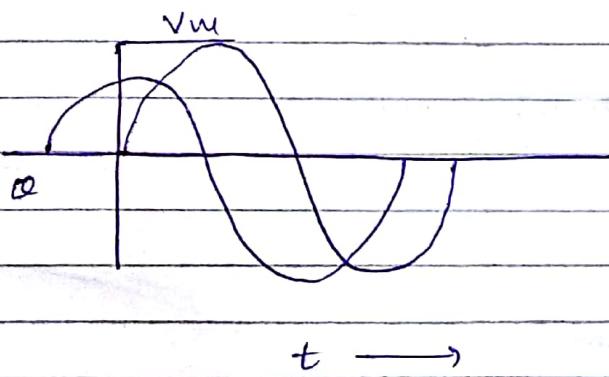
widely  
different  
phase



$$V(t) = V_u(t) + V_c(t)$$

$$\cancel{V_m \sin 2\pi f t} = i(t) \times R + \frac{1}{C} \int i(t) dt$$

$i(t)$  phase will be different



$$i(t) = I_m \sin 2\pi (ft + \phi)$$

$$R^2 + \left(\frac{1}{C}\right)^2 \phi = \tan^{-1} \frac{R}{\frac{1}{C}} = R.C.$$

pure tone / signal  $\rightarrow$  ek hi frequency phase

Medium ki bandwidth  $BW_M \rightarrow 5f$

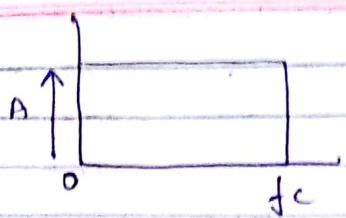
signal ki  $BW_S = 10f$

nhi jaega distortion  
hoga

$BW_S \leq BW_M$

vo parakar jaega

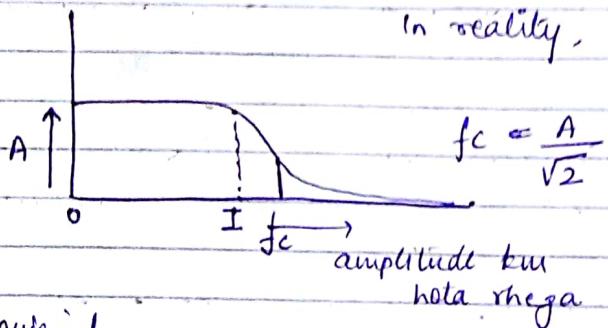
chahiye frequency ka phase kuch bhi ho.



e.g. max frequency that can be passed from the medium is  $f_c$ .

so signal with bandwidth  $\leq f_c$  will be passed without distortion

In reality,



$fc = \frac{A}{\sqrt{2}}$

amplitude kru  
nota rhega

original  
power  
is

$$P = KA^2 \quad \text{for medium}$$

at point

I to power me kuch

degradation ni hega

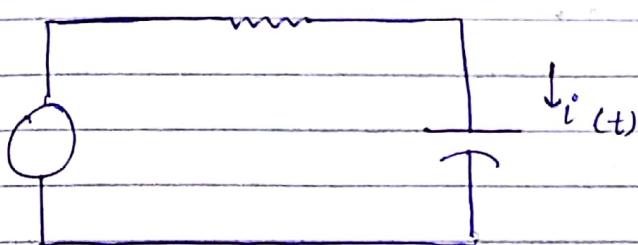
par after I the amplitude will start decrease.

$$k \left( \frac{A}{\sqrt{2}} \right)^2 = \frac{KA^2}{2}$$

half power point

Composite frequency signal - consisting of many frequencies

Behaviour of communication medium is frequency sensitive  
higher frequency - power loss hota hai  
islie communication medium ka ek frequency hota hai use zyada bheja to powerless.



$$i(t) = I_m \sin 2\pi ft$$

$$V_r(t) = \underbrace{R \times I_m \sin 2\pi ft}_{V_m}$$

Voltage se shuru karega

$$\cancel{\frac{V_u(t)}{i_u(t)}} = R \quad ]$$

does not discriminate

frequency

It considers every frequency the same

$$V_c(t) = \frac{1}{C} \int i(t) dt$$

$$= \frac{1}{C} \int I_m \sin 2\pi ft dt$$

$$= - \frac{I_m}{C 2\pi f} \cos 2\pi ft$$

$$\frac{V_c(t)}{i(t)} = -$$

netlab nhi hai phare ki  
waja

$$\frac{V_c(\text{mean})}{I_c(\text{mean})} = \text{Int.} \cdot \frac{1}{2\pi f c} \cdot \frac{1}{I_0}$$

$$\Rightarrow \frac{1}{2\pi f c} = X_C \rightarrow \text{reactance of capacitor}$$

$\frac{V_c(t)}{i_s(t)}$ ,  $R \rightarrow \text{constant}$   
does not change  
with frequency.

note  $\propto X_C$  less current

$$\frac{V_c}{I_c} = X_C$$

$$\frac{V_c}{X_C} = I_c$$

$$X_C \cdot \frac{1}{2\pi f C} \quad T \\ \text{constant}$$

$$\frac{1}{2\pi f c}, X_C$$

$\downarrow$   
increase  $X_C$  decrease

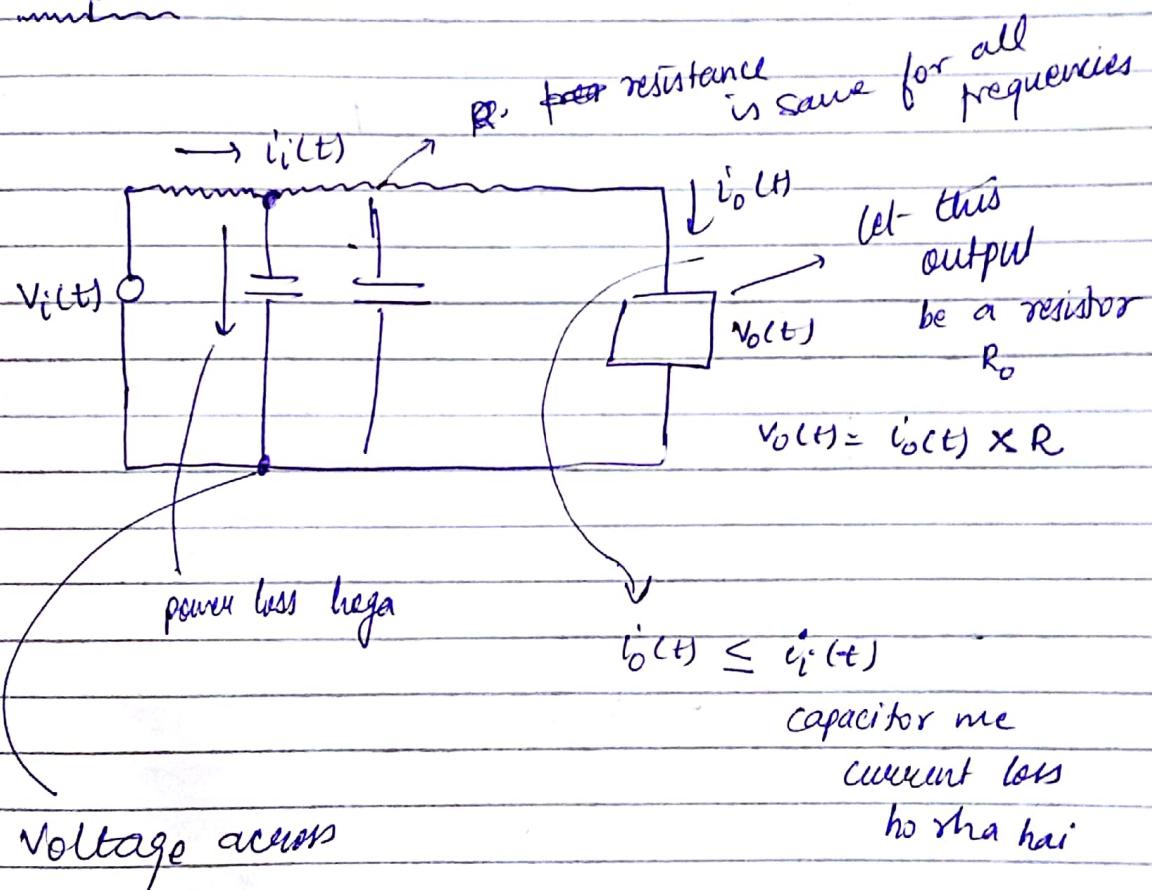
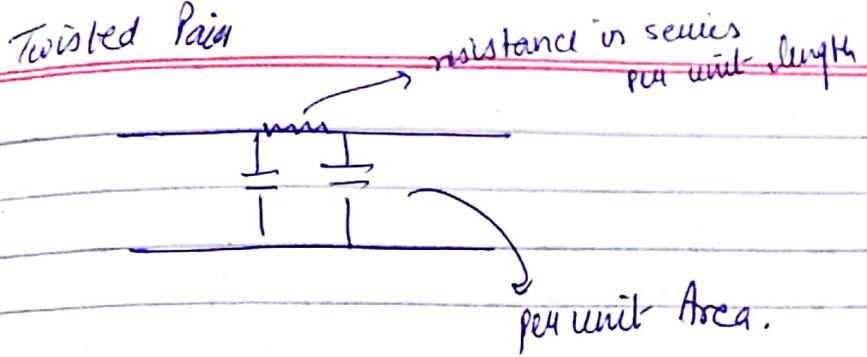
so current increases

$X_C$  will be  
less

$I_c$  will be  
zada

Higher Frequencies ko kaise block kar shai hai

Twisted Pair



$$\frac{V_C}{X_C} = I_C \uparrow \text{leakage current}$$

$$f \uparrow X_C \uparrow$$

different frequencies hai

let

$$i_i(t)_f = 10 \text{ A}$$

$$i_o(t)_f = 8 \text{ A}$$

because of leakage lein wajah se

$$V_o(t)_f = 8 \times R_o \quad \uparrow \text{increasing frequency} \quad \begin{matrix} \text{output} \\ \text{voltage} \end{matrix}$$

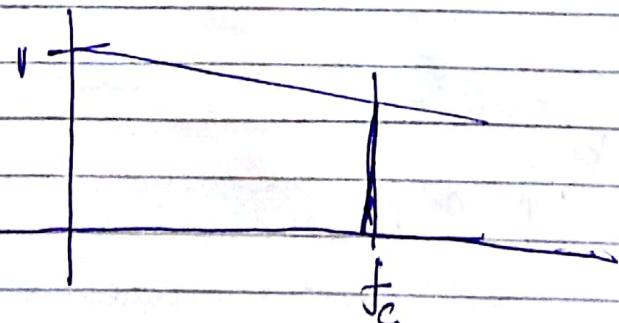
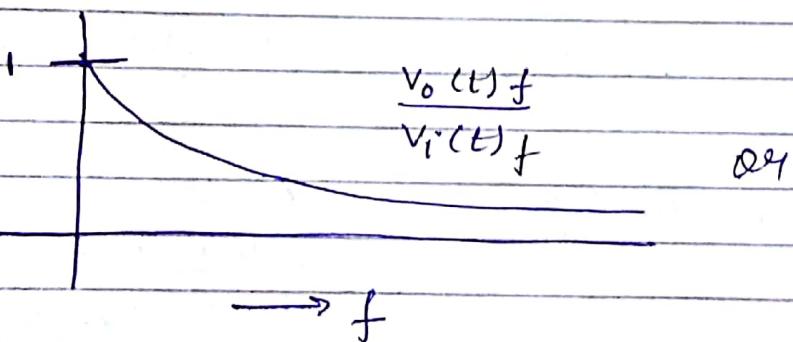
$$V_o(t)_{2f} > 7 \times R_o$$

Gain factor =  $\frac{V_o(t)_f}{V_i(t)_f} < 1$   
 $V_i(t)_f \rightarrow$  constant for  
 all frequencies

0 frequency

$$\Rightarrow \frac{V_o(t)_f}{V_i(t)_f} = 1$$

$\downarrow$   
 capacitance no hoeger.



against  
 capacitance  
 no hoega

to infinite  
 bandwidth

power is half of the  
 original

Cable ki banda - 1 GHz

so signals of frequency  $> 1 \text{ GHz}$

attenuation

ryada hogा

and signal distort-  
hoga.

signal to carry information must change  
laws of superposition

18/1/19

### Magnetic Field

According to the law of magnetic law of attraction

$$\vec{F} = \mu \frac{m_1 m_2}{r^2}$$

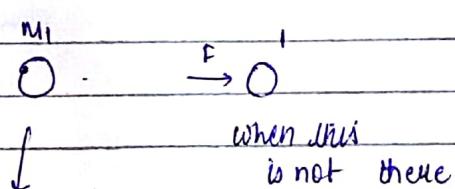
perviability  
of the  
medium

north pole  
strength



$m_1$   $\xleftarrow{F}$  repel kaunje

$$r$$



when this  
is not there

yeh  
koi bhi  
force ~~lga~~ lga पा  
ni laga

unit of pole strength  $m$

$m$  jb yeh 1 point

$\rightarrow$  to  $H$  hogा ins tead of  $P$

$$\vec{H} = \frac{\mu m(t)}{r^2}$$

time varying

$$\vec{h}(t) = \frac{\mu m(t)}{r^2}$$

$m(t)$  varying in sinusoid manner  
others are constant

$$m(t) = M \sin 2\pi f(t)$$

so

$n(t)$  is varying

$$\vec{E} = \frac{1}{\epsilon_0 \mu_0} \vec{Q}$$

time varying

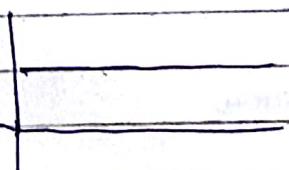
$$\vec{E}(t) = \frac{1}{\epsilon_0 \mu_0} \vec{Q}(t)$$

(to yeh bhi)

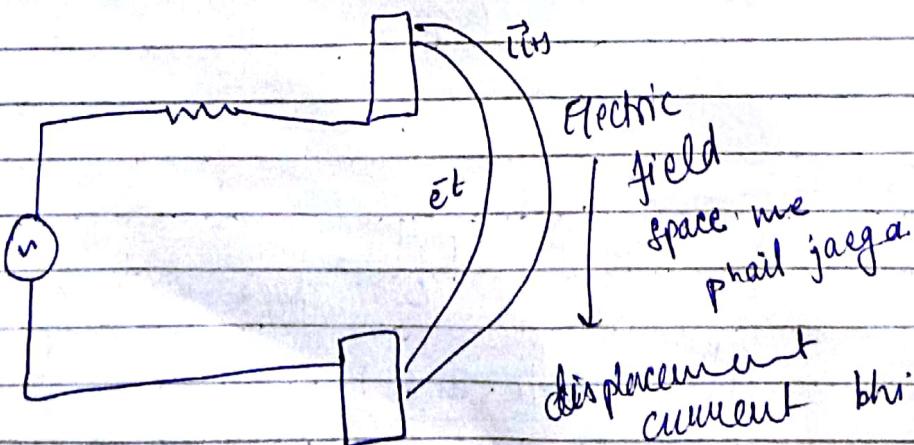
$$q(t) = Q \sin 2\pi f t$$

yeh sinusoid

DC frequency 0  
~~time~~  
change ni  
hota

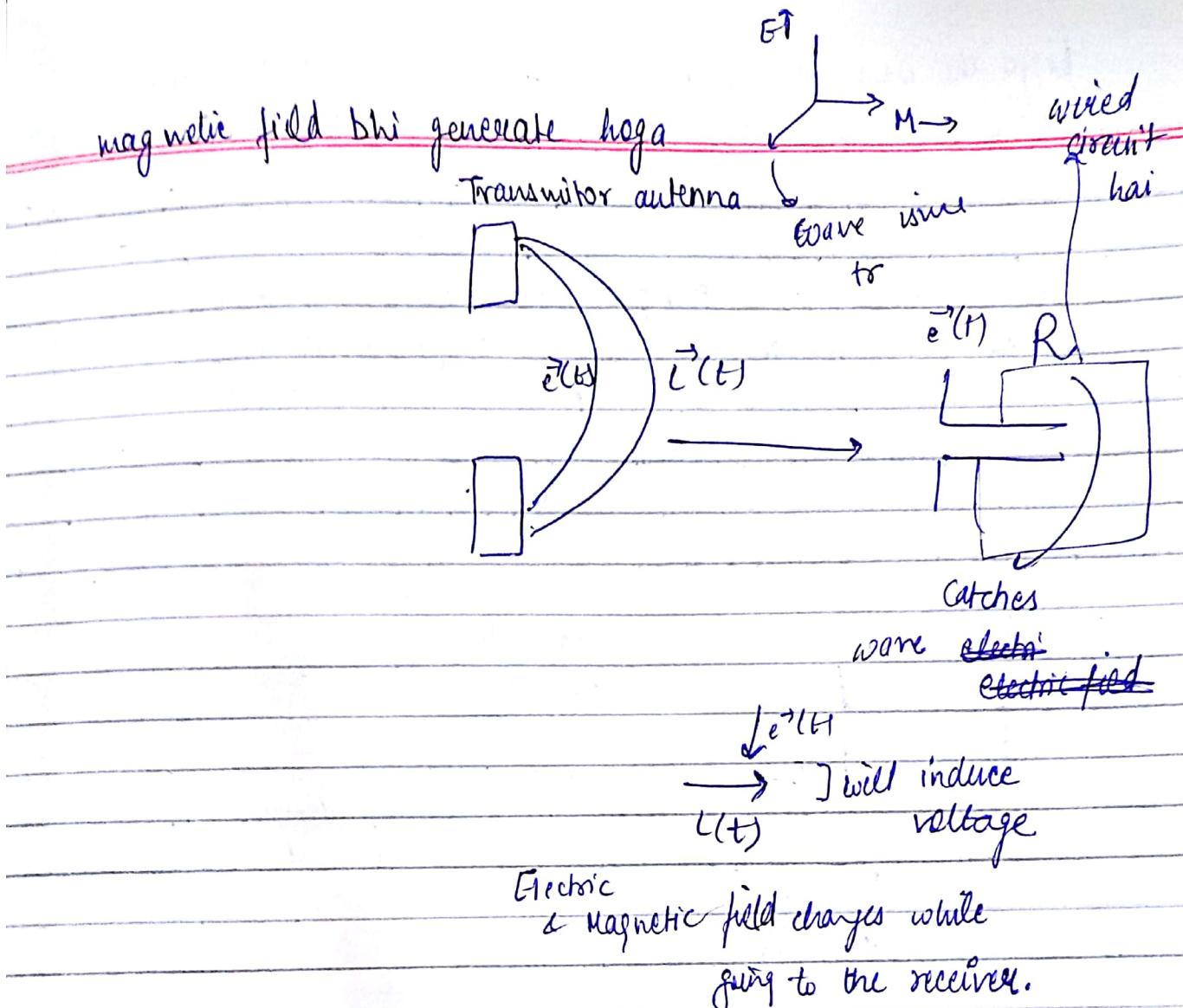


$$f = \frac{1}{T} = \frac{1}{\infty}$$

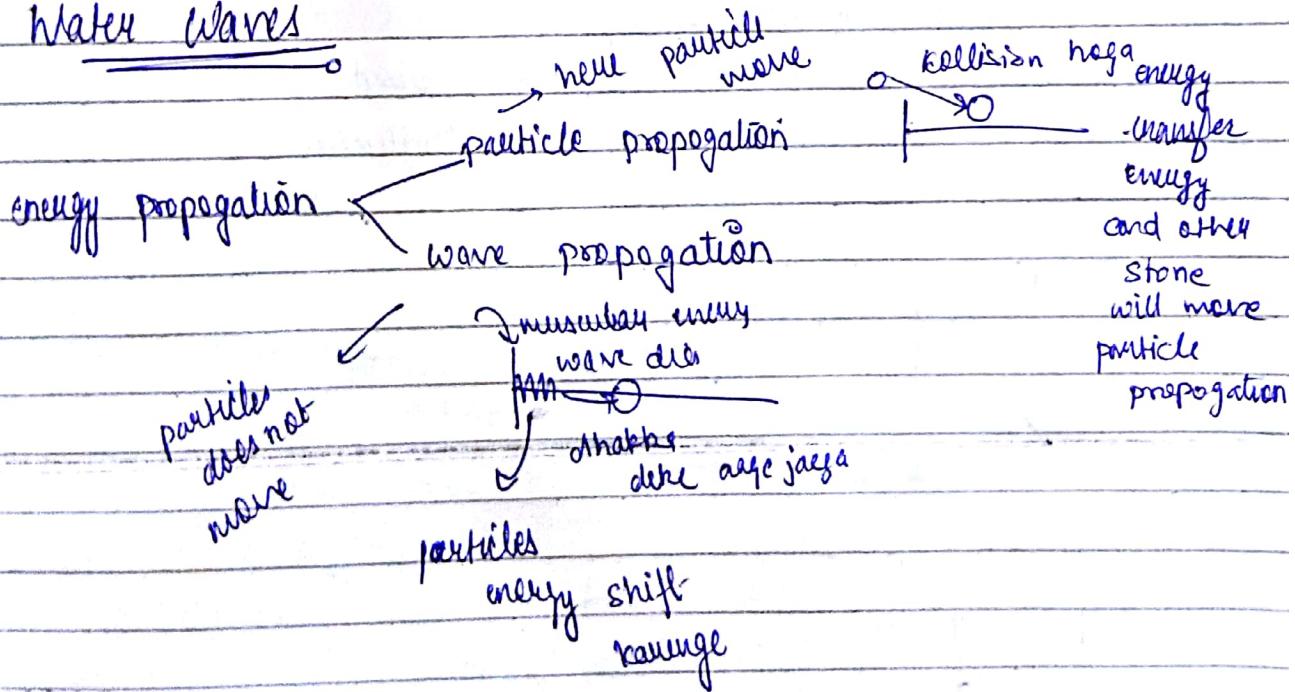


$$\vec{E}(t) = E \sin 2\pi f t$$

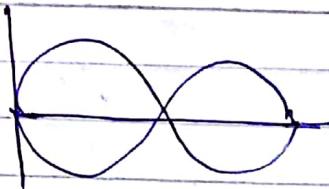
$$i_d(t) = I \sin 2\pi f t$$



## Water Waves

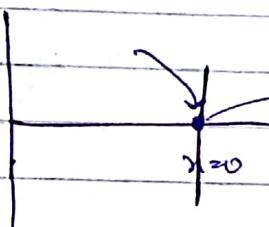


bigger the mass less wave nature more particle nature  
smaller the mass more wave nature less particle nature.

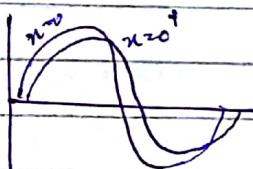


2 light with same frequency  
different phase  
produces darkness.

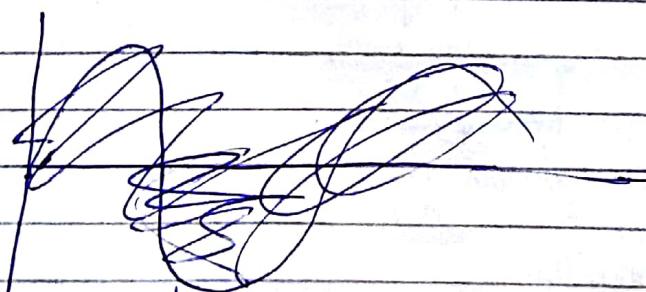
dual nature of light.



water particles  
oscillate  
between  
apne neighbours  
less energy dense

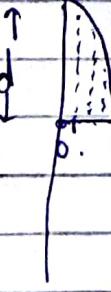


particle is not moving  
only energy is transferred.  
and particles start oscillating

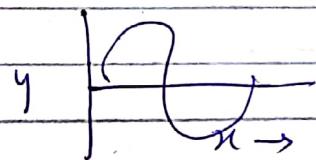
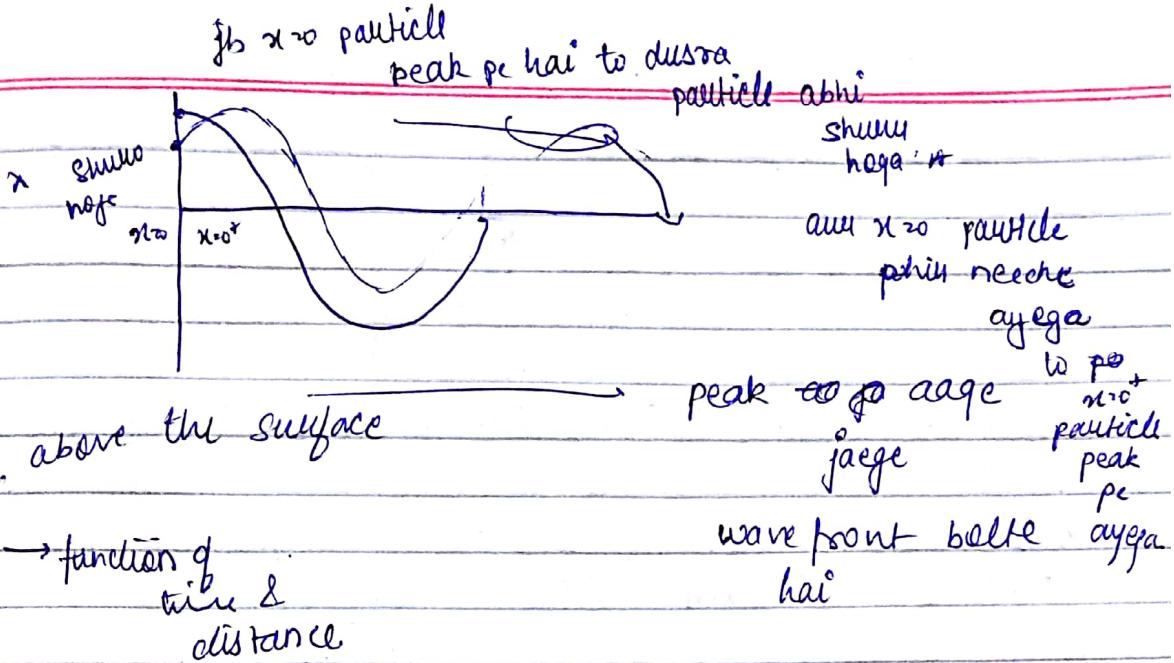


$t = t_1$

$\uparrow$



$x$



looking y with x will also be sin wave



looking y with t will also be sin wave  
↑ man peak length

$$y(t, x) = Y \sin(2\pi ft - \beta x)$$

when  $x=0$

$$Y \sin(2\pi ft)$$

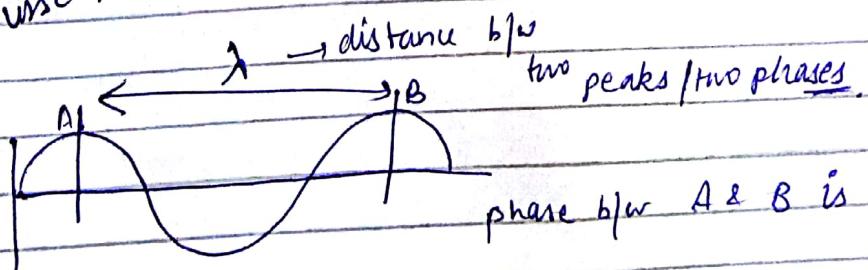
phase change by unit distance



$$x_2 - x_1$$

$$Y \sin(2\pi ft - \beta x_1)$$

one phase change हुआ  
use  $\beta$  change



$\lambda$  distance pe  $2\pi$  phase change

1

$$\frac{2\pi}{\lambda}$$

$$\therefore \beta = \frac{2\pi}{\lambda}$$

$$y(t, x) = Y \sin \left( 2\pi f t - \frac{2\pi}{\lambda} x \right)$$

$$\vec{e}(t, x) = E_m \sin \left( 2\pi f t - \frac{2\pi}{\lambda} x \right)$$

$$\vec{h}(t, x) = H \sin \left( 2\pi f t - \frac{2\pi}{\lambda} x \right)$$

antenna se field varying hai

$$\vec{E}(t, x) = E_m \sin \left( 2\pi f t - \frac{2\pi}{\lambda} x \right)$$

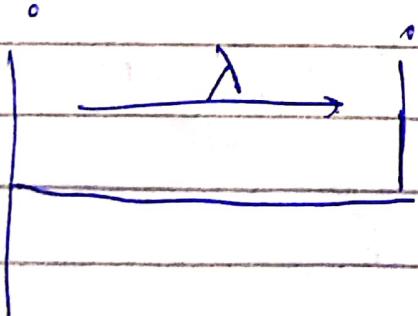
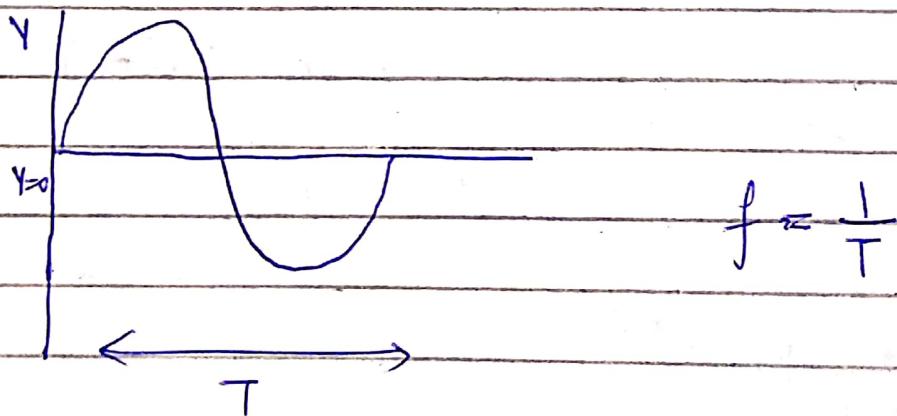
$$\vec{h}(t, x) = H \sin \left( 2\pi f t - \frac{2\pi}{\lambda} x \right)$$

antenna se field varying hai

29/1/18

Function of  $t$   $\rightarrow$  water particle movements

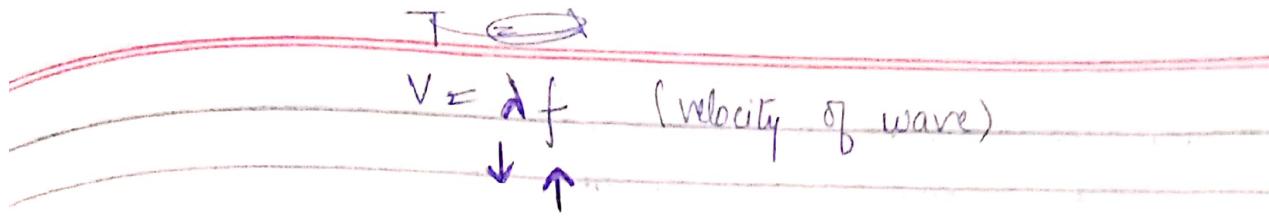
$$y(t, x) = Y \sin \left( 2\pi f t + \frac{2\pi}{\lambda} x \right)$$



$T_{sec} = \lambda m$  (distance)

$$f_{sec} = \frac{\lambda}{T}$$

$$f = \frac{1}{T} = \lambda$$



### Electro Magnetic Field

$$\vec{E}(x,t) = E \sin\left(2\pi ft + \frac{2\pi}{\lambda} x\right)$$

magnitude is changing ~~sus~~ said

$$\vec{H}(x,t) = H \sin\left(2\pi ft + \frac{2\pi}{\lambda} x\right)$$

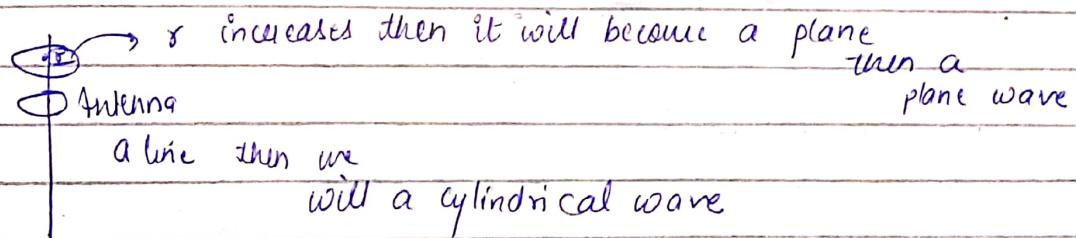
Parallel points are in a circle (phase to join same pe)

Wave front will form a circle

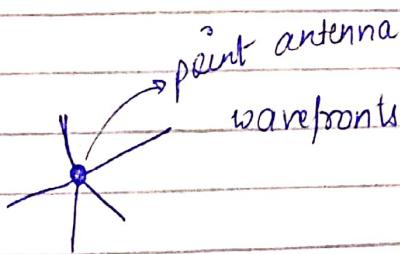
If the distance increases then the wavefront will become a line.



That will be a line wave or a plane wave.



If antenna



point antenna

wavefronts will be spherical

so spherical wave.

distance ~~of~~ away from antenna ~~plane~~ wave is coming as a plane

$$V = f \lambda$$

Signal → whatever we are passing through a medium  
(space)

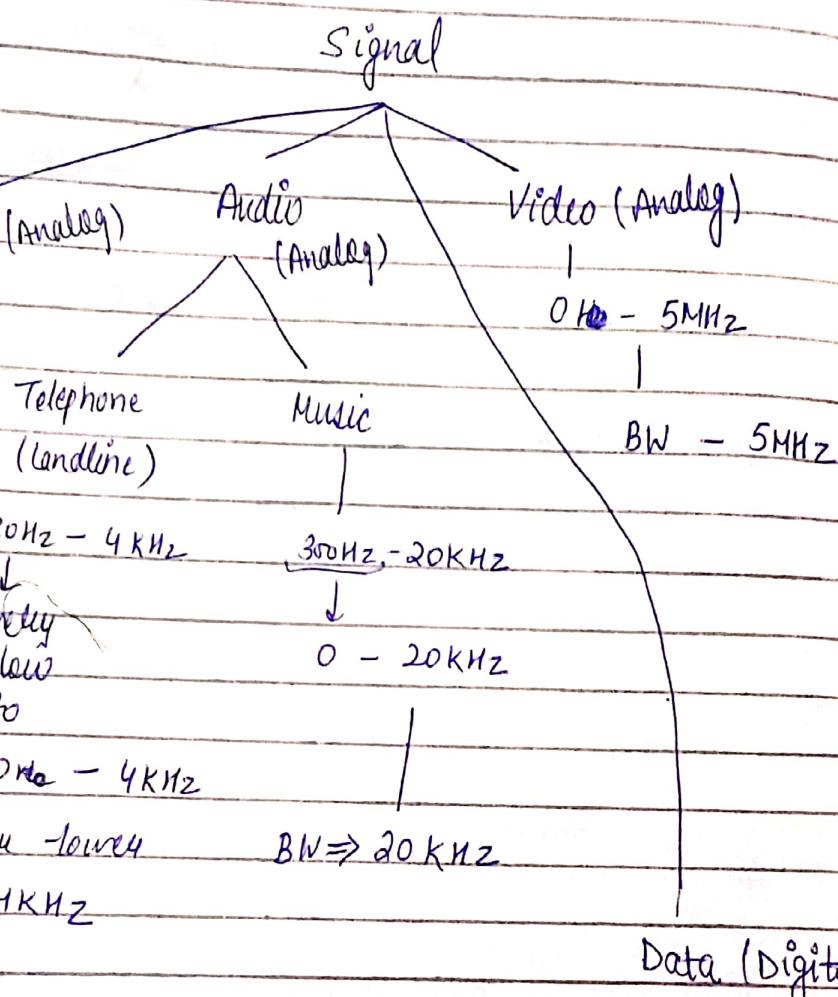
Audio  
Human produced  
by ear/nun

100 kHz range  
Audio + Video  
20 kHz + 5 MHz

~~(5.22 MHz)~~  
(5.02 MHz)

Simplicity  
(5.1 MHz)

$$BW = \text{upper - lower} \\ = 4 \text{ kHz}$$



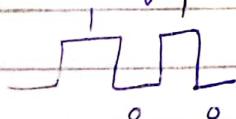
Signal

Analog → varying continuous

Digital

(discrete)

Digital



$$\lim_{t \rightarrow a^-} s(t) = s(a) = \lim_{t \rightarrow a^+} s(t)$$

continuous at point  $a$

if  $s(t)$  is continuous

at all  $a$ 's

↳ analog signal

(many sinusoidal signals added together)

Signal function of t

then  $f(t) = f(t+T) \forall t$

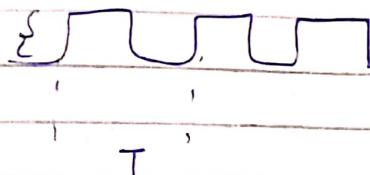
then it is periodic signal

$\nexists f(t+T) \rightarrow$  non periodic

Fourier series

Any periodic (analog / digital) signal  
then

$$g(t) = C + \sum_{n=1}^{\infty} A \sin 2\pi nft + \sum_{n=1}^{\infty} B \cos 2\pi nft$$



$$g(t) = 0$$

$$0 < t \leq T/2$$

$$f = \frac{t}{T}$$

$$= 5$$

$$T/2 < t \leq T$$

$$g(t) = \frac{5}{2} + k \sin 2\pi ft + \frac{k}{3} \sin 2\pi 3ft$$

$$BW = \omega - 0 = \infty$$

We cannot such medium with high bandwidth

~~Signal of high frequency distort hoga~~  
but amplitude reduce ho rha hai to power km hoga.

signal bheja distort hu hogga to that

Digital data not m

K.

$\frac{K}{3}$

$\frac{K}{n} \uparrow$  increase

frequency  
ka mtlab  
ni rahega

$$BW_e = 1/f = 0$$

$$= 1/f.$$

$$9f \leq 1/f \Rightarrow f$$

DC - amplitude  
of 0 frequency

pure tone - 1 frequency

1 frequency signal

BW = 0

cavvien signal

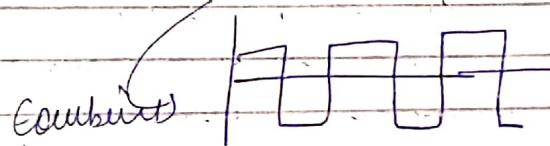
low pass signals

when all radio

(hoga so)

cannot send directly

f, 3f, 5f



periodic  
wave ko  
divide karun  
frequency

to they will come from  
f, 3f, 5f

formule

signal

s(t) → Mod

kr Since 0 frequency nhi hai to ulta signal the side  
ke pe jaega to ulta negative side

Breaks the signal into sin components to frequency spectrum  
pta lagega, bandwidth

any signal

bb within

sine wave

$f_c >$

Bandwidth jini logi almost utni frequency the signal  
bhej pate hai.

on signal p

series

Composite periodic → using formula, we can find the frequency  
of components

Non-Periodic → We cannot calculate that frequencies (hit and trial)

Scanned by CamScanner

signal bheja distort hua to bandwidth increase kia agar pass hogya to that is the bandwidth.

Digital data not natural but we are passing more bandwidth.

carrier signal  $\rightarrow$  technique by which we send our original signal.

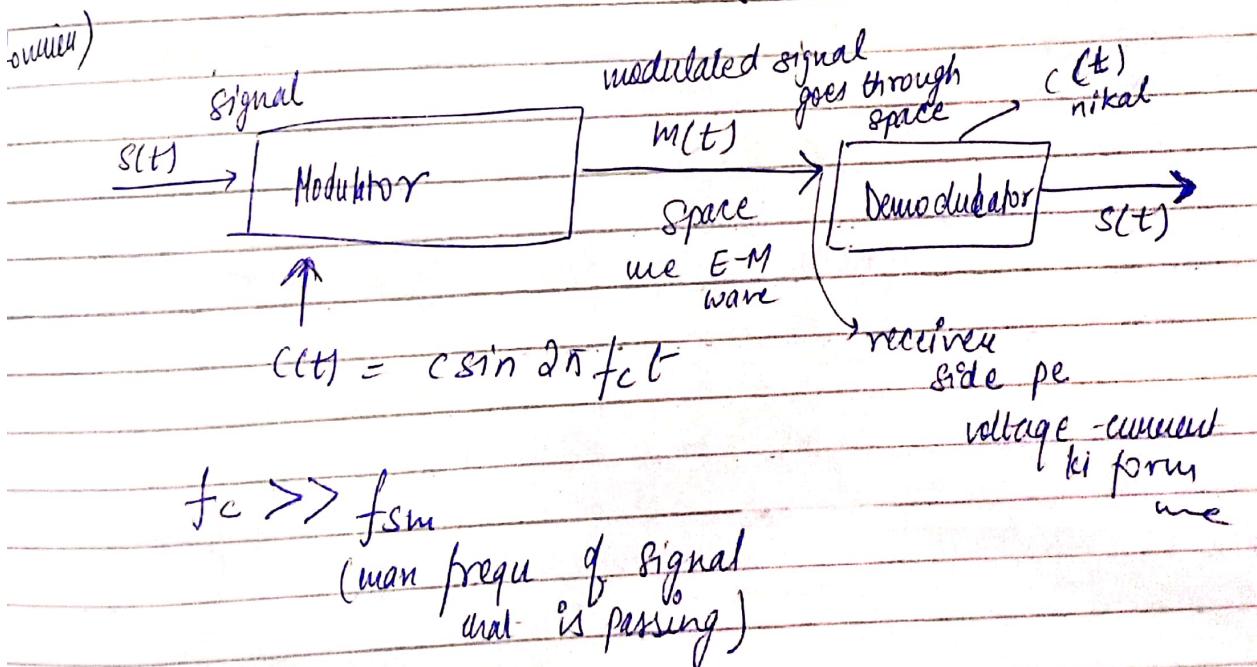
low pass signals.

when all radio signals are passed together to distribution (hoga so I need to send these signals separately)

cannot send directly

to send through a carrier signal  
(modulate it)

and at receiver  
(demodulate it)



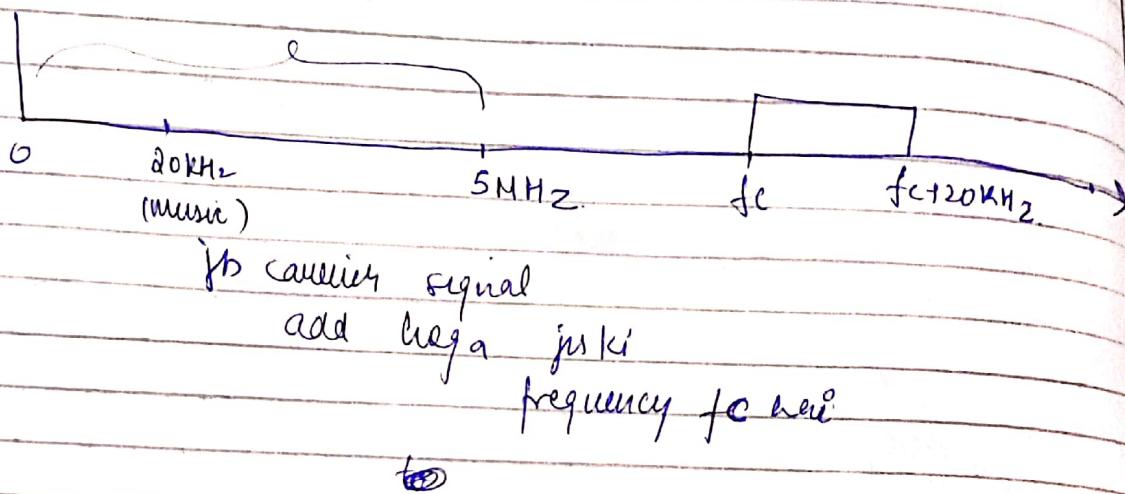
$$f_c > f_{sm}$$

(man freq of signal  
that is passing)

carrier signals generated through oscillator

What all signals pass through space

Natural signals



Medium Wave - 525 kHz - 1705 kHz  
(local radio) (near the surface travel hata hai)

Short wave  $\Rightarrow$  3900 kHz - 7450 kHz  
ionosphere reflect hata hai (for more region cover)

High Frequency Wave - 32-30 MHz

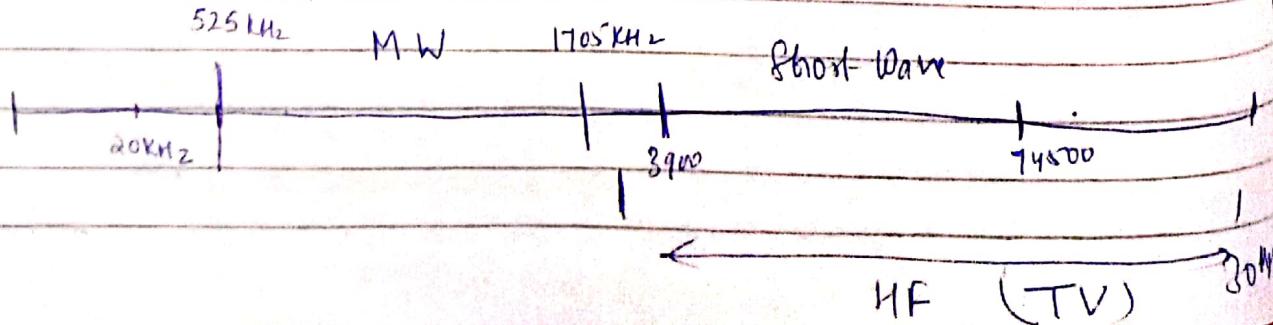
FM radio  $f_{90} - 108 \text{ MHz}$

VHF (TV)

(radio)  
300 MHz

(Wi-Fi)

(Satellite  
Comm)



1 GHz - 3G/4G

3G spectrum.  
4G spectrum.

VHF

- 30 - 300 MHz

UHF

- 300 MHz - 3 GHz

(3000 MHz)

→ (890 - 960 MHz) - 2G comm.

GSM mobile communication

→ 1800 MHz - 1900 MHz (Range check ?)

3G/4G

2 GHz and onwards is a microwave

(wireless (WIFI)  
LAN)

✓ waves in line of sight

2.4 GHz

ray

sharp transmission beam

2.4 + 0.80

(Bandwidth)

gas over

I  
microwave  
heat-

(microwave

oven will operate in)

atom heat  
uniformly  
all over

(satellite communication  
bhi hota hai°)

carrier frequency - 40 GHz

th generate kau  
pae hai°.

FM)

MHz L Band - 1-2 GHz

(WiFi) S Band - 2-4

(satellite C Band - 4-8

communication) X - use ni

Ku - satellite

ITU

E.

Optical Fibre -  $10 \text{ GHz}^{8.6}$   
open 100 layers

International Telephone Union  
↳ so that no interference  
↳ country to band data hai.  
Government  $5 \times 10^6$   
User divide  $1000 \times 10^6$   
bandwidth to  $5 \times 10^6$   
ISP.

FDM

Frequency Division

Multiplexing if audio is

200 channels  
TV

$1000 \times 10^6 \times 10$

$100 \times 10^5$

$10^7$  channels for  
radio.

International

— FDM

Nation

— FDM

ISP

— FDM

(within this TDM.

Geostationary  
satellites.

for satellite  
communication  
beam should  
be very sharp

because on reaching satellite  
it may spread and  
not reach properly

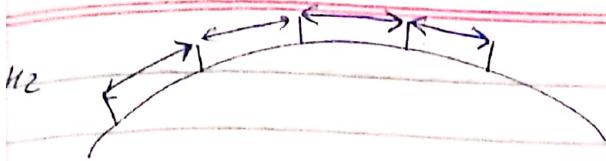
2 types of line of sight microwave

(3 GHz<sup>+</sup>)

Telephone

terrestrial

satellite



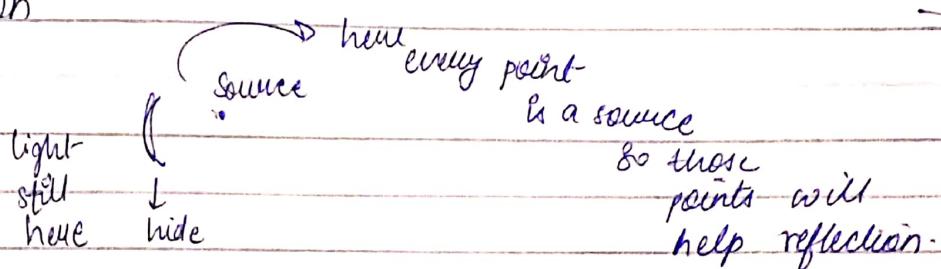
## Microwave Propagation Characteristics

- line of sight communication
- reflection (metal)
- refraction
- interference

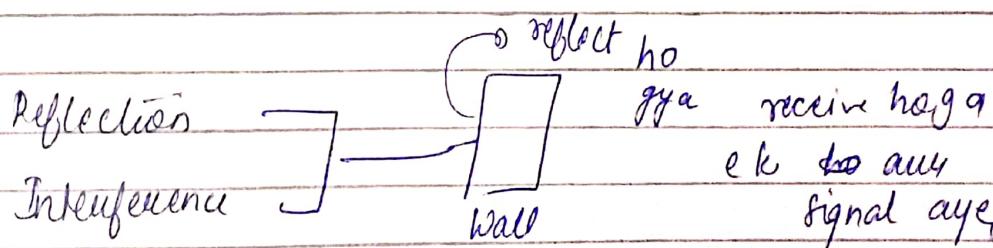
(Two way in opp phase → o signal strength)

" " same phase → double strength

### Diffraction



→ Absorbed by Rain, Fog, Trees



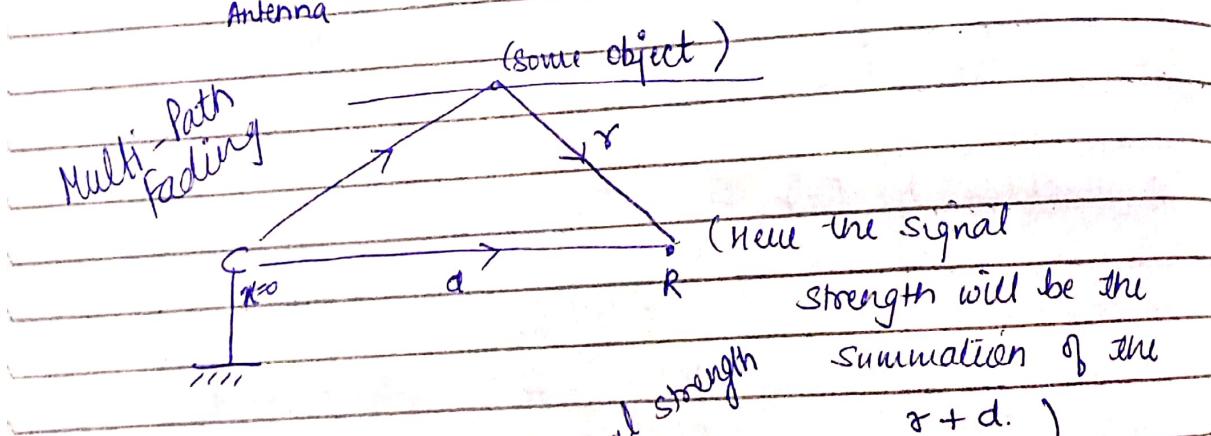
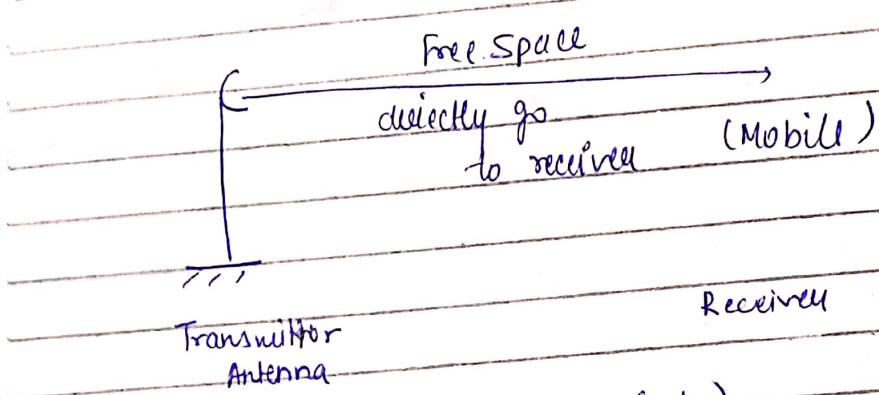
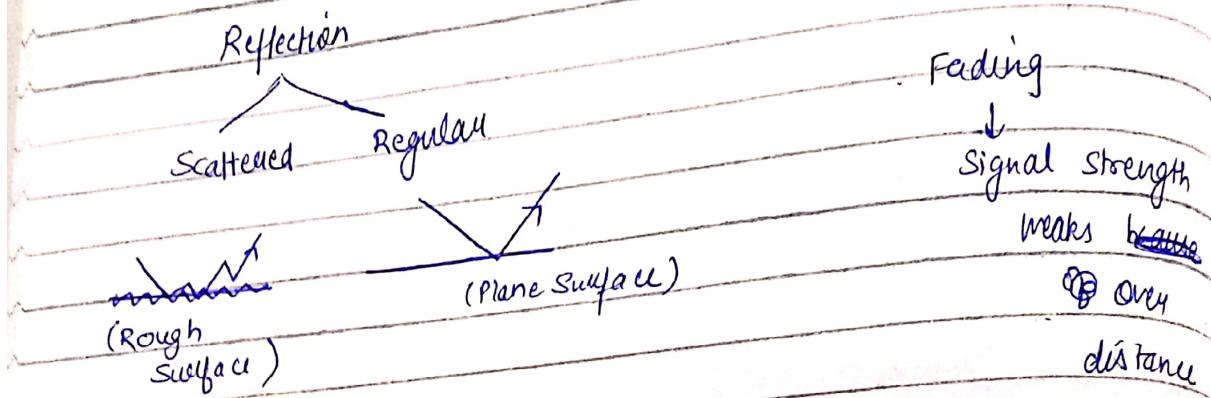
fading → Multi-path fading → if same phase add nhi to 0.

Distance fading

(Distance increase

signal weaken.)

1/2/19



$$e_d(t, d) = E \sin\left(2\pi f t + \frac{2\pi}{\lambda} \cdot d\right)$$

$$e_u(t, d) = E \sin\left(2\pi f t + \frac{2\pi}{\lambda} \cdot x\right)$$

$$\bar{e}_e = e_d + e_u$$

(e effective)

$$u - d = \lambda \rightarrow u = \lambda + d$$

$$\bar{e}_e(\lambda) = \vec{e}_d + \vec{e}_u = E \left( \sin 2\pi ft + \frac{2\pi}{\lambda} \cdot d \right)$$

$$+ \sin \left( 2\pi ft + \frac{2\pi}{\lambda} \cdot (d + \lambda) \right)$$

$$= E \left( \sin \left( 2\pi ft + \frac{2\pi}{\lambda} \cdot d \right) + \right.$$

$$\left. \sin \left( 2\pi ft + \frac{2\pi}{\lambda} d + 2\pi \right) \right)$$

$$\frac{\sin(2\pi)}{2}$$

$$= E \left( \sin \left( 2\pi ft + \frac{2\pi}{\lambda} d \right) + \sin \left( 2\pi ft + \frac{2\pi}{\lambda} d \right) \right)$$

\*  $2\pi$  phase difference gives the same wave

$$= 2E \sin \left( 2\pi ft + \frac{2\pi}{\lambda} d \right)$$

$$\bar{e}_e(\lambda_2) = 2E \left( \sin \left( 2\pi ft + 0 \right) \right)$$

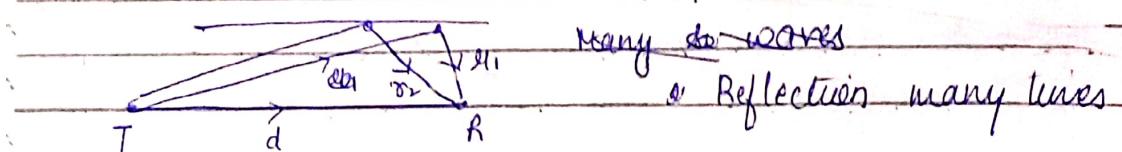
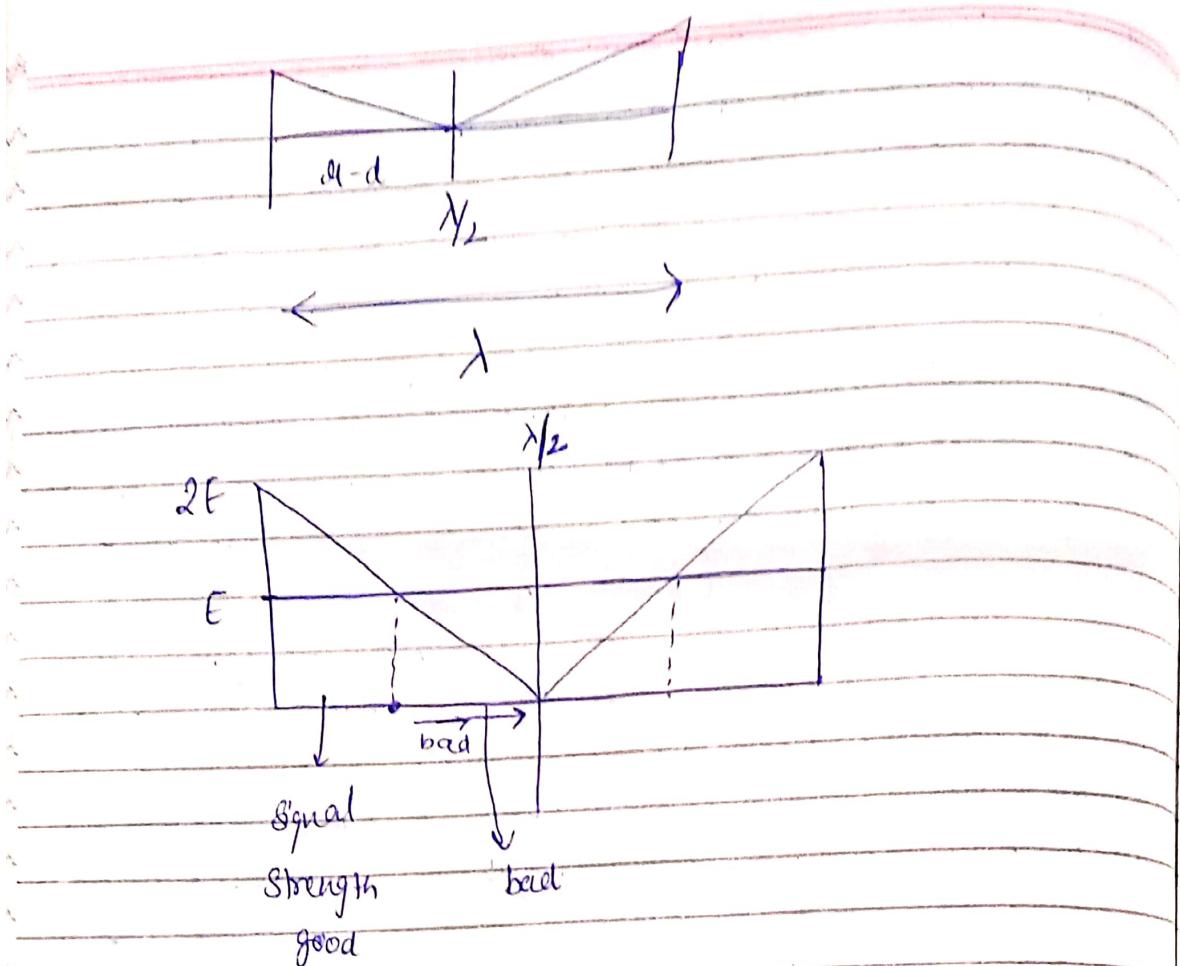
If two waves have same phase difference of  $1/2$   
so signal strength will be 0



Strength double  
↑

Strength 0

Phase difference between  $0 \leq r-d < \lambda/2$



$$\vec{e}_d \text{ and } \vec{e}_{\theta_1} \text{ and } \vec{e}_{\theta_2}$$

So

$$\mu_1 = \theta_1 - d = \lambda/2$$

$$\mu_2 = \theta_2 - d = \lambda$$

$$e^*(\lambda) = \vec{e}_d + \vec{e}_{\theta_1} + \vec{e}_{\theta_2} \quad (\text{This is summed graphically})$$

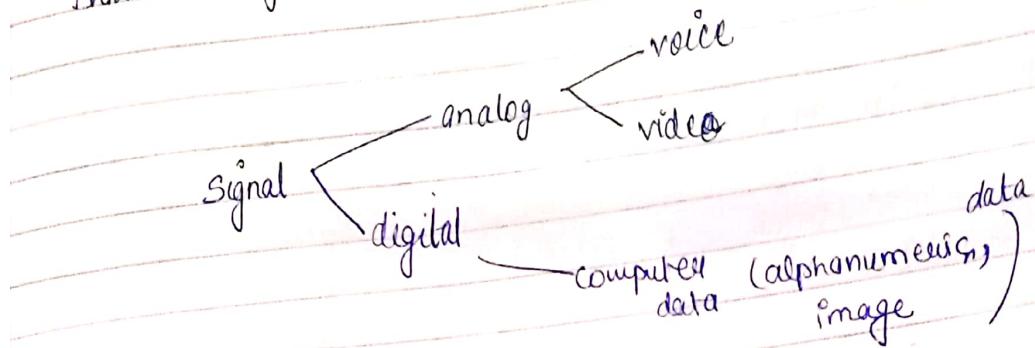
$$e = e_d + e_{\theta_1} = 0 \quad \text{beacause}$$

$$e_d = e + e_{\theta_2} = 0 \quad \text{calculation becomes difficult}$$

(Signal strength becomes same again)

Underground buildings receives less signal because of multi-path fading.

Intrinsic digital signal (coming natural)



Voice is 20 kHz (0 - 20 kHz)

signal is modulated eq. with 500 KHz

then modulated

Signal will become 520 - 520 KHz.

Three types of modulation

(amplitude modulation)  
AM

$S_1$  ————— 0 - 20 kHz  $f_C$  500 KHz ————— (500 - 520 KHz)

$S_2$  —————  $f_C$  525 KHz ————— (525 KHz - 545 KHz)

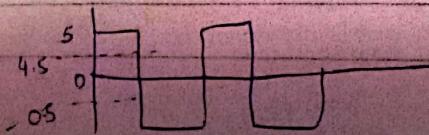
Problems of Analog Communication

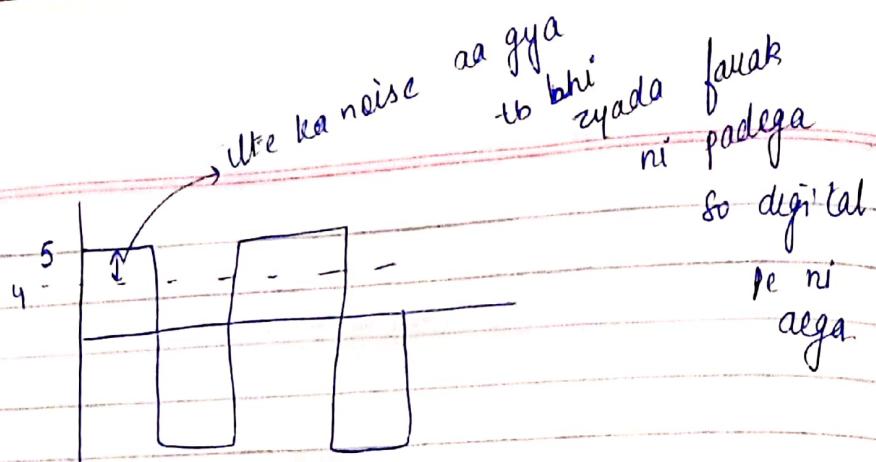
→ Modulated signal combine with noise

when demodulating the signal is not the original one.

(digital signal pe ni hota zyada)

→





If digital data gets errored then error recovery mechanism is  
(check and detect)

Data Encryption

Digital signal can be receipt and amplify it  
this is digital regeneration

noise filter

repeater/ does this  
regenerator

amplifier

Since digital data can be stored and processed we can do

Error Detection - CRC

Correction - Hamming Code (transmission  
we usually  
ni hola)

detect kia

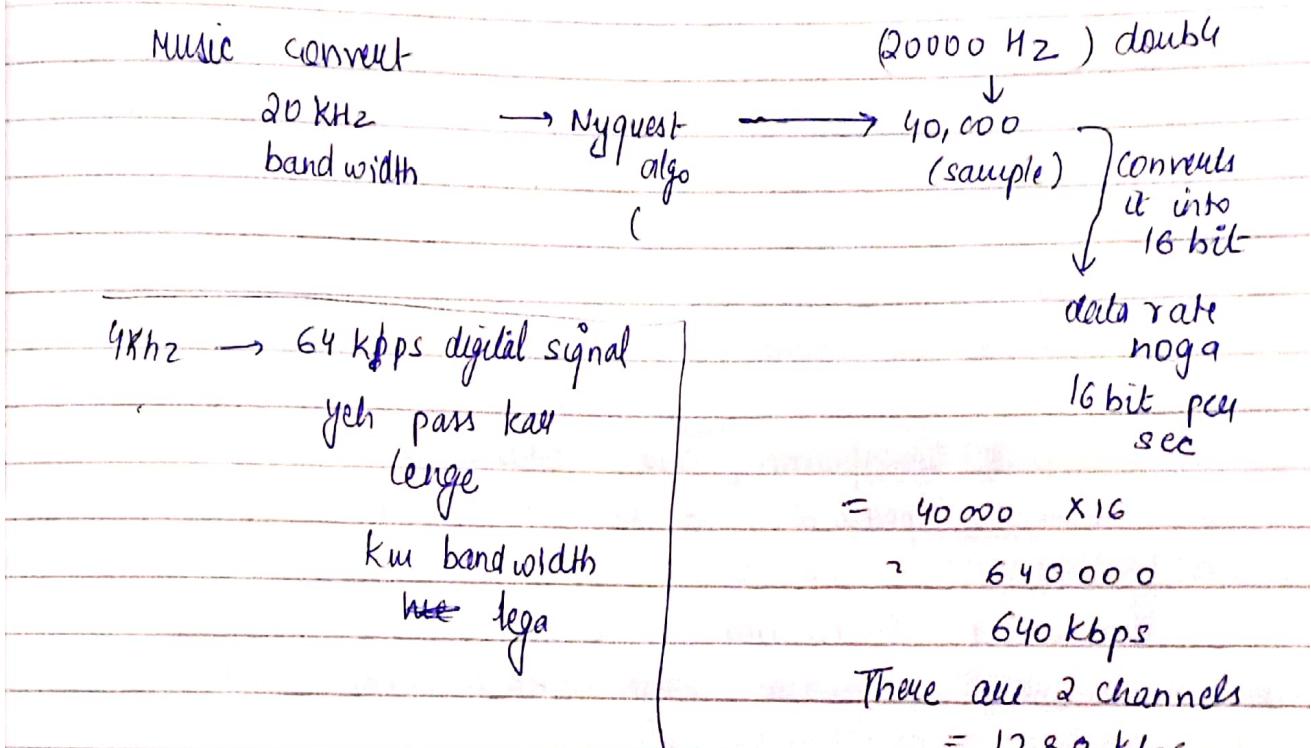
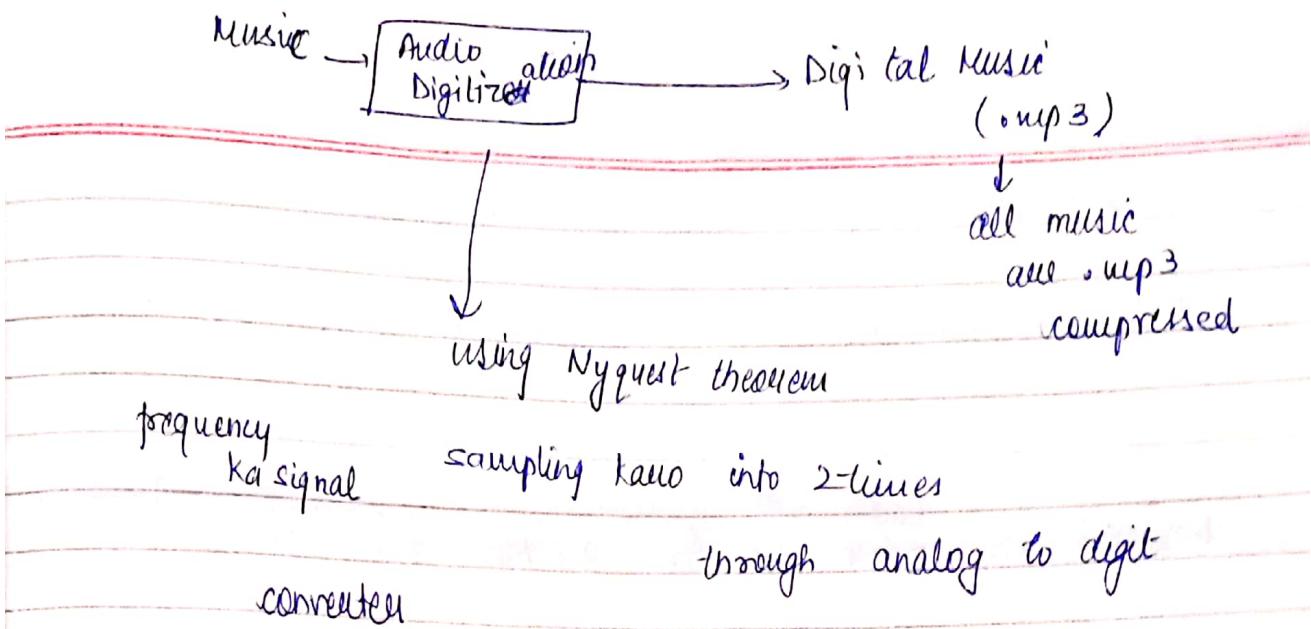
Digital data can also be  
compressed & decompressed  
& recover  
using bandwidth kia.

Video Digitization

~~convert~~ video into digital signal  
(video)

Video  
Digitizer

digital video



## Video Compression

### Video format

Suppose -  $1024 \times 768$

↓      ↓  
points lines  
in each line

to compress ~~baaga~~ kauenga i.e. using .mp3 compressor

Resolution → no. of pixels per frame

We will get three analog signals for 1 pixel.  
red, blue and green

Red decoded into 8 bit values will be 0-256  
11<sup>say</sup> for Blue & Green

1 pixel ko ~~color~~, ke lie 24 bits use hue  
Code K me  $\rightarrow$  2<sup>24</sup> colors can be displayed  
11<sup>say</sup>

So ~~cost~~ for  $1024 \times 768$   
 $= 1024 \times 768 \times 24$  bits  
will be required to store the frame

### Optical Illusion

If two frames come within a period.

Cannot be separated (indistinguishable),  
If two frames, <sup>does not</sup> come ~~within~~ & within that period can be separated

per second 50 frames generated  
so each frame come within  $\frac{1}{50}$  second

so they will be indistinguishable due to persistence of vision.

no ~~frames~~ of frames per second  
 $1024 \times 768 \times 24 \times 50$  bps

943718400 bps

921600 Kbps

9035 ~~Kbps~~ Mbps

Divide by 64 Kbps that many channels of voice will be required.

## Video Compression & Decompression

Compression

↳ If video is fast first frame is send, then diff of second & first frame is send and so... on  
(bits)

Lossy Decompression

First frame is received then second frame added and so... on.

Lossy Compression / Lossless Compression

↓                    ↓  
Diff b/w          No diff in ~~compression~~ sending & receiving  
sending & receiving picture

picture (time distortion  
hoga signal ka)

video

↳ has  
variable  
bandwidth

Digital bandwidth expressed in bps, kbps,  
mbps

Analog bandwidth in Hertz

analog

digital data passed through a medium requires bandwidth

↓  
bandwidth zyada then analog bandwidth bhi  
zyada nilega.

Mobile voice me analog bandwidth nhi dia jata, <sup>zyada</sup>  
lossy compression hota hai uske signal & loss  
hota hai.

whereas

landline voice me compression ni hota

→ goes through space then also signal gets distorted bcoz of noise.

Lightning does not affect the signal bcoz lightning sound frequency is very high.

But motor of industries generates high voltage spikes containing all type of frequencies if met with a signal to . Signal gets distorted

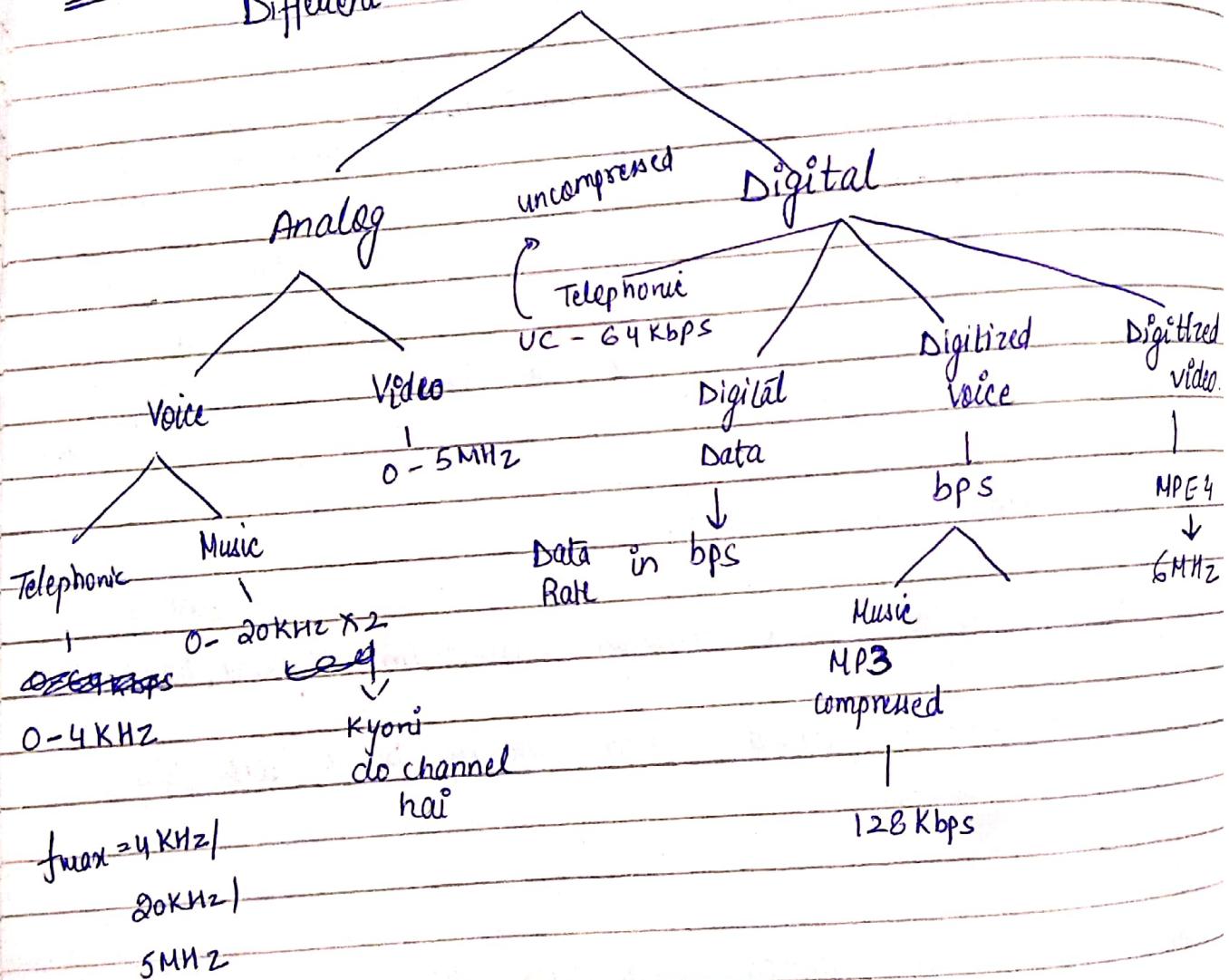
2i

~~Lightning does not affect us, sound frequency is very high.~~

But motor of industries generates spikes (containing all type of frequencies) if met with a signal to . Signal gets disturbed

21  
6/1/19.

## Different Communication



Modulation done using  $f_{\text{max}}$

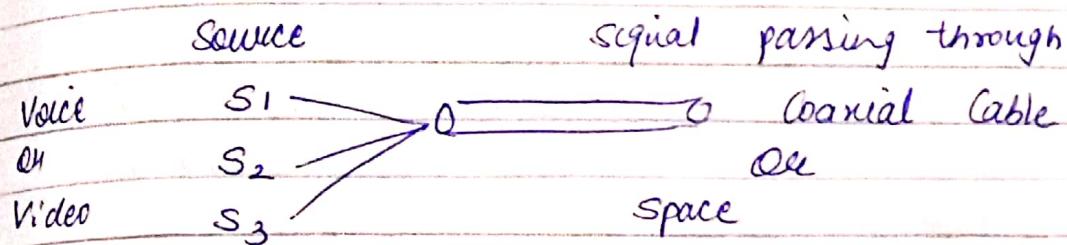
$$BW_s = f_{\text{max}} - f_{\text{min}}$$

In analog telephonic

$$\begin{aligned} BW_s &= 4 \text{ kHz} - 0 \\ &= 4 \text{ kHz} \end{aligned}$$

Transmit Data into space

low pass  $f_{max}$  =  $0-f_{max}$



multiple sources  
sends signals together (Interference hog)  
collision hog.

0                  4 GHz  
0

Bandwidth of coaxial cable

and for space

0 - 40 GHz  
frequency

40 GHz frequency signal is generated.

S<sub>1</sub> sends signals occupy 0-f<sub>max</sub>

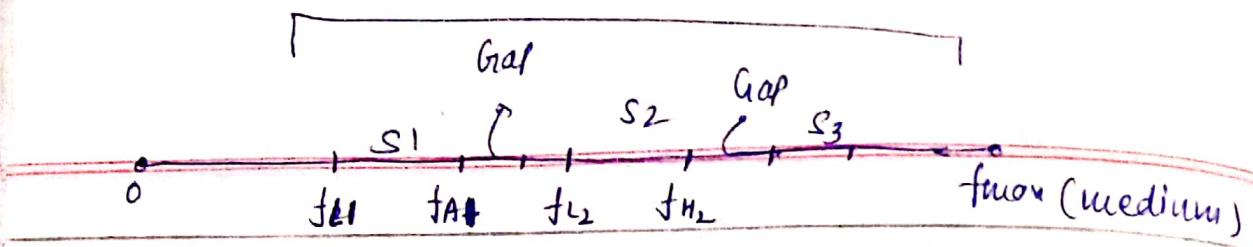
S<sub>2</sub>                  0 f<sub>max</sub>

S<sub>3</sub>                  0 f<sub>max</sub>

all signals will occupy only this bandwidth (max frequency of f<sub>max</sub> (medium) medium)

rest will go waste.

→ Modulation



Bandpass signal

↳ non-zero low frequency and  
non-zero high frequency

channel is low pass

but within it will be band pass  
with  $f_L$  and  $f_H$

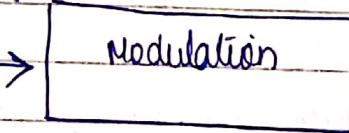
composite signal

low  
frequency

high  
frequency

(Analog) → this is  $f_{max}$

$$s(t) = S \sin 2\pi f t \quad (\text{Bandwidth} = 0 - f_{max})$$



CARRIER

frequency

$$c(t) = C \sin (2\pi f_c t + \phi_c) \quad \begin{matrix} \rightarrow \text{phase of} \\ \text{carrier} \end{matrix}$$

Amplitude  
of carrier

carrier  
frequency

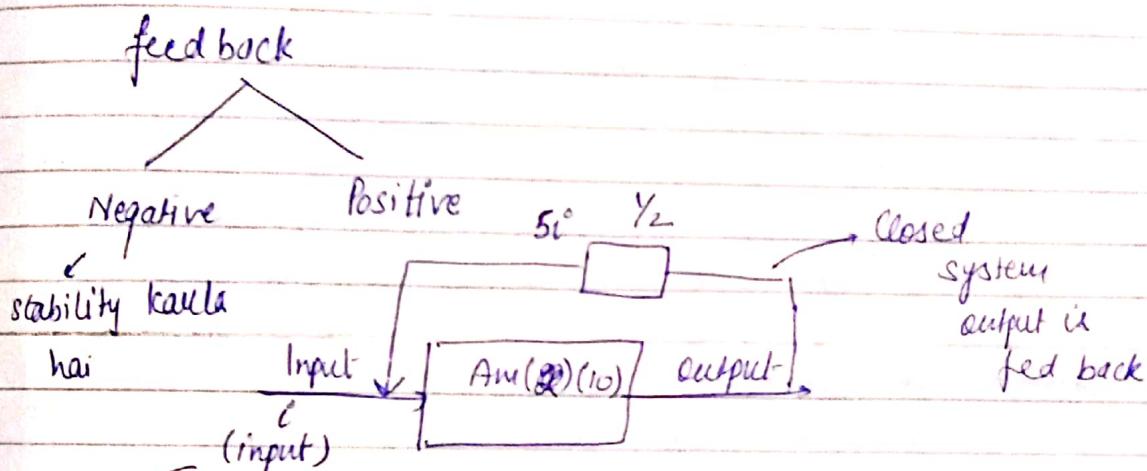
Noise is bad as signal

↳ but bcoz of this we are able to create oscillator and high frequency signal

Oscillator + feedback

Tuning frequency depend of Inductor & conductor in oscillator

$$\text{Tuned frequency } f = \frac{1}{2\pi} \frac{1}{\sqrt{LC}}$$



If input amplified  $\rightarrow$  amplifier

If input decreased  $\rightarrow$  decipitor

Amplitude effect =  $Au(10)$

→ If input is amplitude 10 times so 10 i°  
but when feed back is reduced by  
 $\frac{1}{2}$  so input is then 5i°

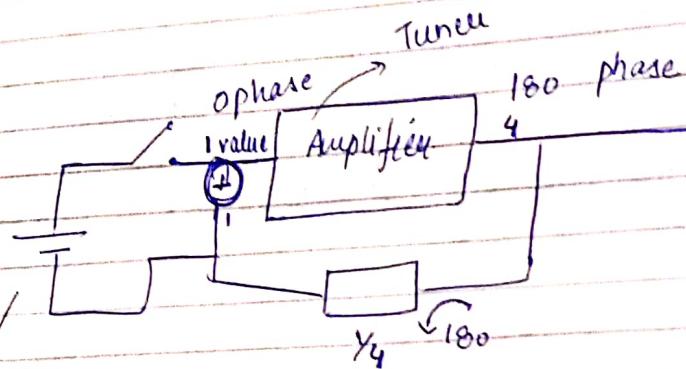
Negative feedback

↳ is good bcoz the ~~sig~~ input signal which was increased by the amplifier due to the negative feedback the signal is reduced.

~~sig~~ output signal is subtracted by a factor

- that is said to be negative feedback.

Positive feedback  
 Input dia amplify hua + positive feedback add hua  
 ie positive feedback



power circuit

When o generates

~~frequency noise~~ (containing all frequency)

~~Noise will vanish~~

Tuner will pick any one frequency and then amplified and noise vanish

source ~~o~~ gayab ho jaega

$$f = \frac{1}{2\pi} \frac{1}{\sqrt{LC}}$$

change this G

so it will be

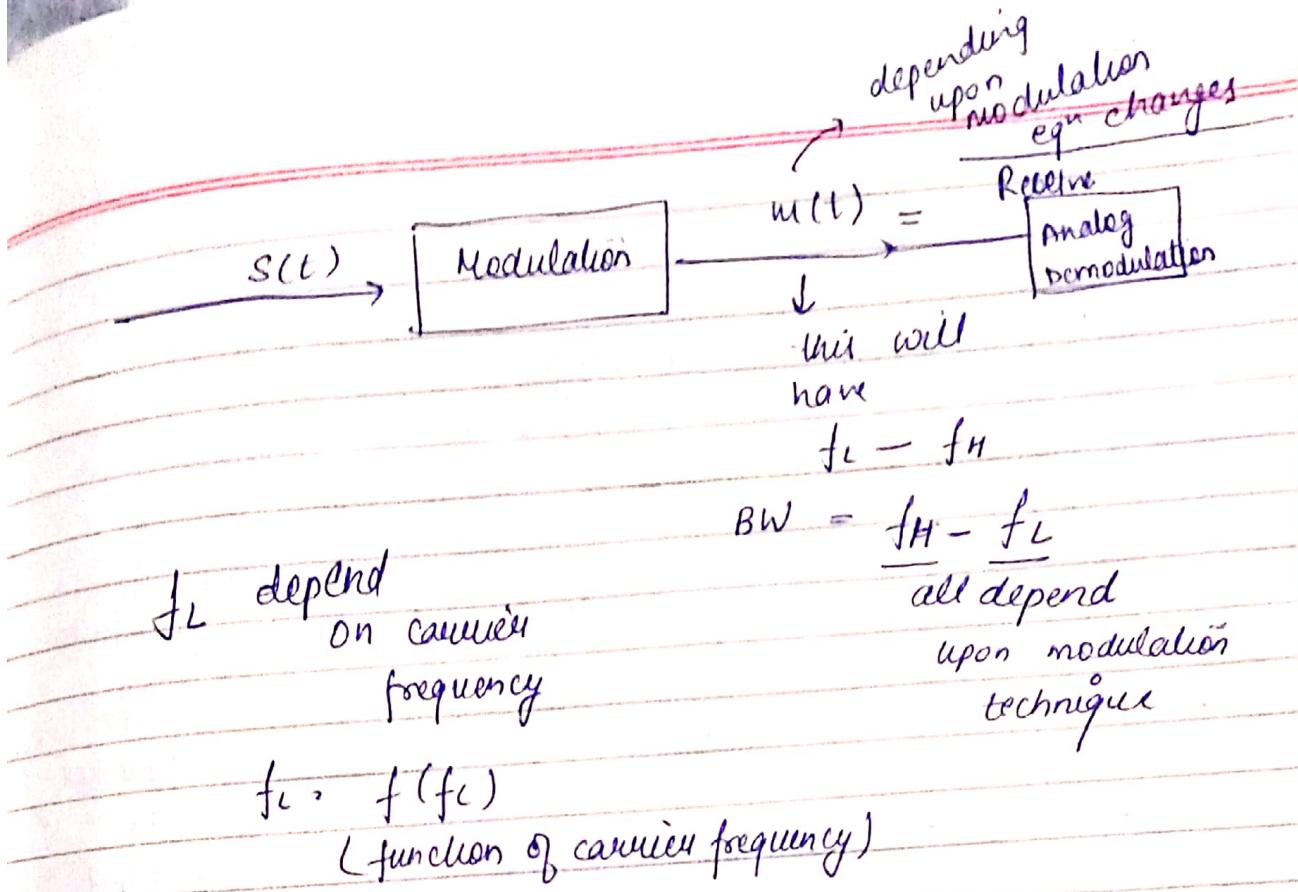
$$f_C = \frac{1}{2\pi} \frac{1}{\sqrt{LG}}$$

Amplifier

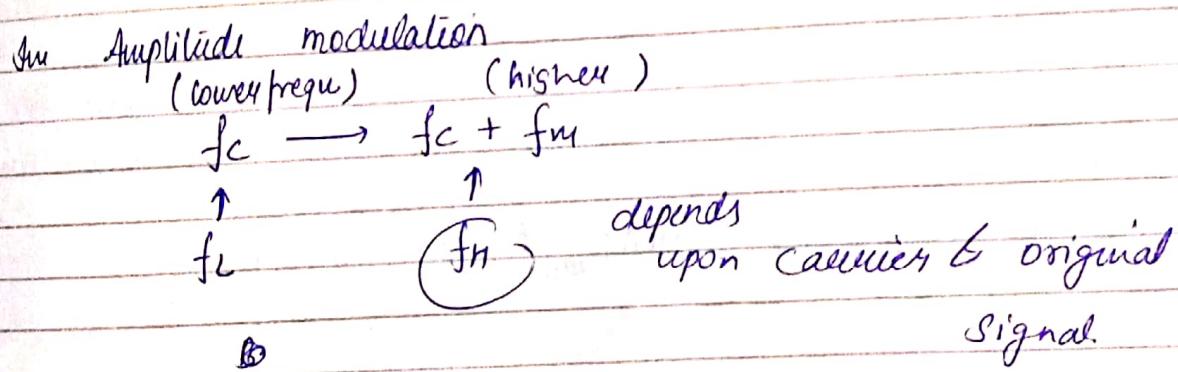
Analog Commu

S(t)

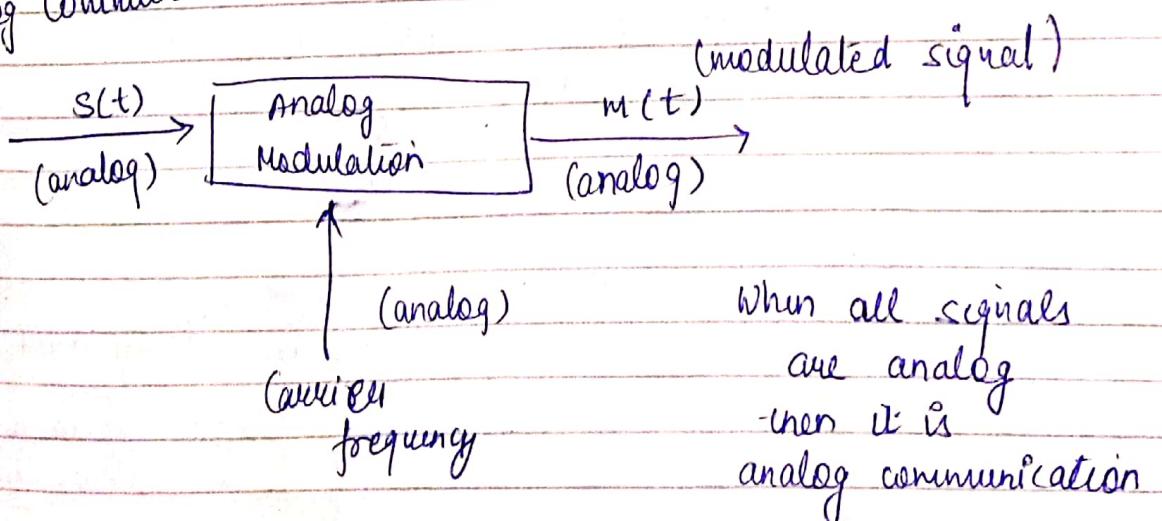
(analog)



$BW \rightarrow$  depends upon  $f_{max}$



## Analog Communication



## Amplitude Modulation

CARRIER frequency  $f_c$ ,  $\phi_c$  are fixed

Amplitude  $C$  will vary

constant of amplitude modulation

$$C_{\text{mod}}(t) = C + K_a s(t)$$

proportional

amplitude varied  
to the instantaneous  
value of the signal

depend on depth of  
modulation

(carrier power  
input  $K_a$   
ratio  $h_o$ )

$$m(t)_a = (C + K_a s \sin 2\pi f_{\text{mod}} t) \sin (2\pi f_c t + \phi_c)$$

$\uparrow$   
 $= 0$   
when  
amplitude  
modulating

2 Frequency

$C$

$f_{\text{mod}}$

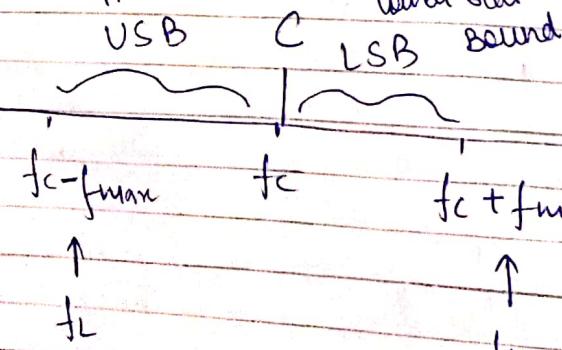
$f_{\text{mod}}$

$$m(t)_a = (C + K_a s \sin 2\pi f_{\text{mod}} t) \sin (2\pi f_c)$$

$\downarrow$

3 frequencies are engi  $m(t)$  me

Upper side band



All will have certain amplitude\*

depending on carrier & original frequency

$$m(t) = C ($$

$\downarrow$  their

so some bandwidth  
kya nega

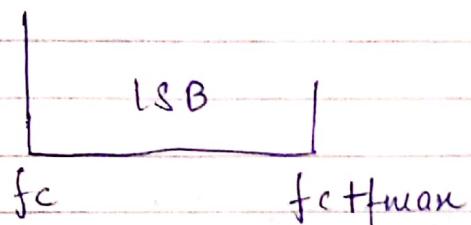
frequency  
with opposite  
phase

$m(t)$  Bandwidth

$$BW = f_H - f_L$$

$$= 2f_{\max}$$

In  $m(t)$  we are sending



$$\text{so } BW = f_{\max}$$

$f_c t + \phi_c$ )

↑

$= 0$   
when  
amplitude  
modulating

2 frequency Modulation

$C, \phi_c \rightarrow \text{constant}$

$f_{cm}$  (Modulated carrier frequency)

$$f_{cm}(t) = f_c + k_f s(t)$$

$$m(t) = C \left( \sin 2\pi (f_c + k_f s \sin 2\pi f_{\max} t) t \right)$$

↙ they will have many frequencies

Is there bandwidth  
kya hogा  $\omega \rightarrow \infty$  - ?  
No,

higher frequency amplitude

$k_m$   
hogा

Amplitude  $\rightarrow \infty$

frequency  
with opposite  
phase

~~$k_m$~~

inversely  
proportional  
to  $N^{\text{th}}$  frequency

Actual Bandwidth =  $\infty$

Effective Bandwidth

$$BWe \text{ U/L} = n f_{man}$$

sideband

If take upper band + lower band.

$$BWe = 2n f_{man}$$

FM audio quality is very high

↓  
costly w.r.t to bandwidth

Video is 5 MHz when FM then if  $n=5$

bandwidth will become 25 MHz

AM is not costly w.r.t to bandwidth

Studio -  $20 \text{ kHz} \times 5 = 100 \text{ kHz}$  Bandwidth

$$\begin{aligned} & \text{2 channel bai} \\ & = \frac{100 \times 2 \text{ kHz}}{2} \\ & = 200 \text{ kHz} \end{aligned}$$

### 3 Phase Modulation

$C, f_c \rightarrow \text{constant}$

$$\phi_{cm}(t) = \phi_c + k_p s(t)$$

Not used.

proportional

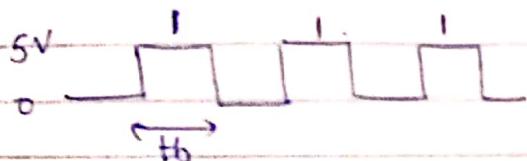
bps  $\propto$  band rate of transmission

~~DSP~~

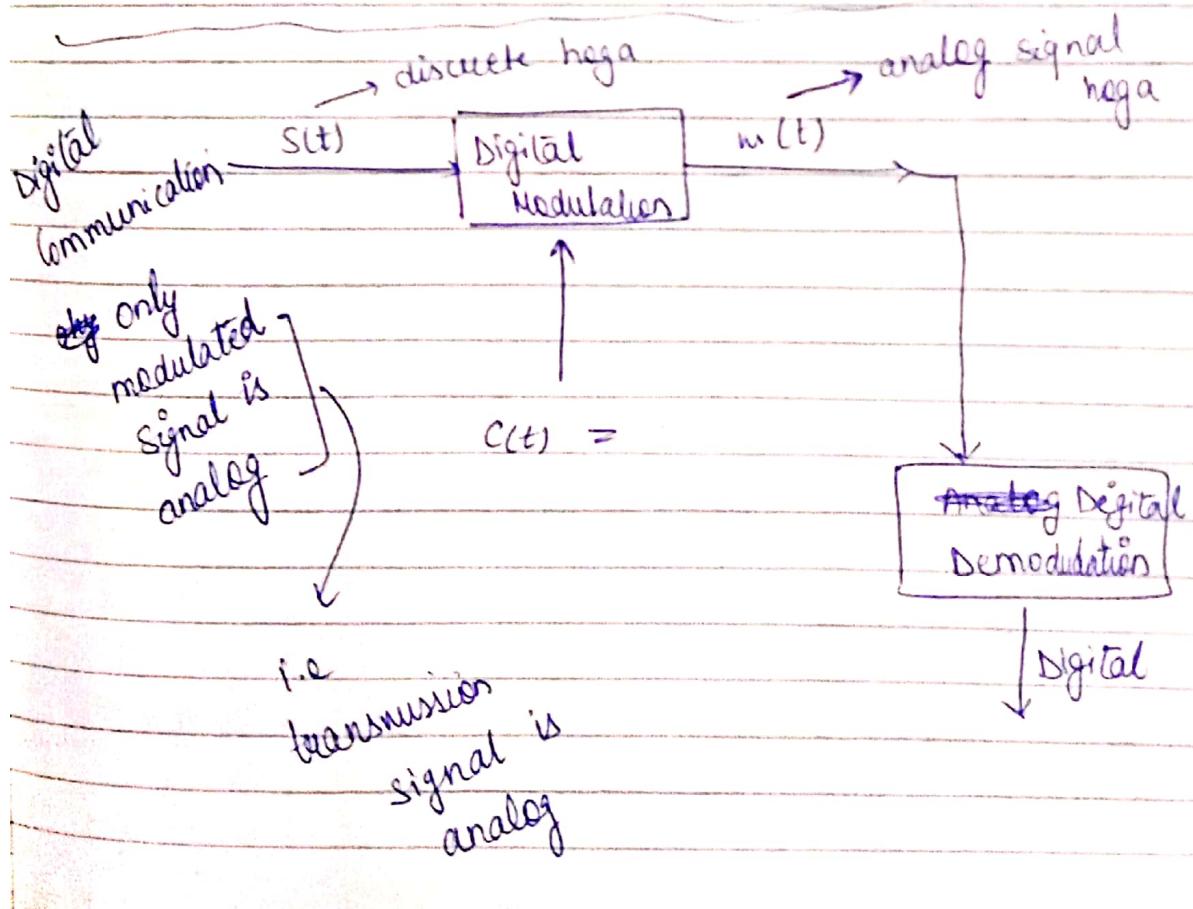
$$s(t) = 5V$$

Digital  
Digital Data (original)

Digitized music  
Digitized video



$$s(t) = 5V \quad 0 \leq t$$



Amplitude Shift Keying

$\Rightarrow C$  varies  $f_c, \phi_c \rightarrow \text{constant}$

Frequency shift keying

$f_c$  varies  $C, \phi_c \rightarrow \text{constant}$

Phase Shift Keying

Amplitude Shift Keying  
Frequency Shift Keying

$\Rightarrow C$  varies  $f_c, \phi_c \rightarrow$  constant

Phase Shift Keying

$f_c$  varies  $C, \phi_c \rightarrow$  constant

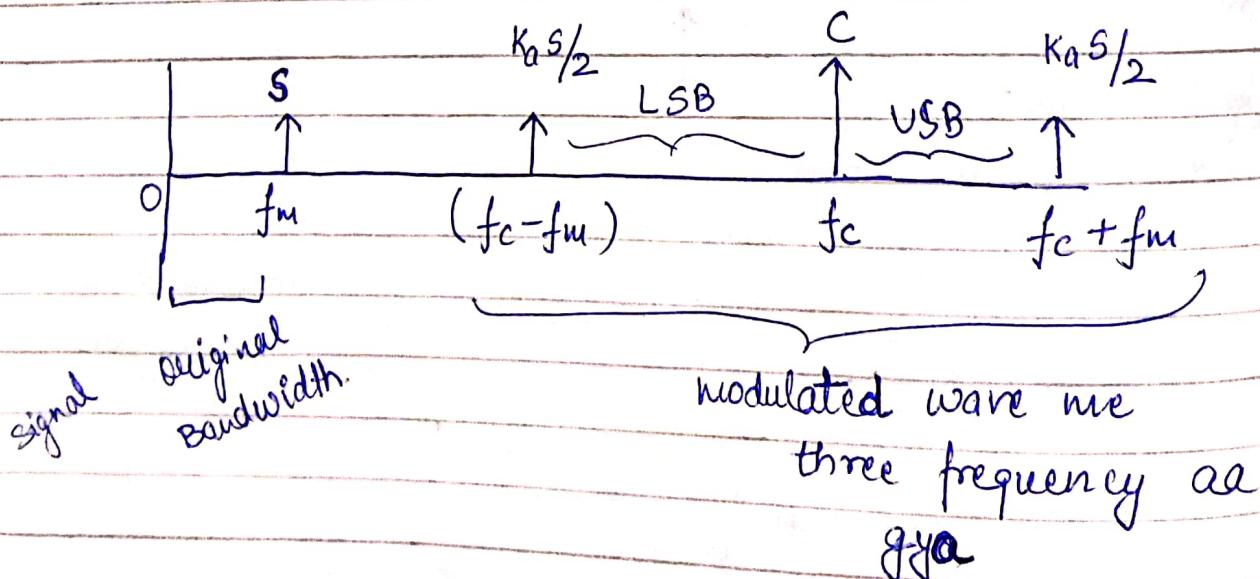
8/2/19.

$$w(t)_a = A(t)$$

$$\begin{aligned} w(t)_a &= (C + K_a S \sin 2\pi f_m t) \sin 2\pi f_c t \\ &= C \sin 2\pi f_c t + K_a S \sin 2\pi f_m t \sin 2\pi f_c t \\ &\quad \parallel \qquad \parallel \\ &= C \sin 2\pi f_c t + \frac{K_a S}{2} \cdot 2 \sin B \sin A \end{aligned}$$

$$= C \sin 2\pi f_c t + K_a S (\cos 2\pi (f_c - f_m) t - \cos 2\pi (f_c + f_m) t)$$

Dfg  
Grauer  
Band



$$B_{WS} = f_c + f_m - f_c + f_m = 2f_m$$

$$\text{original BW's} = f_m - o = f_m$$

LSB me we have carrier & signal

hum hta denge

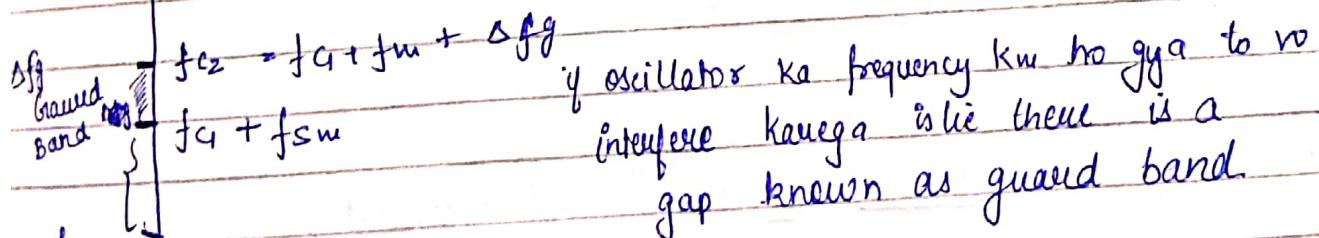
so we have VSB

$$\text{BW (m)}_e = f_m + f_c - f_c \\ = f_m$$

We have many sources

$s_1, s_2, \dots, s_n$

$f_{c1}, f_{c2}, \dots, f_{cn}$

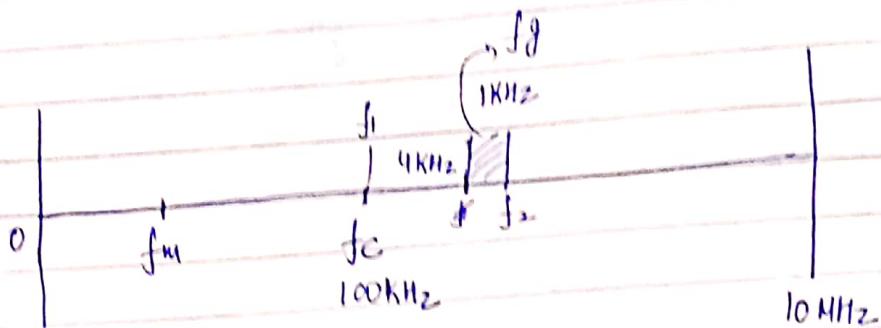


$$f_{cn} = f_{cn-1} + f_m + \Delta f_g$$

$$f_{cn} < f_{cn \# off} (\text{median})$$

12/2/19.

Voice Quality ke lie frequency modulation  
ke lie amplitude frequency



$$f_m = 4 \text{ kHz}$$

$$\text{guard band } f_g = 1 \text{ kHz}$$

$$= 10 \times 10^3 - 100 \rightarrow \text{frequency (here it can be) } 20 \text{ kHz}$$

$$4n + n - 1$$

$$= 5n - 1$$

$$= 5n$$

$$(10000 - 100) = \frac{9900}{5} = n \text{ (no. of channels)}$$

FM

$$\text{Stereo Music} = 20 \text{ kHz} \times 2 + 1 \text{ kHz}$$

$$= 41 \text{ kHz}$$

$$\approx 40 \text{ kHz}$$

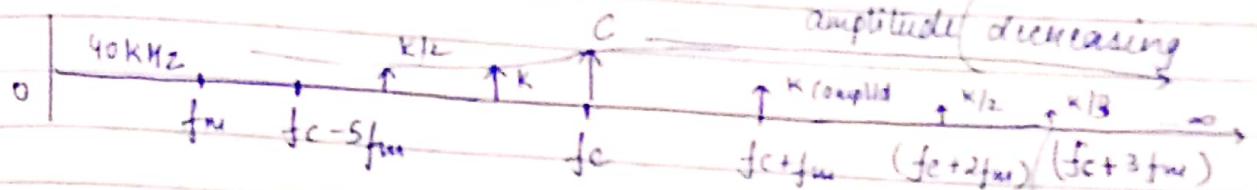
$$m(t)_f = C \left( \sin 2\pi (f_c + k_f s \sin 2\pi f_m t) t \right)$$

there is a sin within a sin

can be broken using high  
techniques

$m(t)$

power  
 $\frac{1}{4}$  decreased



$n = 5$  next after this power will not be considered.

We will stop at  $f_c + 5f_m$

2 approximation

- 1)  $\infty$  bandwidth tha use kai kia  $10f_m$
- 2) then we took lower and upper side band.

$$BW_C(e) = 5f_m$$

$$BM_{\text{music channel } e} = 20 \times 5 \\ = 100 \text{ kHz}$$

Amplitude ~~not~~ modulation me demodulation me noise  
frequency modulation me demodulation me noise  
ni ata  $\rightarrow$  quality adi ati hai

Telephone voice

Given

$$BW \rightarrow 100 \text{ kHz} - 5 \text{ MHz}$$

$$\text{Music channel} \rightarrow 20 \text{ kHz} \times 5$$

$$= 100 \text{ kHz}$$

$$90 \text{ MHz} \quad \text{FM channel} = 100 \text{ kHz}$$

$$108 - 90 = n \times (0.1) + (n_i)$$

Bandwidth

frequency of clear

+ 1 MHz

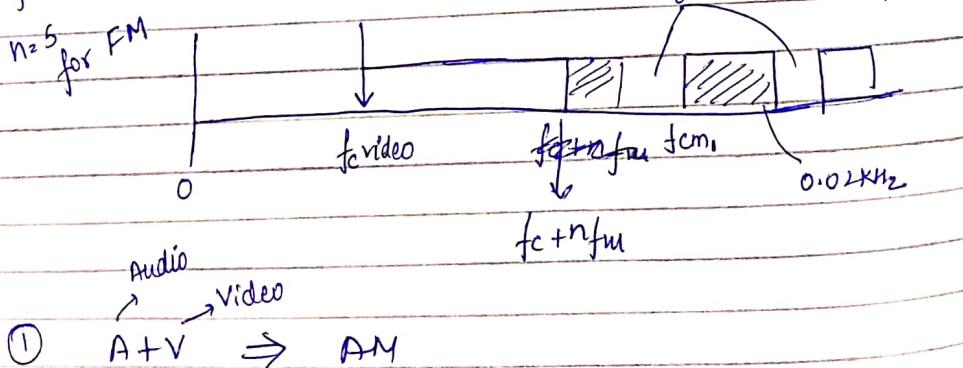
108 MHz

This is frequency division multiplexing

13/2/19: Stereo music transferred through FM

$n=1$   
for AM

$n=5$   
for FM



①  $A + V \Rightarrow AM$

$$5 \text{ MHz} + \underbrace{0.02 \text{ MHz}}_{\text{Music ko AM kia}} + 0.02 \text{ MHz} = 5.04 \text{ MHz}$$

Music ko AM kia

to quality ni ayega to

yeh karenge hi nhi

2) A+

frequency of  
guard  
band  
 $(10 \text{ kHz})$

3. A (FM) + V (

0.2 MHz is  
FM for audio.

Hence TV ka bandw

In TV transmis  
04

Q. Why do we do

When source is di  
analog and  
This is known

102 PM

$$2.) A + V \Rightarrow FM$$

$20 \text{ kHz} \times 5$   
+  
 $20 \text{ kHz} \times 5$

$$25 \text{ MHz} + 0.2 \text{ MHz}$$
$$= 25.2 \text{ MHz}$$

(huge bandwidth  
as compared to)  
AM

frequency of  
guard  
band  
0.1  
(10 kHz)

Yeh bhi disragaud kavunge,  
but audio ko FM kavunge  
pr video ko AM kavunge

$$3.) A(FM) + V(AM) = 0.2 \text{ MHz} + 5 \text{ MHz}$$
$$= 5.2 \text{ MHz}$$

0.2 MHz is very less & so we can afford  
FM for audio.

Hence TV ka bandwidth km ho gya.

Q. Why in TV transmission which is modulation  
or amplitude technique is used.

Q. Why do we do FM for audio.

When source is digital medium transmission is  
analog and destination is digital  
This is known as digital communication.

(source, medium and destination are analog)

### 1.) Fully analog communication

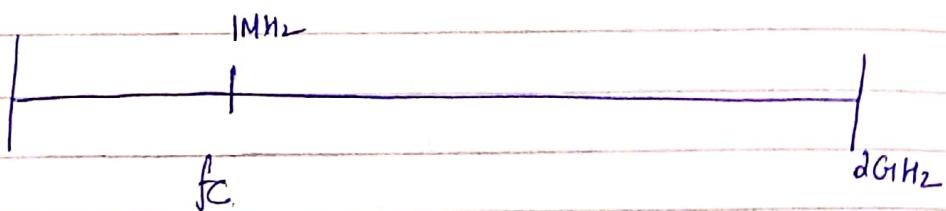
space

wired medium (cable)  
(coaxial cable)

Cable TV

### Coaxial Cable

#### Cable Bandwidth



$$\begin{aligned}\text{Total Bandwidth} &= 2 \text{ GHz} - 1 \text{ MHz} \\ &= 1999 \text{ MHz}\end{aligned}$$

\* 1 channel take 5.2 MHz

~~1 MHz~~ = guard band

$$1999 = (5.2)n + (n-1)(0.1)$$

$$n = \left\lfloor \frac{1999}{5.2} \right\rfloor$$

### Twisted Pair

↳ Telephonic line no modulation done

0 - 4 kHz

Pulse Code Modulation se 64 Kbps ~~mechanical~~ digital ban jata hai

then this reversed onto analog using reverse pulse code modulation.

## Digital Communication

not fully digital

communication

source and destination digital  
medium analog

- on wired medium

- twisted pair      0 - 1 GHz
- coaxial cable      0 - 2 GHz
- optic fibre      0 - 20 GHz

Single cable & a single source  
no problem.

But when multiple sources

then min ho jāega  
using time division  
multiplexing  
04

STDM

Synchronous TDM



t=0 pe \$S\_1\$ ka data line pe jāega

then after  $\Delta T$  \$S\_2\$ ka data  
and till \$S\_n\$.

If some source has no data then dummy data

is send

and at receiving end after some time ~~at~~  
signal demodulated i.e. given to ~~source \$S\_1, \dots, S\_n\$~~  
 $S_1, S_2, \dots, S_n$  given at  
destination  $D_1, D_2, \dots, D_n$

Analog me wire me FDM hoga. eg. Cable TV  
Digital me wire me TDM hoga.

~~On space~~ fully digital

wired me source, destination  
and medium bhi digital  
hoga

Fully digital me digital regeneration

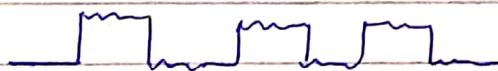
Source & Destination pe processing hoga  
error detection / recovery

compression      decompression

and many more

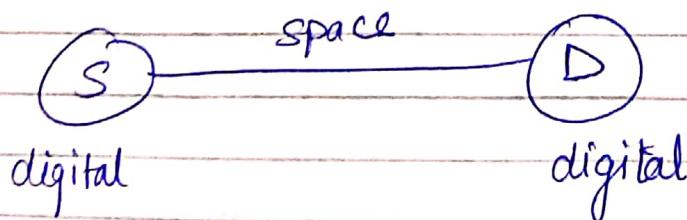


during medium  
noise can come

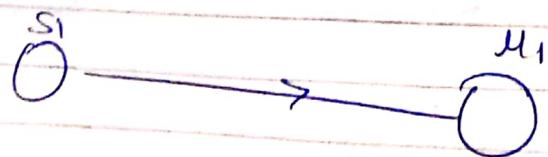


bt noise hata ke  
we can amplify

~~On space~~

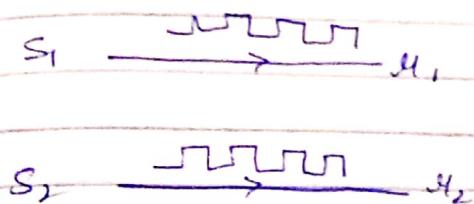


if single source & receiver

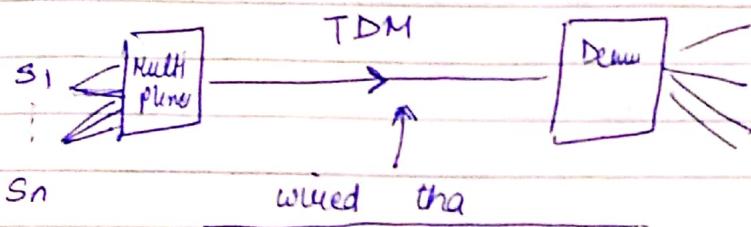


then digital data can be send

But when multiple sources



but both digital min  
no jaega influence  
hoga.

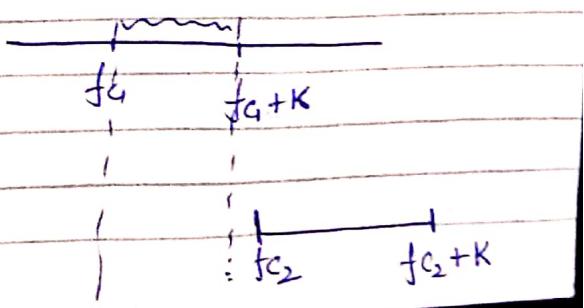
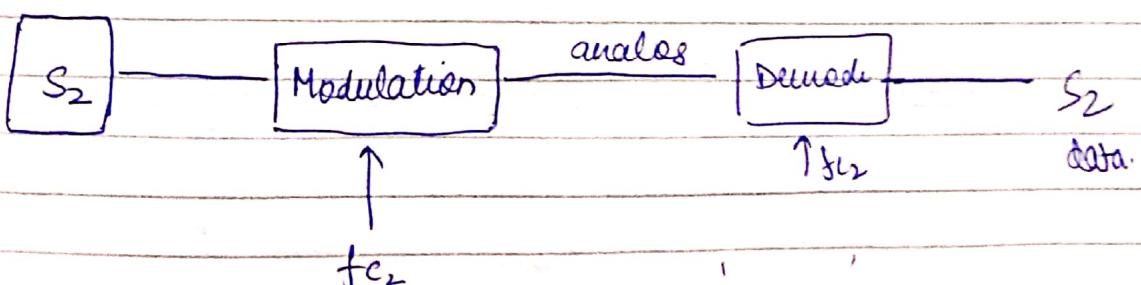
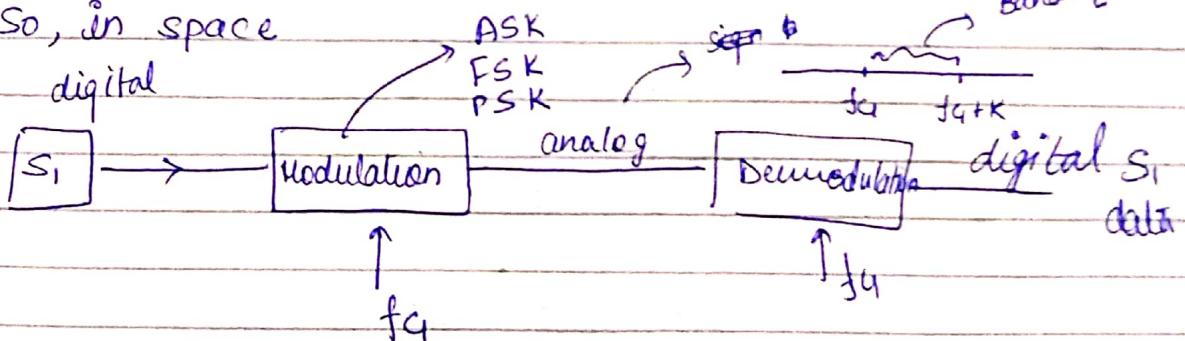


wired tha  
but space is vast  
anyone can mix malicious ~~code~~ <sup>data</sup>

Hence, direct digital comm. is not possible in space.

Hence digital data is not sent through space.

So, in space



This is hybrid communication

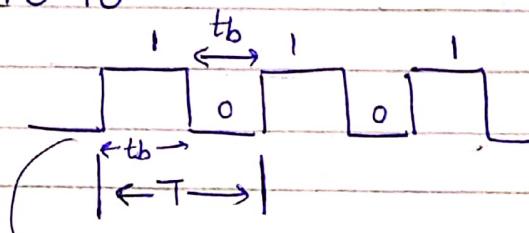
not fully digital, not fully analog

during transmission signal regenerate nahi ho paega. bcoz analog. hai during medium.

but error detection / recovery at source and destination can be done.

1) 1010

→ this data is changed maximally to iske lie



bandwidth bw

zada hogा.

bit line is controlled using digital clock

Data Rate

$$t_b \xrightarrow{\text{sec}} \text{time me } 1 \text{ bit is send}$$
$$1 \text{ sec} = \frac{1}{t_b} \text{ bit}$$

BRT → bit rate transmission

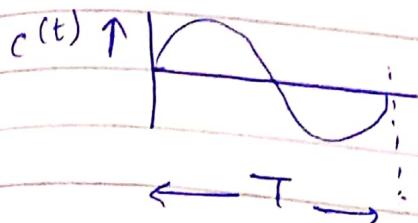
= digital rate of transmission

$$= \text{digital BW} = \frac{1}{t_b}$$

$t_b \downarrow$  to BW ↑

agar ~~jaldi~~ ~~jaldi~~ jaldi jaldi  
bhejoge to medium change  
zada hogा.

## 2.) Signalling Element

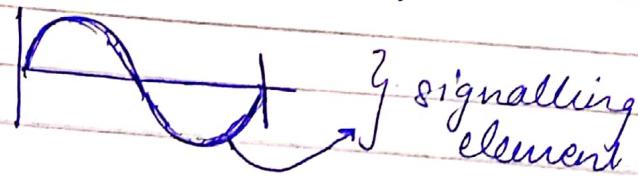
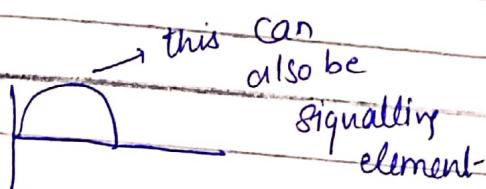


(We will use carrier to send digital data)  
Carrier signal

$$f_c = \frac{1}{T}$$

but digital signal is discontinuous and carrier signal is continuous

We will transfer 1 bit on 1 cycle of carrier



In ASK,

if bit is 0

at t so signal  $s(t) = 0$

Amplitude of carrier

$$m(t) = C + k_a \times 0$$

so sending 0 the amplitude not change

so for 0

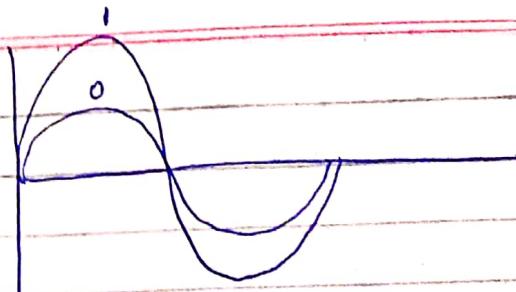


but 1

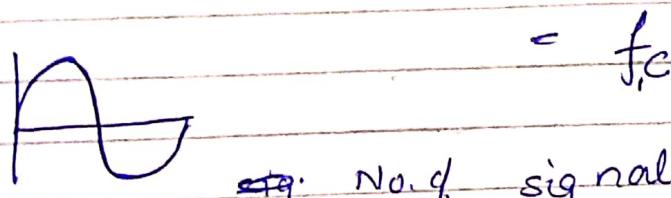
$$\text{so } m(t) = C + k_a \times 5V$$



1 signalling element  
= 1 Band



3) Signalling rate = <sup>Band Rate</sup>  
No. of signals transmitted per rate



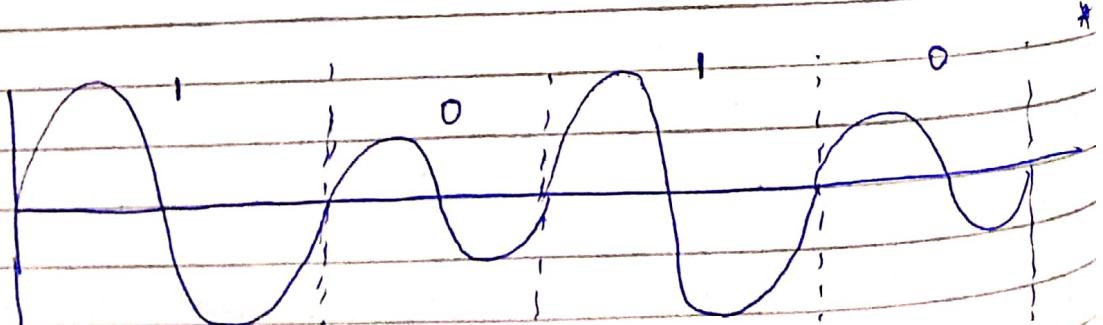
e.g. No. of signal =  $f_c$

But if



So, no. of signals =  $2f_c$

ASK sending 101010



1 signalling element me 1 bit

2 level of signalling bunge to bit rate =  
baud rate

Gaussian ka ek frequency  
so Bandwidth is 0.

Simple ASK -

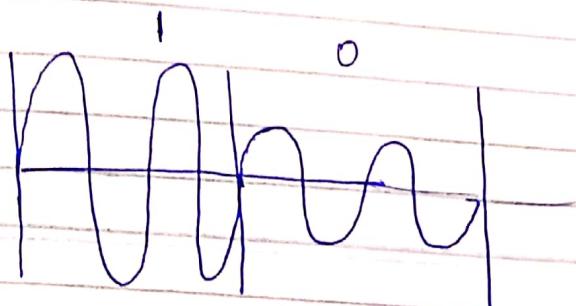
2 level of signalling

(i.e. using 2 level we can send digital data)

- \* This is a composite analog signal containing different frequencies. and since not single wave so bandwidth is not 0.

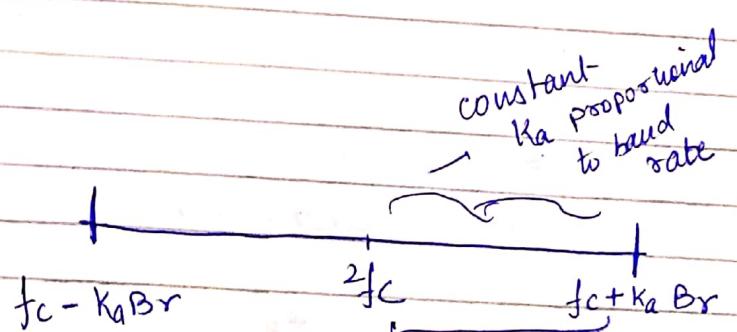
Baud rate is doubled.

So

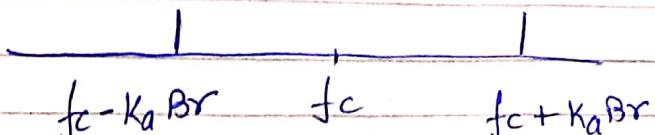


→ signalling rate  
 $= 2f_c$

↓  
→ yha pe change  
faster  
hoga



→ signalling  
rate =  $f_c$  ] →



$$BW_{CH} = 2K_a Br$$

We need to choose  $K_a$

we choose 0 then  $BW_{CH} = 0$  then no

information is send

When  $K_a = 1$

$$\text{Bandwidth} = 2 \text{ BaudRate}$$

This is very high

So, value of  $K_a$  (optimal)

$$K_{a_{op}} = \frac{1}{2}$$

So, in ASK

$$\boxed{BW_{CH} = B_s}$$

information can be modulated / demodulated easily

$b_s$  = bit rate

$B_{BAUD}$  = Baud Rate

$BW$  = Bandwidth

$$B_{BAUD} = BW \quad \text{when } K_a = \frac{1}{2}$$

when 2 level of signalling then

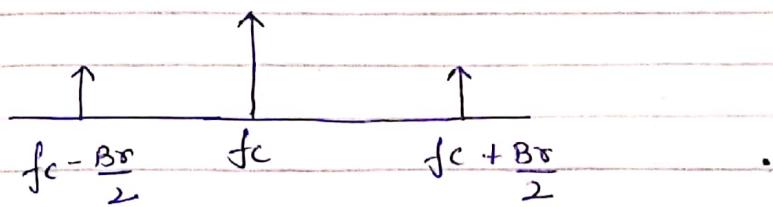
$$b_s = BW$$

↓

Also known as  
digital bit rate.

15/2/19

ASK



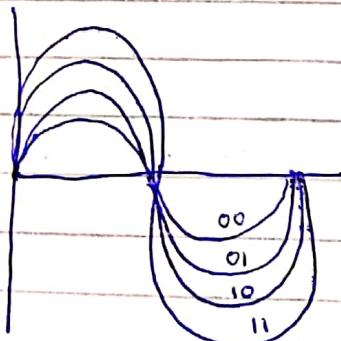
$$BW_{CH} = Br = f_c + \frac{Br}{2} - f_c - \frac{Br}{2}$$

Digital transmission  $\propto$  Bandwidth of the channel

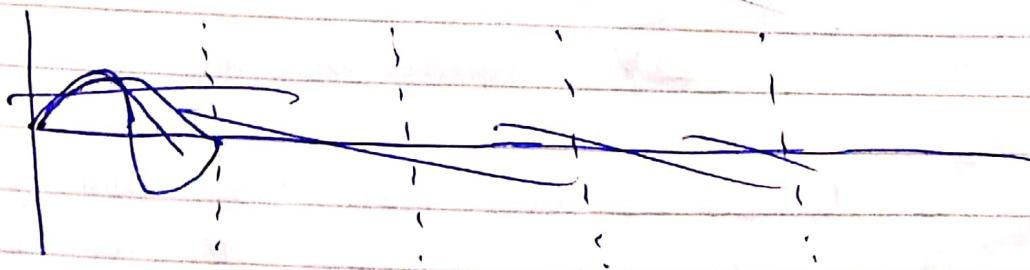
4 level of signalling

2 bits will be choosed for signalling element-

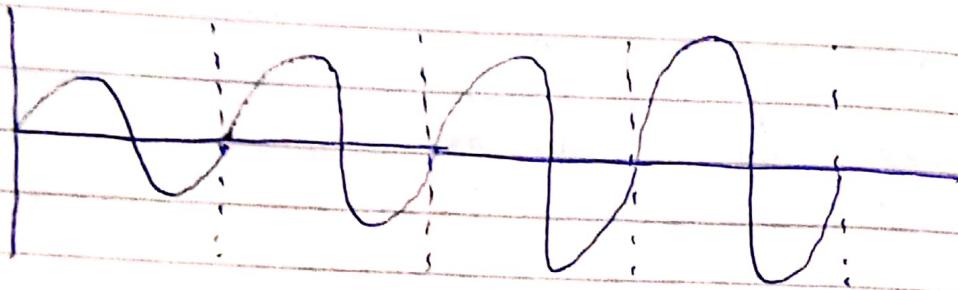
1 baud me 2 bits send kar rhe hain.



0 0 0 1 1 0 1 1



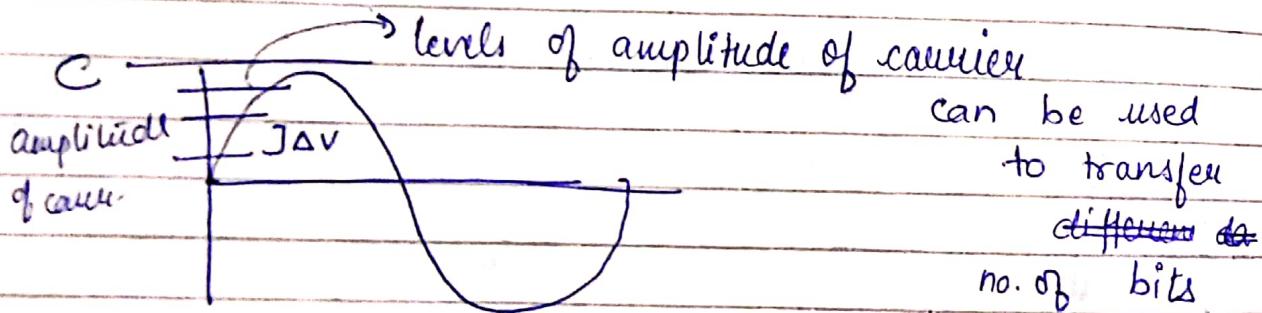
00 01 10 11



$$b_r = B_r \times \log_2 V$$

$$= BW \times \log_2(V) \rightarrow \text{we will change this}$$

Jitni analog bandwidth utna baud rate.  
So, we need to minimise bandwidth and increase the bit rate transmission.



C ko kitne level tk divide karenge - ?  
 $V \rightarrow$  no. of levels

$$V = \frac{C}{\Delta V} \text{ (difference b/w two levels)}$$

If  $\Delta V$  is very less then there are chances of noise because if two levels are very near so, if noise comes then it will not be easy

to distinguish b/w upper level or lower level.  
it is 00 or 01  
it will be difficult to distinguish.

In noisy environment,  $\Delta V$  has to be kept greater.

$\Delta V_{\min}$  = function of noise level  
or signal state  
Noise

$$\propto \frac{1}{\frac{\text{Signal}}{\text{Noise}}} = \frac{\text{Noise}}{\text{Signal}}$$

$$\Delta V_{\min} \propto \frac{\text{Noise}}{\text{Signal}}$$

Noise<sub>env</sub> ↑ to  $\Delta V_{\min}$  ↑ hoga

$$\text{so } V_{\min} = \frac{C}{\Delta V}$$

$V_{\min}$  hoga.

Modem capacity depends on

1) Noisy Environment (Noisy)

But we cannot increase the amplitude of the carrier signal - ?

Ans.

$$P_{\text{modem}} \propto C^2$$

C increases

$$P \rightarrow \text{man}(C^2)$$

$$V \times I = \text{increases}$$

Current J increase

$I^2 \times R \rightarrow$  Power burnt in coil  
increase

परों

current increase ho rha hai to resistance  
decrease karna padega

$$R = f \frac{L}{A} \xrightarrow{\text{constant}}$$

To Area increase karna padega

i.e. wire will be thick so the price of  
modem increases.

So, for a particular modem amplitude of  
carrier will be fixed and so the level.

Q

$$f_c = 4 \text{ kHz}$$

$$B_r = 4000 \text{ Baud / Sec}$$

$$V = 16$$

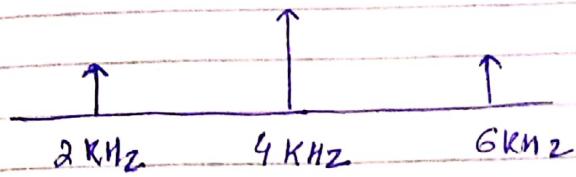
$$b_r = 4000 \times \log_2 16$$

$$= 4000 \times 4$$

$$b_r = 8 \times 16000 \text{ kbps}$$
$$= 16 \text{ kbps}$$

$$BW = 4 \text{ kHz}$$
$$= \boxed{B_r = 4000 = 4 \text{ kHz}}$$

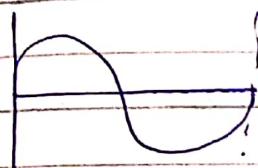
## frequency Spectrum



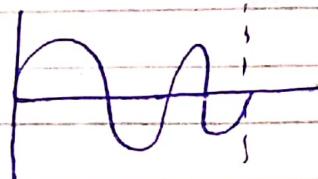
## Frequency Shift Keying (FSK)

✓ no. of frequencies

of two frequencies taken



for 0

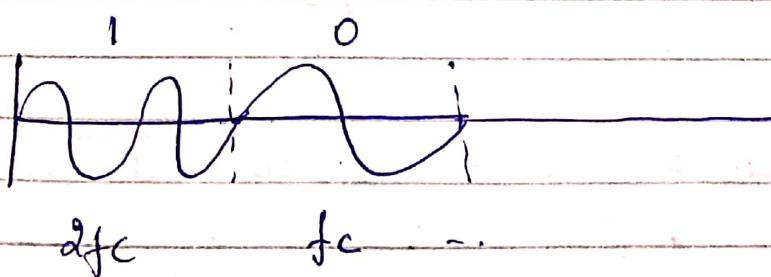


for 1

(frequency increased)

so

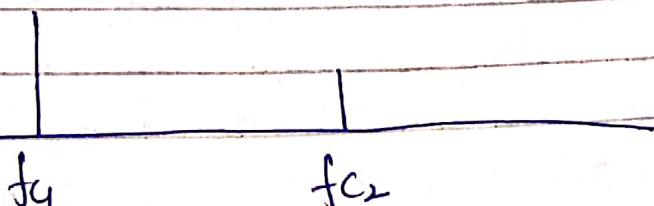
$$f_1 = f_c \quad \text{and} \quad f_2 = 2f_c$$



$f_1, f_2$   
are  
two frequencies

not a pure signal (a composite signal)

$$f_2 = f_4$$

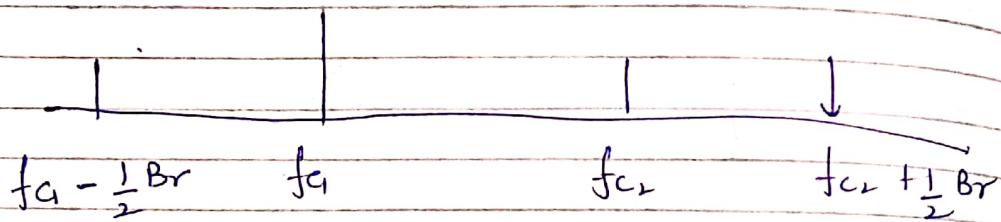


$$f_1 = f_c$$

$$f_2 = 2f_c$$

$f_1$  will be changed to  $f_2$   
 $f_2$  to  $f_1$

so that will generate  
 two more frequencies proportional to  
 band rate



$$\text{BW} = f_2 + \frac{1}{2} Br - f_1 - \frac{1}{2} Br$$

$$= (f_2 - f_1) + Br$$

$$Br = \text{BW} - (f_2 - f_1)$$

$$= \text{BW} - (2f_c - f_c)$$

$$Br = \cancel{\text{BW}} - \cancel{f_c} \quad (\text{from } 2 \text{ level})$$

level 8 = 3 bits are sent per signal.

$$Br = (BW - (f_H - f_L)) \log_2 V$$

$V \uparrow$  to frequency  $bh^{\circ}$  hagegi

$f_H$  &  $f_L$  ka gap bd jagega

So  $V$  will not be increased

so FSK is no

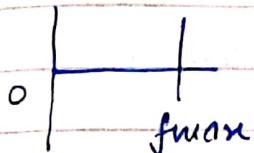
so FSK is not chosen for digit data.

Rough  
book

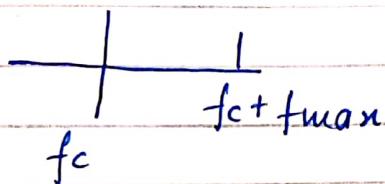
so FSK is not chosen for digit data.

20/2/19

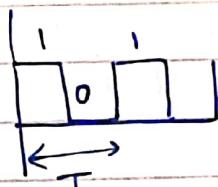
Analog signal ka bandwidth



After modulation, channel bandwidth

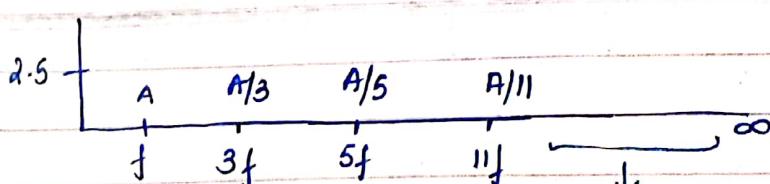


But digital signal



$T$  = Time period of square wave

Digital signal is a no. of sine wave



yha pe signal ka strength  
kii ho jaega uska  
Significance nhi uahega

So spectrum ka ban gya form  $0 - 11f$

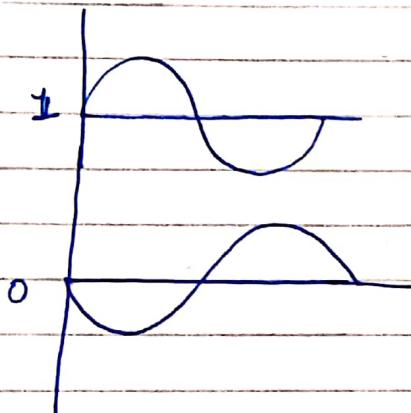
$$BW_S = 1/f - 0 = 1/f = \frac{1}{T}$$

FSK

$$BW_{CH} = (f_{CH} - f_{CL}) + Br$$

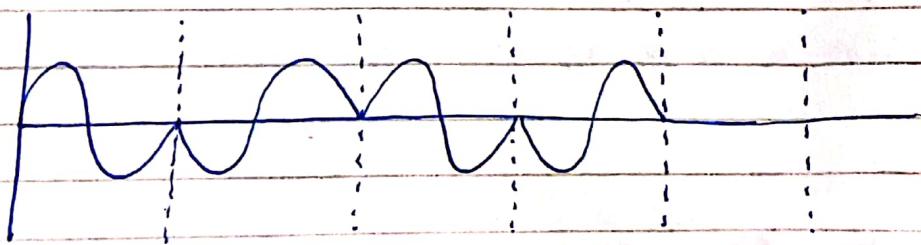
Q why FSK is not used for digital data?

PSK → Phase shift Keying



1 0 1 0 1 0

sign signal  
changes at  
the change  
of  
band rate  
so, not  
a sin  
wave



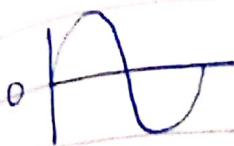
$$fc - \frac{Br}{2} \quad fc \quad fc + \frac{Br}{2}$$

$$BW_{CH} = Br$$

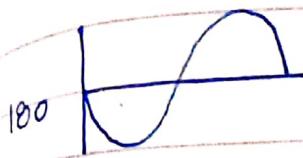
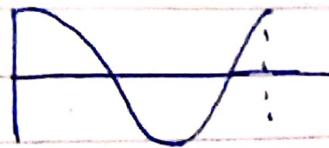
How to increase band  
rate?

We take 4 phases

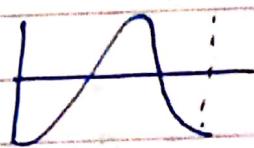
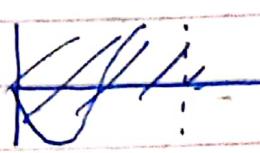
90°



90°



180°



00 → phase 0

01 → phase 90°

10 → phase 180°

11 → phase 270°

when sending  
2 bits per baud

10 11 01 00

10 11 , 01 . 00

Baud rate kbaud  
kia to  
pattern jaldi  
change hoga

Baud rate zyada

kia to

pattern slowly  
change hoga. - ?

$$d.u = B_r \times \log_2 4$$

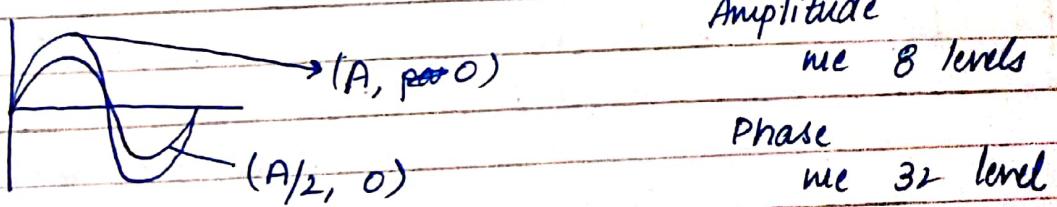
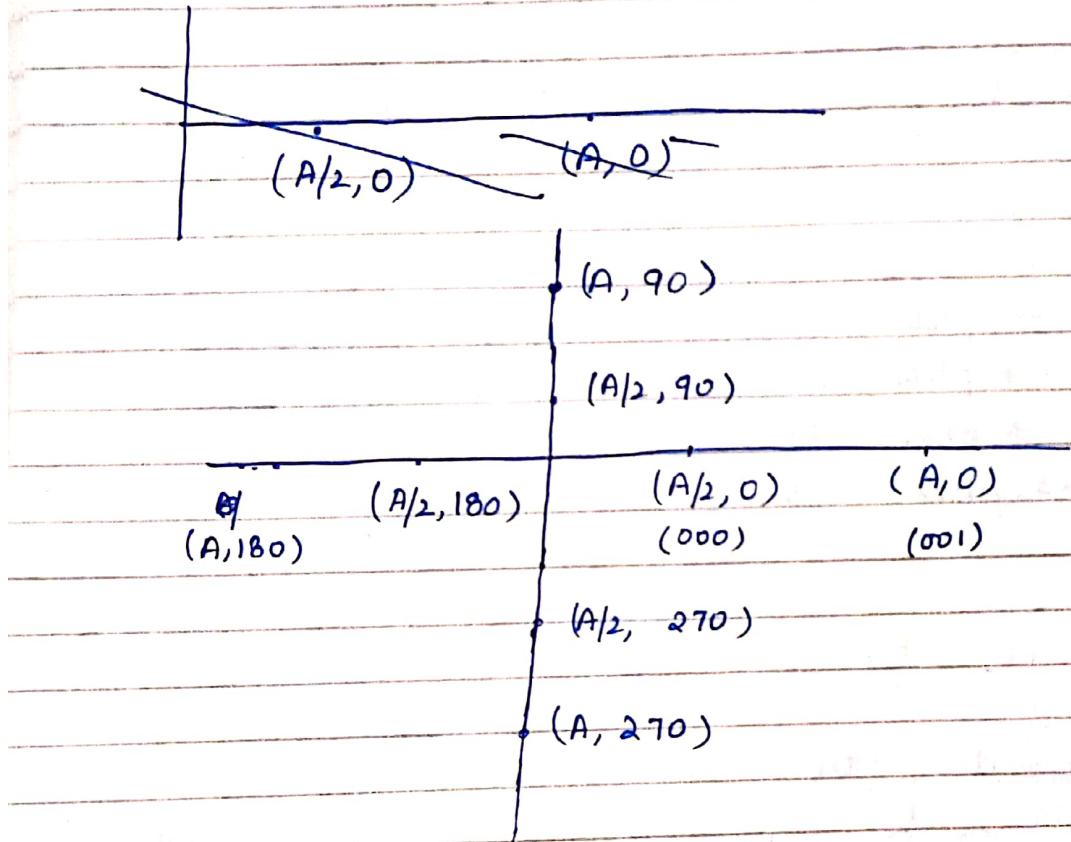
$$BW \times 2$$

Phase since not related to power. So  
we increase V much more than ASK.

0, 45, 90, - 270

8 phase 3 bits can be send.

ASK as compared to PSK  
ASK is also discarded.



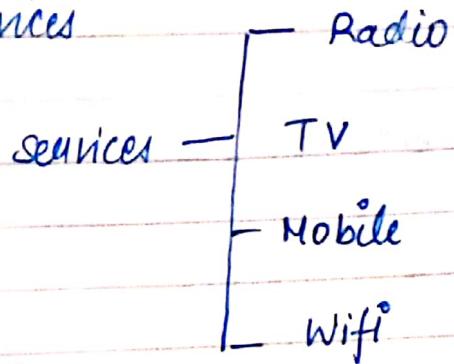
$$V = 2^3 \times 2^5 \text{ levels}$$

aa jaenge

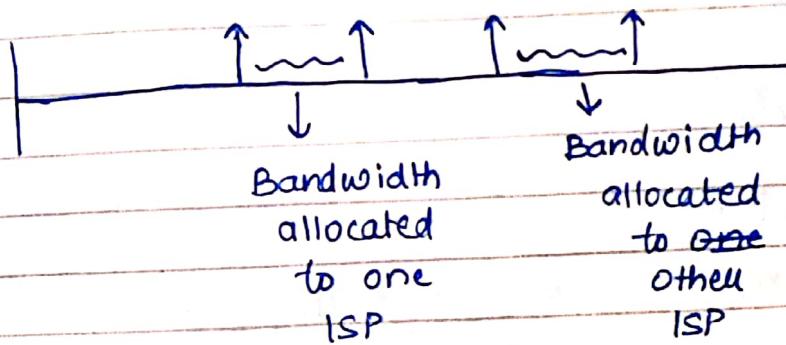
This is the quadrature  
amplitude modulation  
(QAM)

# Multiple Access Schemes

wireless communication to provide services



ISP provide these services and they should get bandwidth in space depending upon the service



Frequency Division Multiplexing  
is used to divide bandwidth

Mobile Communication ke lie

1800 - 1900 MHz

890 - 960 MHz allocate kota hai

For Wifi

2.4 GHz -

5 GHz -

These zones are license-free

## Multiple Access Schemes divided into

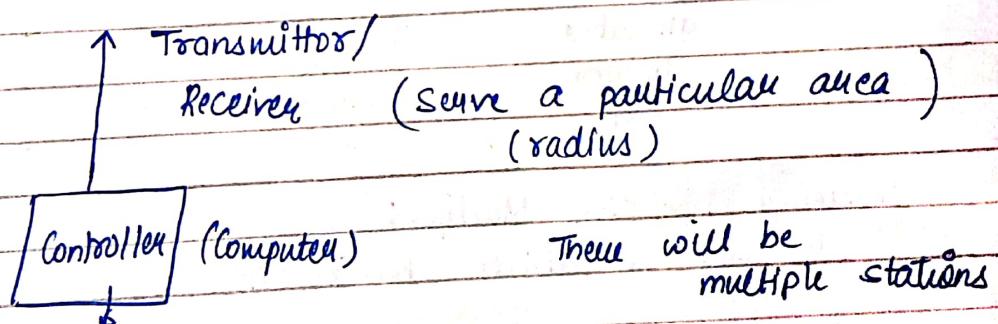
### → Centralised

(central coordinator who will create and allocate channel to user i.e mobile device )

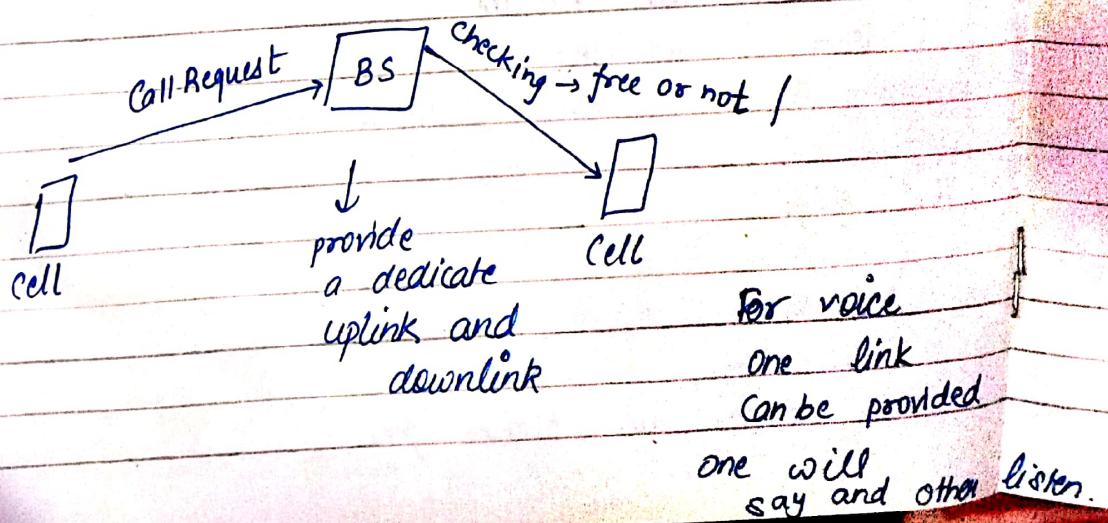
e.g Mobile Communication  
= Base Station

For mobile voice / data communication

BS ka two parts



There will be multiple stations

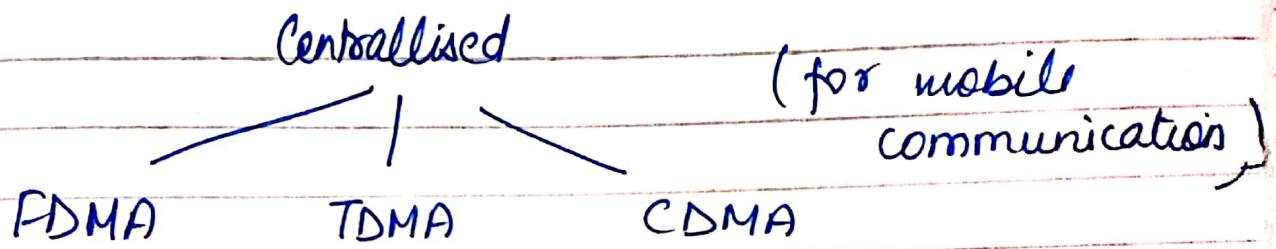


for data

we require 2 link

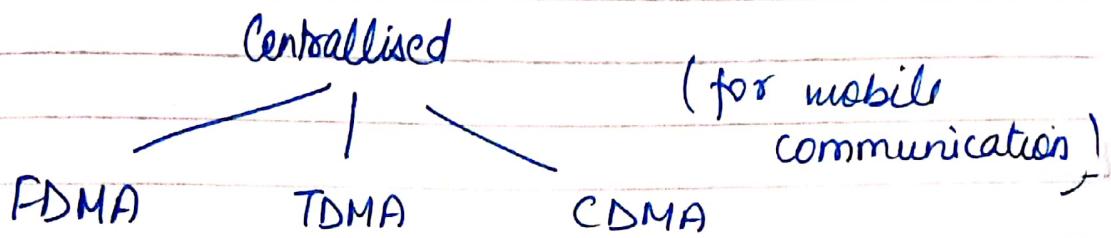
one for sending, one for

receiving



Antenna for both transmitting/receiving  
known as trans-receiving  
antenna.

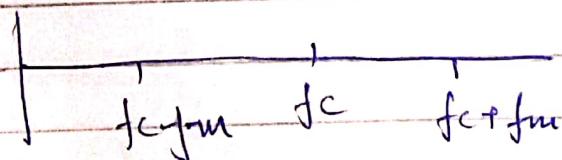
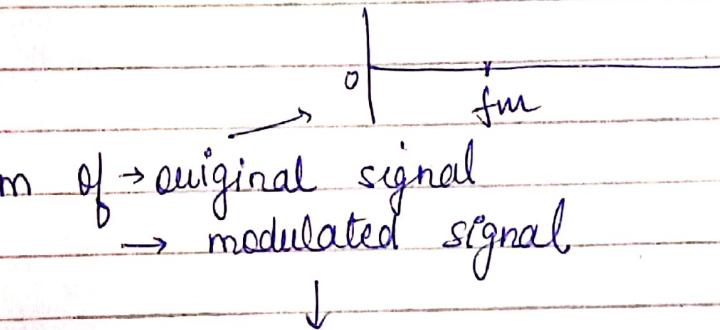
For data  
we require 2 link  
One for sending, one for  
receiving



Antenna for both transmitting/receiving  
known as trans-receiving  
antenna.

23/2/19

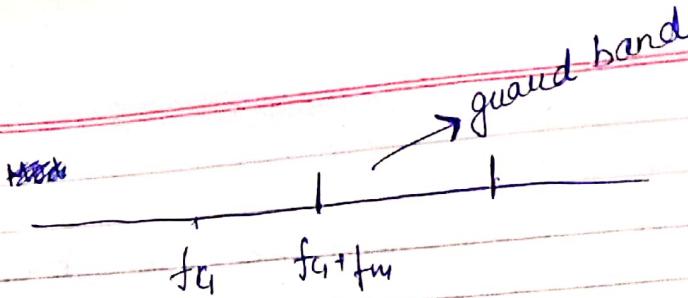
Frequency spectrum of → original signal  
→ modulated signal



$BW_{CH(a)}$  = Bandwidth of the channel  
in amplitude modulation.

If we include LSB + USB  
then bandwidth is doubled.

But, taking one band helps ~~demod~~ separate  
original signal from carrier at  
receiver.



$$\text{Next frequency } f_{c2} \geq f_c + f_m + \Delta f_g$$

$f_g \rightarrow$  ek frequency denote karta hai  
bt since guard band is not a single frequency so  $\Delta f_g$

CARRIER frequency is generated by oscillator.

So, frequency km zyada hoga to interfere hoga.

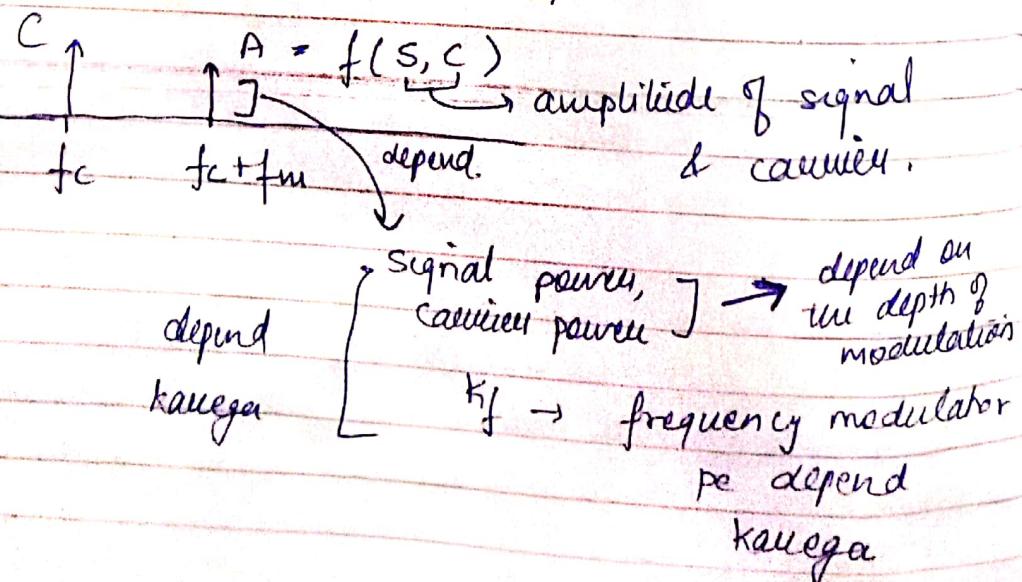
So, guard band jukte

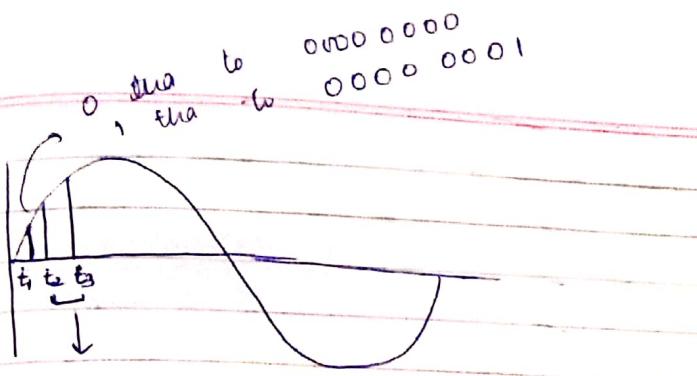
In frequency modulation

we have a sine wave within a sin wave we cannot separate it using trigonometric functions.

says

JO  
uska





ADC  
Analog  
Digital  
Converter  
using 8 bit.

① sampling.

digitization  
acha hoga

1 cycle ko sample

Karne ke lie

bit stream ka

value

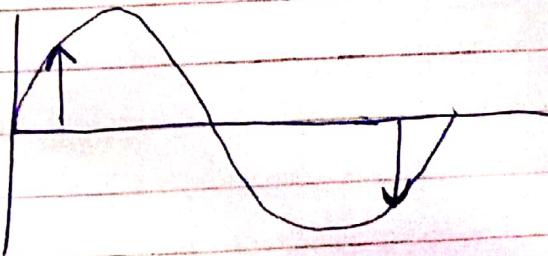
infinite  
hoga



because  
analog  
ko  
continuous  
rakhna.  
hai

says infinite sampling nhi.

Jo maximum frequency hai usse ko  
iske cycle me 2 bar sample karao



Sampling rate =  $2f_m$

Each sample ko multiply by  $n$

i.e.  $2f_m \times n$

no. of bits

Telephonic

$$f_m = 4 \text{ kHz} \times 2 = 8 \text{ kHz}$$

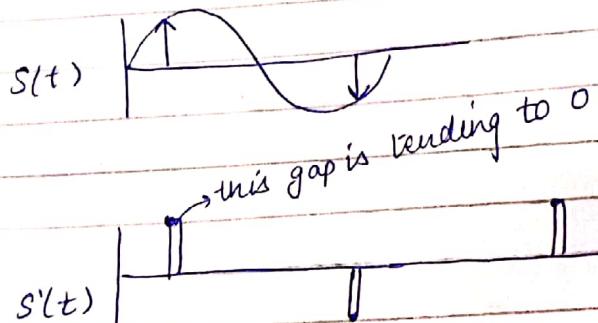
Sampling rate

$$= 8000 \text{ samples}$$

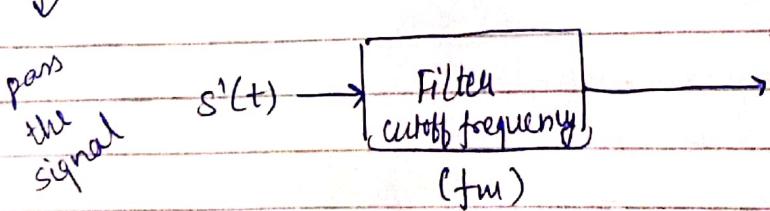
8 bit used to encode analog

$$= 8000 \times 8$$

$$= 16000$$



(But this is not the original signal)



0 —  $f_m$  — highest frequency

↳ Filter will take signal with frequency with  $f_m$  and <sup>signal frequency</sup> <sub>waves</sub> lower than  $f_m$

Higher frequencies to eliminate noise diff

sample  
Music channel

sample twice the maximum frequency  
and use 8 bits for each sample.  
i.e decode each sample using 8 bits.

Music  
Channel 1

$f_m$

20 kHz

Soate = 40000 samples per second

$n = 16$

$$\text{Channel 1} = @ 40K \times 16 \\ = 640 K$$

$$2 \text{ Channel} = 1280 \text{ Kbps}$$

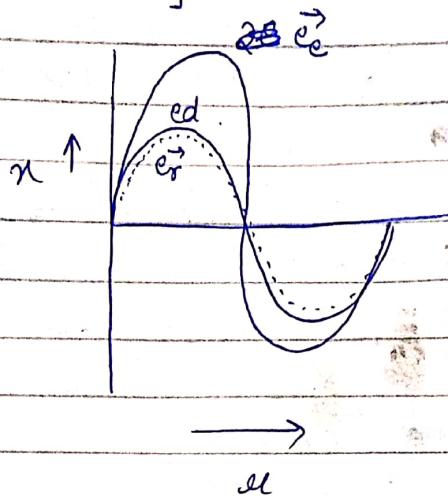
use hata hai

mp3 compression lgaa ke bhejo.

12/03/19

## Multi-Path Fading

Graphically



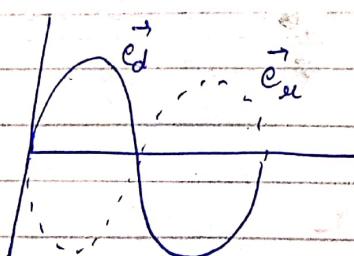
Phase diff. ( $\lambda$ )

$$\vec{e}_e = \vec{E} \sin\left(2\pi ft + \frac{2\pi}{\lambda} \cdot d\right) + E \sin\left(2\pi ft + \frac{2\pi}{\lambda} \left(d + \frac{\lambda}{2}\right)\right)$$

$$= \vec{E} \sin\left(2\pi ft + \frac{2\pi}{\lambda} \cdot d\right) + E \sin\left(2\pi ft + \frac{2\pi}{\lambda} d + \pi\right)$$

opposite  
phase  
  
 -      ~~=~~  
 negative  
hypo

$$\vec{e}_e = 0.$$



Take  $d$

$$d_1 = \lambda/4$$

$$d_2 = \lambda/2$$

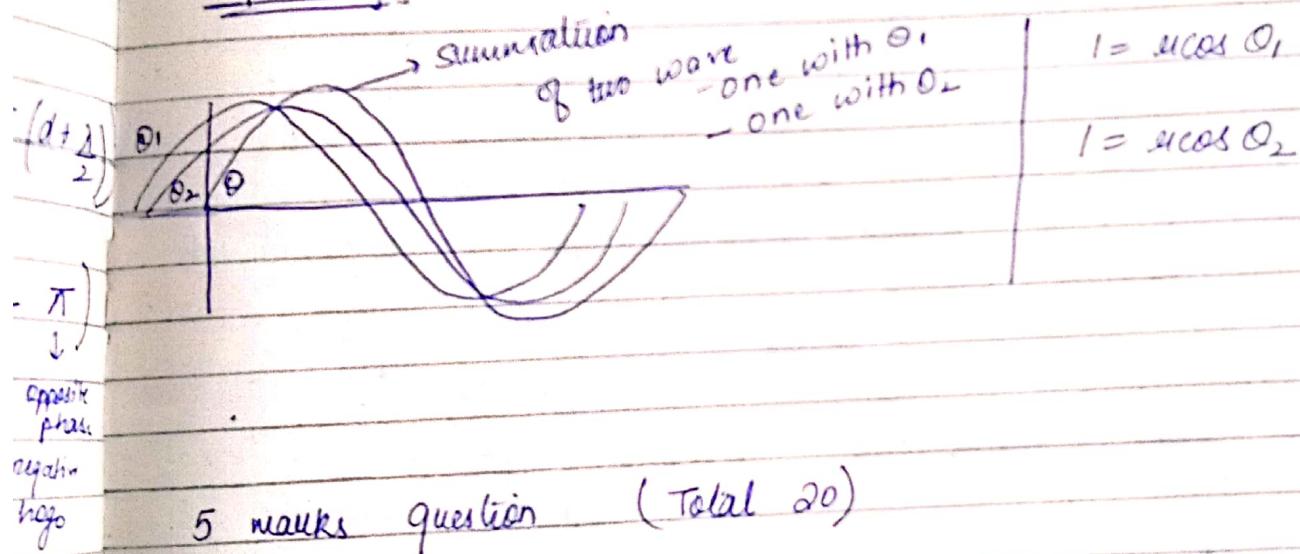
what will be the  
effective wave  $\vec{e}_e$

Q. In a mobile comm., the receiver 3 waves with these phase differences. Comment on multi-path fading.

Summation of two sin wave is a sine wave

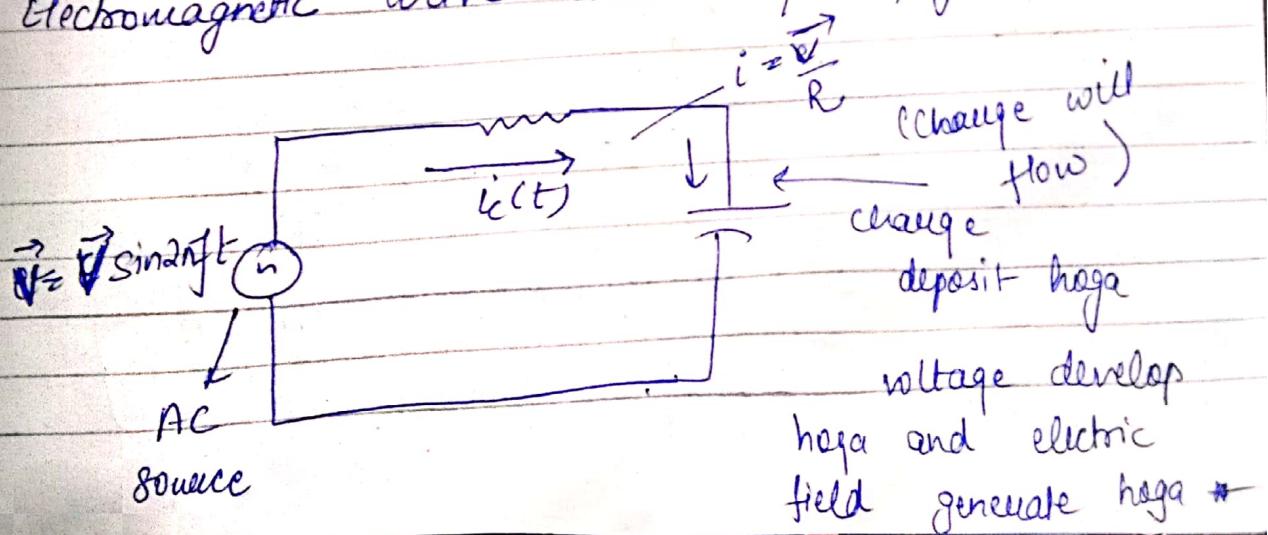
$$= \vec{E} \sin(\omega ft + \phi_1) + \vec{E} \sin(\omega ft + \phi_2)$$

Graphically



Choice (any 4)

Electromagnetic Wave Generation/ Propogation



Initially the capacitor is not charged during switching on. (no voltage)

\* varying electric field generate hoga.

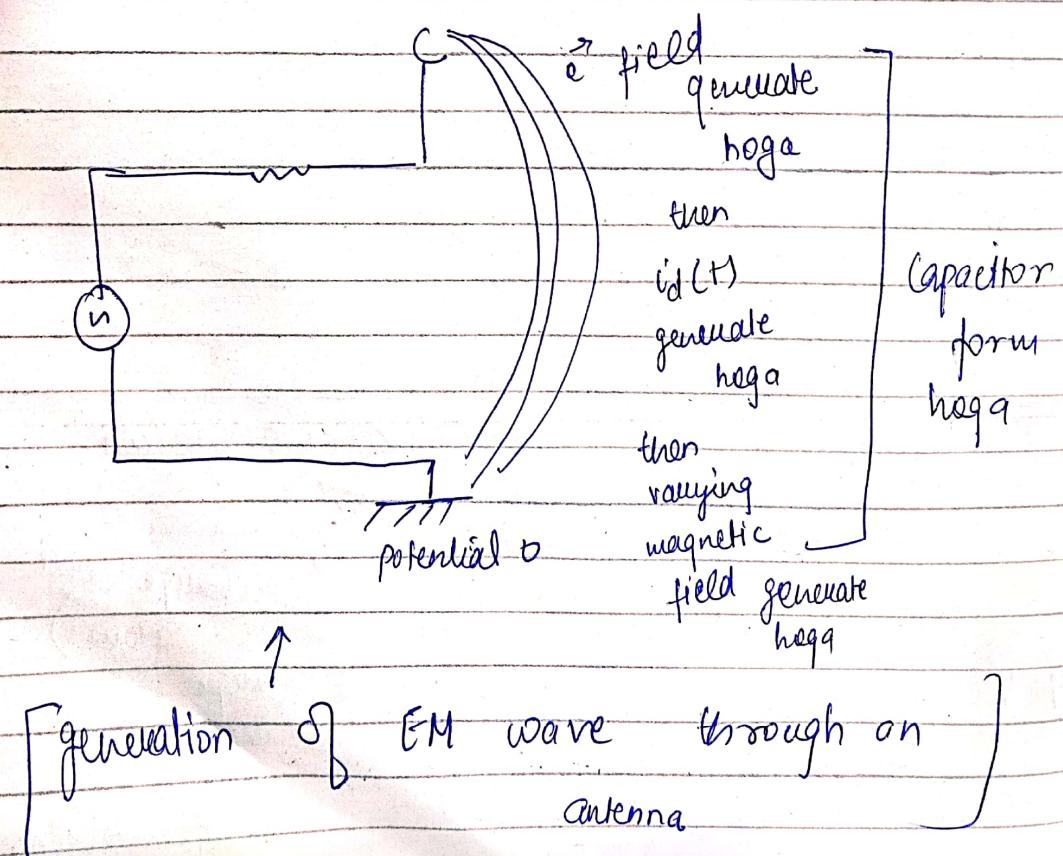
current through capacitor is displacement current-

phase diff. b/w voltage & current.

$$i(t)$$

and current flowing through all resistor is conduction current.

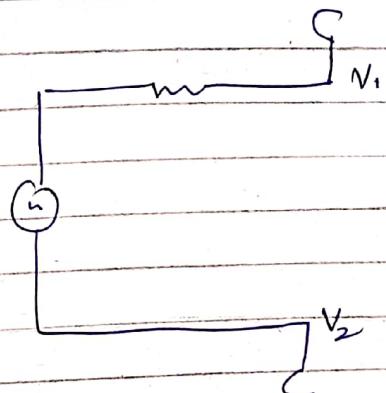
varying voltage, varying current, varying displacement current



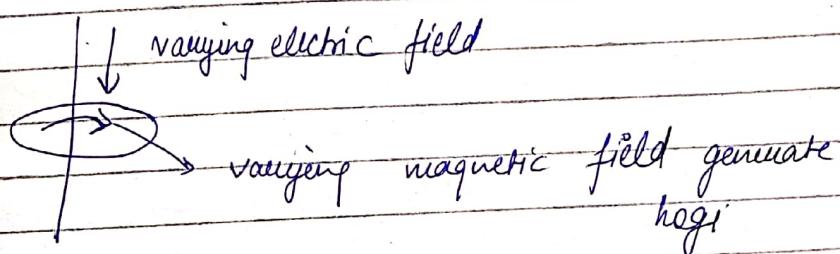
$$V = \frac{Q}{C}$$

$C \rightarrow$  this tends to  $\infty$

voltage tends to be 0.



potential difference  
is  
 $V_2 - V_1$



Faraday laws.

Q State how the E-M wave is generated by  
an antenna  
or received by an antenna.

wave equation

To derive

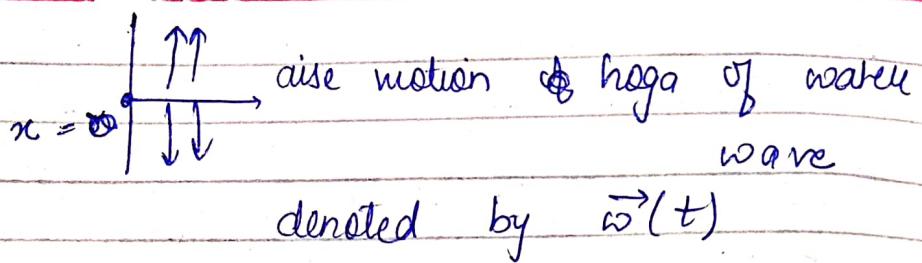
these we require

$$\vec{E}(t, x) = E \sin \left( 2\pi f t + \frac{2\pi}{\lambda} \cdot d \right)$$

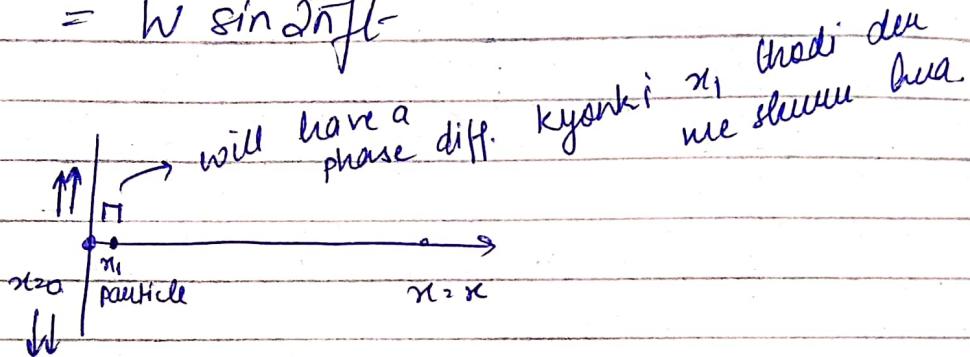
the analogy of water waves

$$\vec{H}(t, x) = H \sin \left( 2\pi f t + \frac{2\pi}{\lambda} \cdot d \right)$$

$\alpha$  direction



$$\vec{\omega}(ft) = W \sin 2\pi ft$$



$\Rightarrow x_1$  will transfer energy to  $x_2$   
and  $x_2$  <sup>ka</sup> harmonic motion hoga  
bt thodi deu me hoga

|| sly  $x_3 \dots$

then

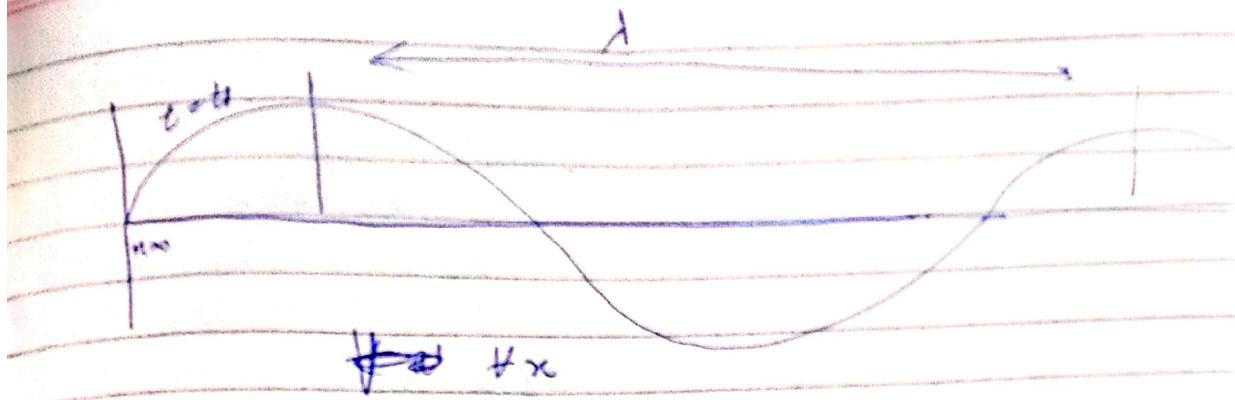
$$\vec{\omega}(t, x) = W \sin(2\pi ft - \phi_x)$$

water  
wave

eq<sup>n</sup> will be  
function of

$t$  &  $x$

kyunki  
particle  
ka motion  
bad me  
shuru hogi  
to run  
hoga.



at  $t = t_0$  wave change ho jaiga

So.

$$\vec{\omega}(t_0, x) = W \sin(2\pi f t_0 - \phi_x)$$

$$\lambda = 2\pi$$

$$l = \frac{2\pi}{\lambda}$$

$$x = \frac{2\pi}{\lambda} \cdot n = \phi_x$$

So.

$$\vec{\omega}(t_0, x) = W \sin\left(2\pi f t_0 - \frac{2\pi}{\lambda} x\right)$$

~~Distance between two consecutive crests~~

T time me  $\lambda$  distance ho rahi hoga

$$l = \frac{\lambda}{T}$$

$$v = \lambda/T \Rightarrow \boxed{v = f\lambda}$$

Capacitor value change karo oscillator se  
different frequency generate hogta

(Tuning circuit)

Speed of EM wave = Speed of light  
~~in space~~

in space, aerospace  
& metal

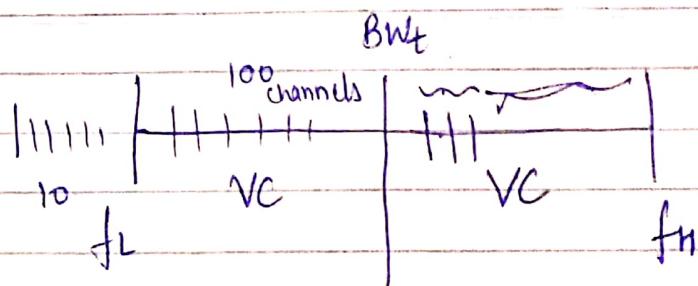
space me  $3 \times 10^8$  m/sec

aerospace me km hujata hai

metal me aur km hujata.

27/3/19

FDMA | FDD



110 channels  
are allocated

110 channels  
are allocated

Uplink - call request

downlink - paging slot, grand channel

BW<sub>UL</sub>      BW<sub>DL</sub>

110 channels  
are allocated

(anybody can)  
access

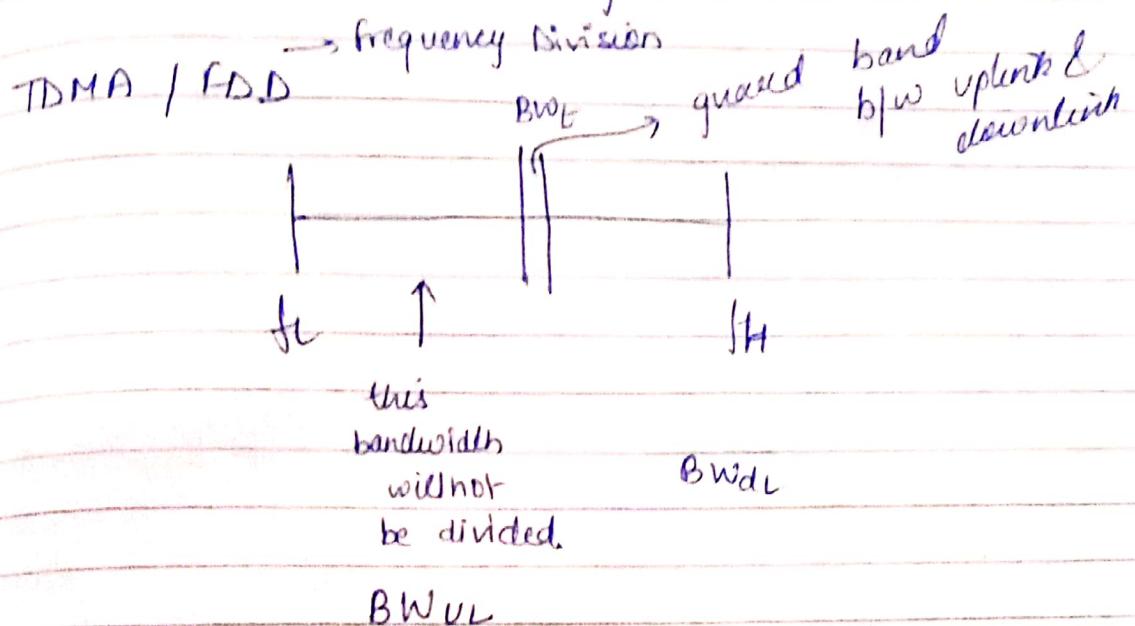
within  
which  
100 channels for  
voice &  
10 for control control  
slot

voice channel are created from  
normal voice & frequency slots

and control channel are created from  
mini frequency slots

## TDMA / FDD

The channel allocated frequency slot allocated to source can be used any time

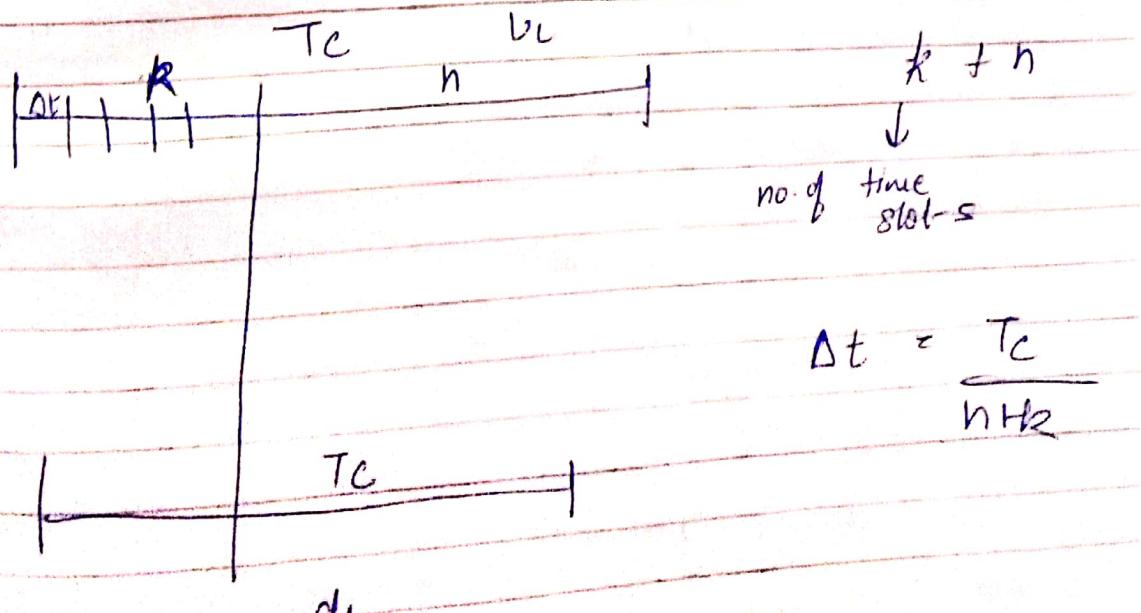


Here, bandwidth is divided into  $BW_{UU}$  &

$BW_{DL}$

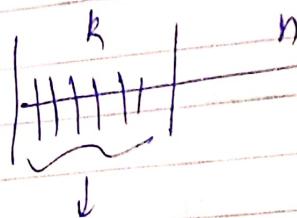
but not further divided.

Uplink and downlink have different frequency slots



$n$  slots will be allocated created as voice channels (uplink)

Then  $k$  to further divide into slots me.



where  
further  
divided

it become

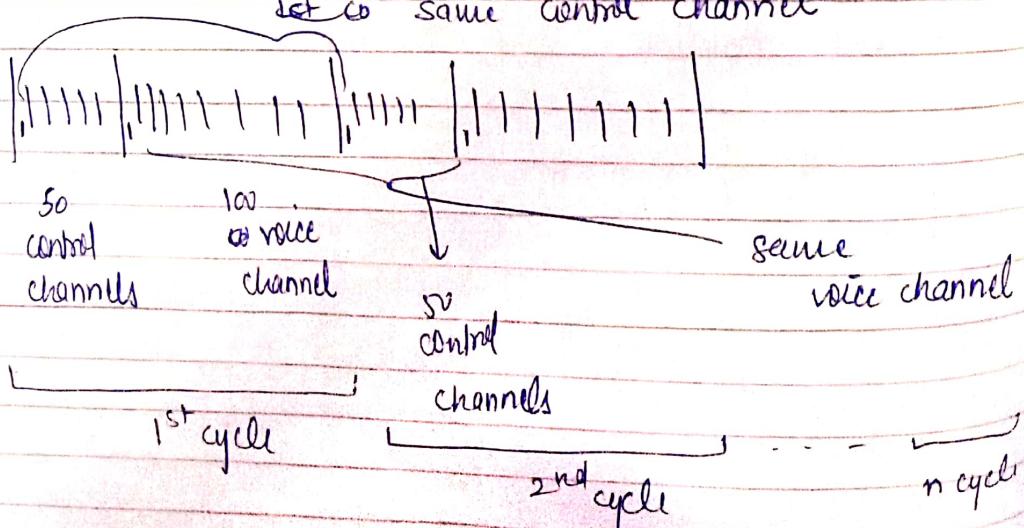
$10^k$

↓  
no. of slots

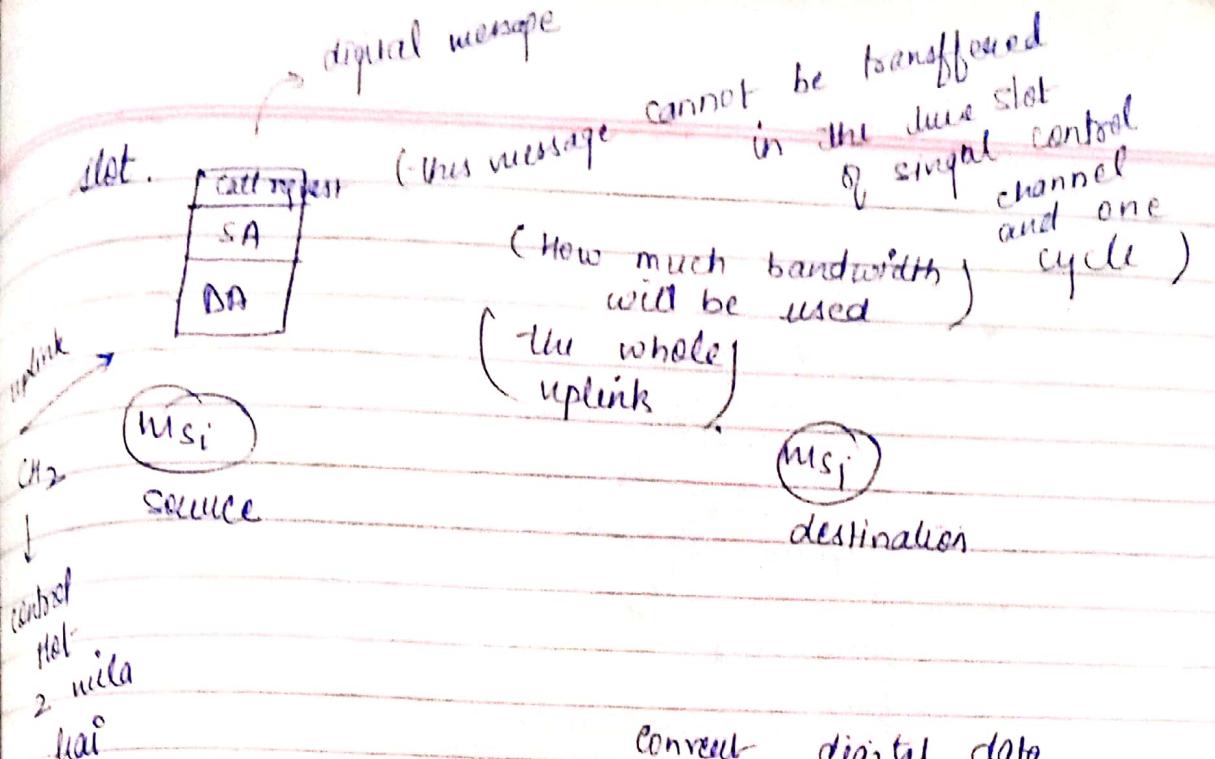
Called mini-time slots

Similarly, for downlink,

set to same control channel



Any one allotted 1<sup>st</sup> control channel (time slot)  
Send the data for 1<sup>st</sup>. then can only send  
data on the 1<sup>st</sup> control channel on next



ROM me  
built in  
beta hai  
control slot  
time calculate  
kavange when  
the time  
with start

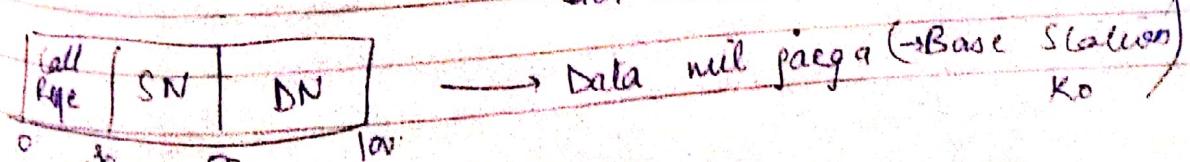
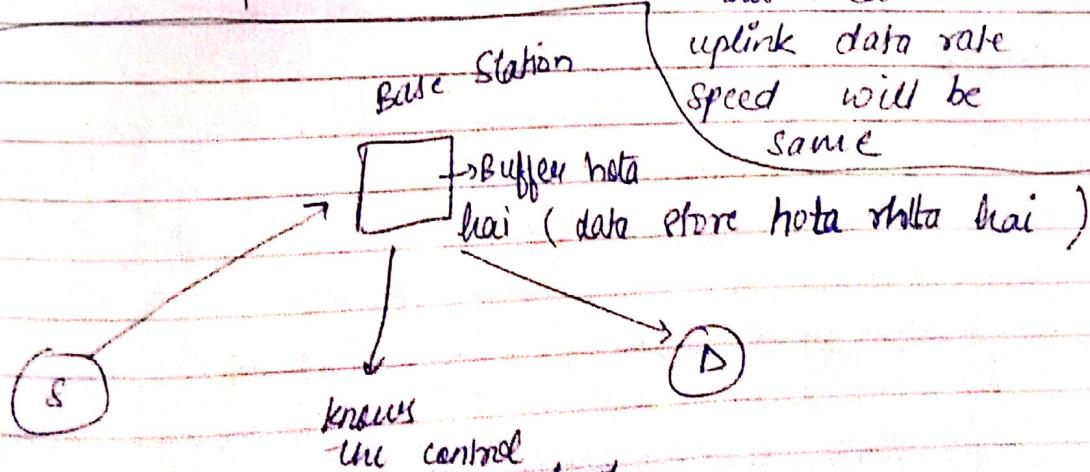
has data  
every message  
is send with the  
same speed or for a  
mini time slot.  
depend on  
the uplink  
bandwidth

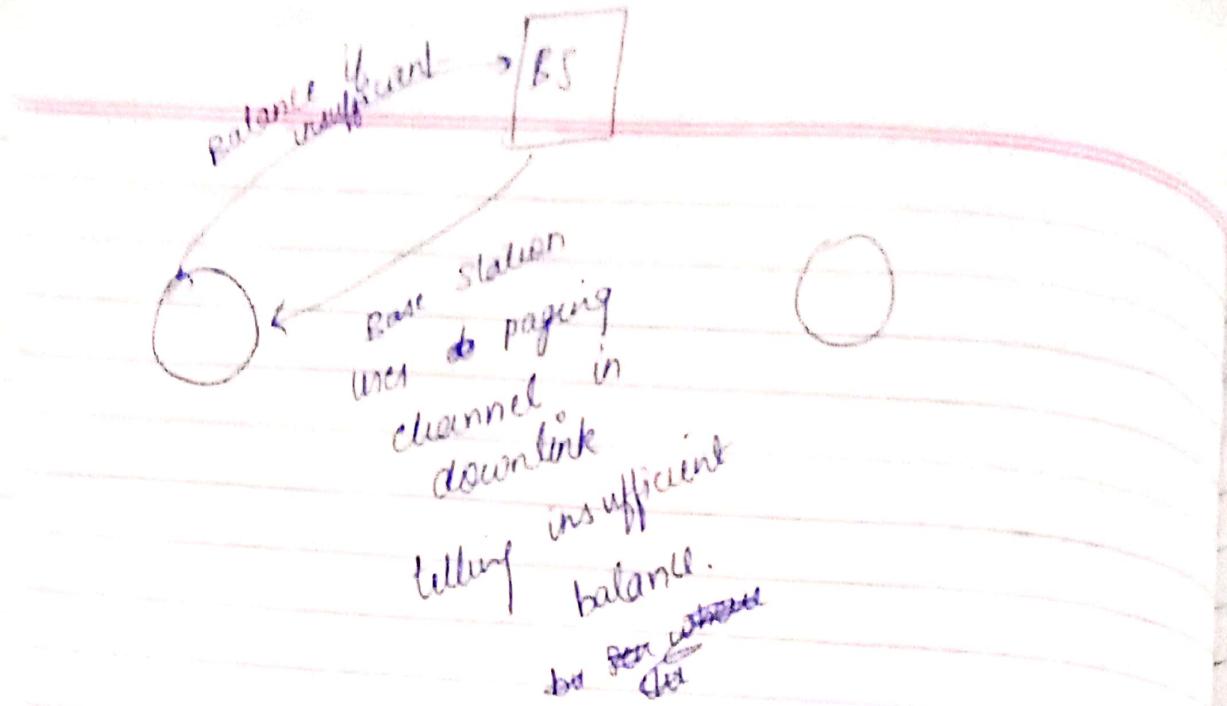
convert digital data

d  
analog  
using (PSK, QAM)

analog bandwidth  
= baud rate.

this means  
uplink data rate  
speed will be  
same





If destination is busy, then also a paging channel will be used to send to sender that destination is busy.

If not busy, sufficient balance then need to allocate voice channel to both sender & destination

check karega

1 pair of uplink and downlink voice channel hai ki yeh ni

Grant channel

If yes

Sender pe paging channel pe

Call successful message hogi

phir grant channel pe - ?

Destination pe paging channel pe message hogi

call coming.

(E)

4 6  
V d

allocated  
to sender  
and desti.

(D)

4 6  
V d

when A



teek signal → to start the transmission  
for uplink

(base station  
midea)

↓ | 1 | 2 | 3 | 4



(4-1) ΔT

after this time

date rate increases  
time cycle reduces.

transmission  
start using  
the speed  
of uplink  
channel  
only

Source & Destination ka  
clock synchronization

(5-1) ΔT

transmission  
start  
using the  
speed of  
downlink  
channel.

FDMA me data speed will  
equal to the frequency slot-

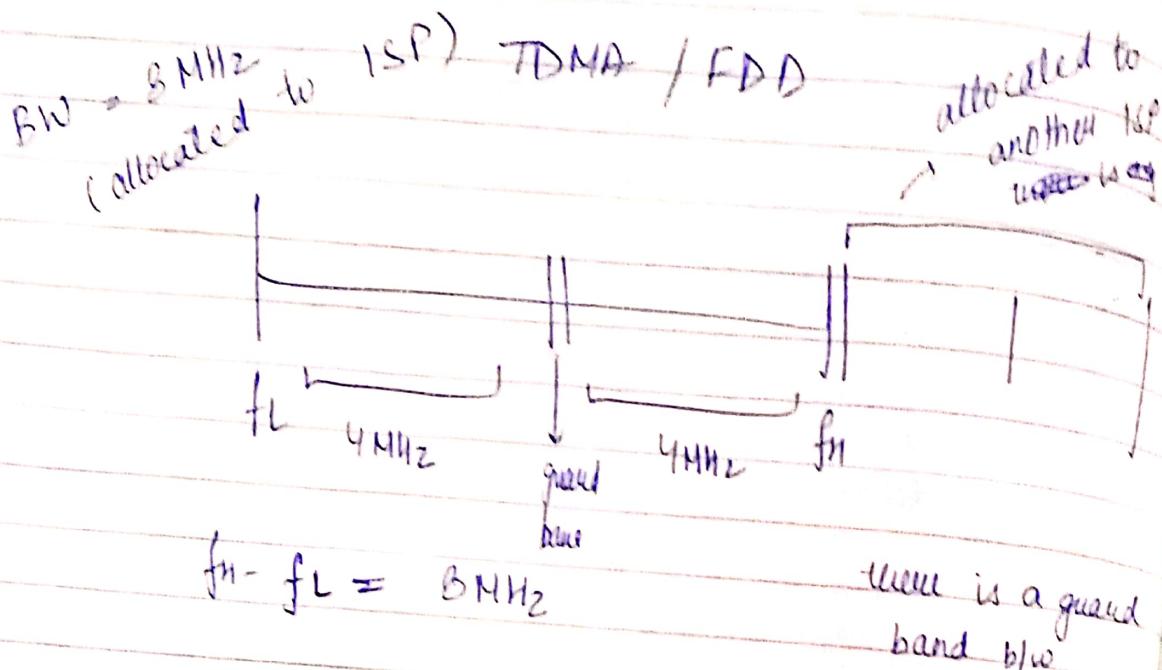
allocated

but in TDMA

it will be the complete  
uplink | downlink.

(But for ΔT  
time)

## No. of Uplink & Downlink Channels



ISP cannot transfer

more than the ~~part~~ bandwidth allocated.

Using 16 QAM  
Arrange no. of levels

Each channel is speed

$$CH = 10 \text{ Kbps}$$

But they will be transmitting with the speed of uplink bandwidth.

No. of uplink channels

$$\text{Average speed of channel} = \frac{\text{Uplink bandwidth}}{n}$$

$$BW_{UL} \rightarrow 4 \text{ MHz}$$

$$BR_U \rightarrow 4 \text{ MBaud/sec}$$

$$\text{Bit rate} = 4 \log_2 16$$

$$= 4 \times 4 = 16 \text{ Mbps}$$

# Emerging Trend in Computer Networking

PPT

96

3

16 group

Min - 2  
Max - 3

ROLL No - Name Group No)

Six ko bana

$$\text{Average speed} = \frac{16 \text{ Mbps}}{n}$$

$$16 \text{ Mbps} = 16 \times 10^6 \text{ bps.}$$

$$\begin{matrix} \text{1 channel} \\ \text{speed} \end{matrix} = \frac{10 \times 10^3}{\text{---}}$$

Total is

$$\text{No. of uplink} \cdot \frac{\text{Total speed}}{\text{1 channel speed}}$$

$$= \frac{16 \times 10^6}{10 \times 10^3}$$

$$= 1600$$

Pehe

uplink & down link same at the time.

Tc

uplink

frequency

same di thi

Tc

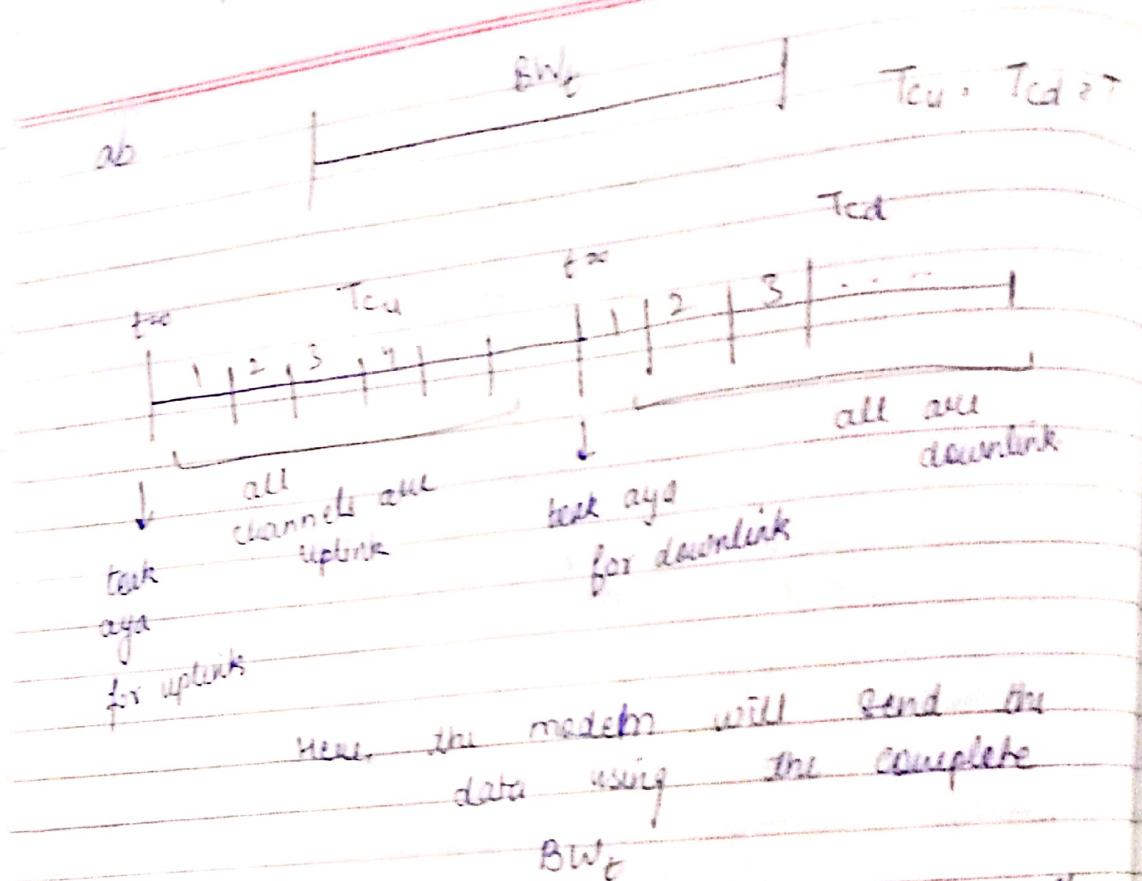
downlink

task aya deno salt shuru ho gye.

overlap kare the thi.

TDMA | TDD

re division  
of bandwidth

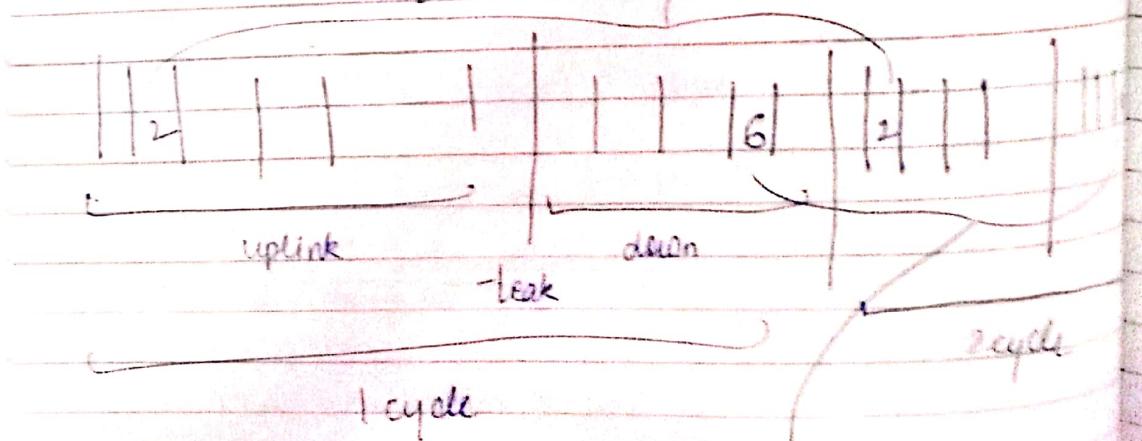


tit source has

$$u_c = 2$$

$$d_c = 6$$

same channel source  
not use for send  
data send  
knowledge



isti re hi  
sent with  
receive data

speed double ho gya.  
channel come ho gya.

$$\left. \begin{array}{c} \text{BWT.} \\ \hline \end{array} \right\} \quad \text{BWT} = 8 \text{ MHz}$$

$$BR_t = 8 \text{ MB/sec}$$

~~bit rate~~  
bit rate =  $8 \times \log 16$   
 $= 8 \times 4 = 32 \text{ Mbps}$

No of channels =  $\frac{32 \times 10^6}{40 \times 2^n}$

total  
channels  
(uplink &  
downlink)

Only for uplink =  $\frac{32 \times 10^6}{2^n} \times \frac{\text{No. of channels}}{2}$

~~No of channels~~

FDMA me bhi

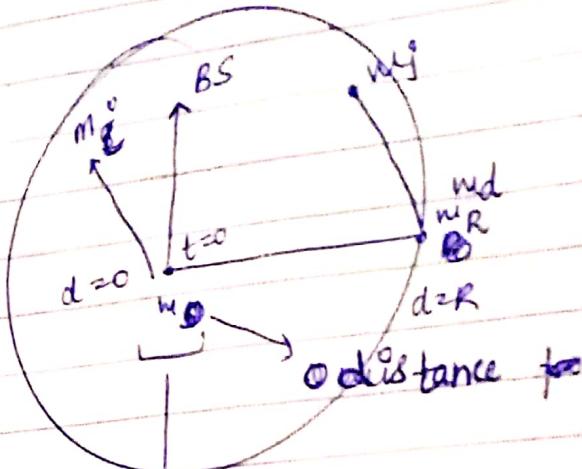


have channel  
ke beech guard  
me bhi r frequency  
plot tha

b/w uplink &  
downlink  
me bhi

## TDMA

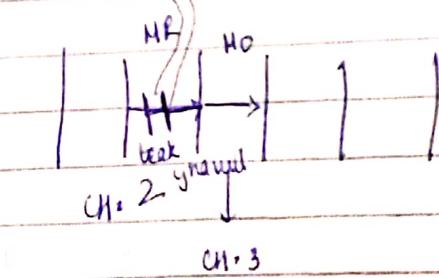
minimum  
will calculate  
noga



is ko teek

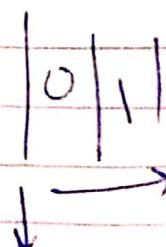
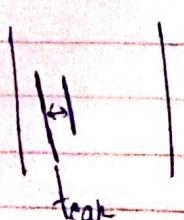
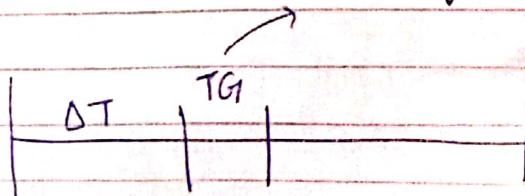
jdi mil  
fay a

$t=0$  pe calculate karte 2 se



$v = \text{EM wave}$   
signal  
speed

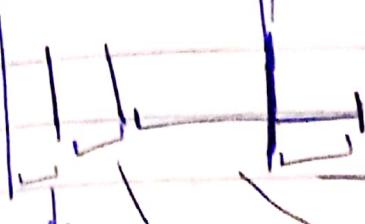
$$TG_1^2 \frac{2R}{v} \quad TD = \frac{R}{v}$$



CH<sub>1</sub> L

MR.

yha pe work will go  
CH<sub>2</sub> G and  
Mo transmission bhi & yhi se share  
hoga



yha

leak

mila

BS se

phir

trans m

iske

bad

transmission

share

hoga

isme

transmission

hoga to the

BS

to and

since

Mo

Ethall BS RO  
Mo & MR

ka

data

mitlega

A

influence

hoga

29/3/19

## CDMA / PDD

Code Division Multiple Access

SDMA

- 8

Space we bandwidth cells have

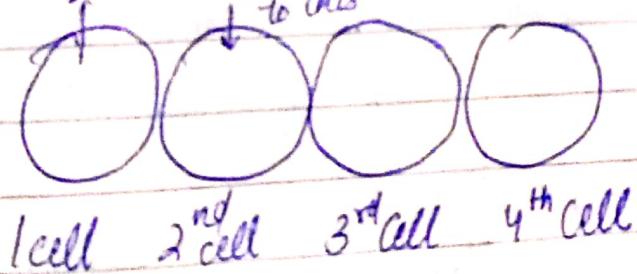
Bandwidth allocated to this cell

will not be

allocated to  
this cell

Bandwidth allocated  
to this is not

allocated to this



Power does not  
reach this

cell, so that

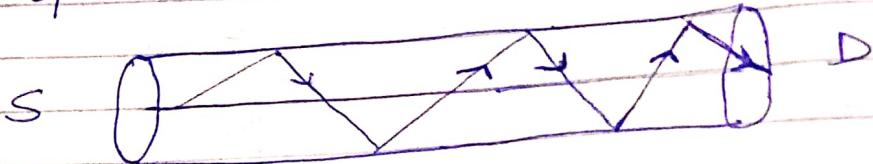
bandwidth can

be allocated

to 3<sup>rd</sup> cell.

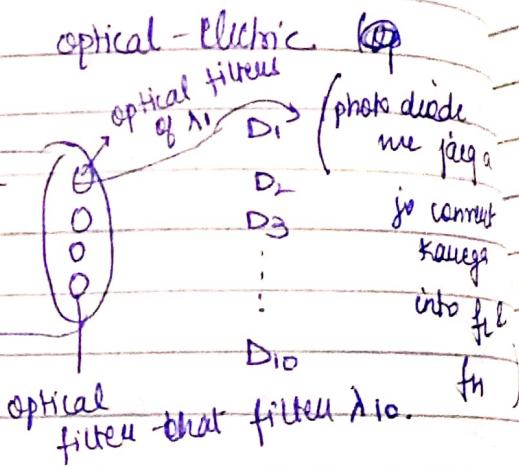
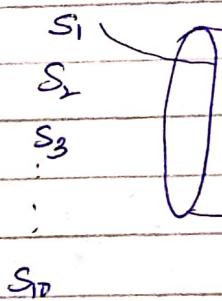
Optical fibre me hum wavelength ki bast karte hain.  
we say we are communicating using wavelength  
(It involves →)

Optical Fibre



$\lambda$  length ka signal bheya (i.e. one color)

electric-optical (laser)



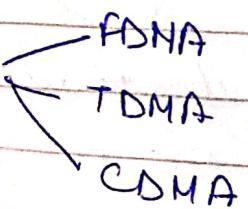
Each source will send signal  $S_i$  with wavelength  $\lambda_i$ ,

then  $S_1$  is  $\lambda_1$

all signals with different  $\lambda$  will mix and at destination signals we will separate signal with their resp.  $\lambda$ 's.

(using optical filters)

Mobile Communication



## CDMA / FDD

In this, we have 'n' codes

A B C D ...  $n^{\text{th}}$  code (uplink)  
i.e. these codes are transmitted using  $BW_u$ .

A B C D ...  $n^{\text{th}}$  code (downlink)  
i.e. these are transmitted using  $BW_d$ .  
11 bit Code

$2^n \rightarrow 2048$  channels

↳ 1848 as voice channels  
200 for control channels

## European Standards

1G - FDMA | FDD - AMPS  
2G - TDMA | FDD - GSM  
TDMA | TDD

2G - CDMA (USA Standard)

3G CDMA - WCDMA  
4G CDMA  
5G CDMA

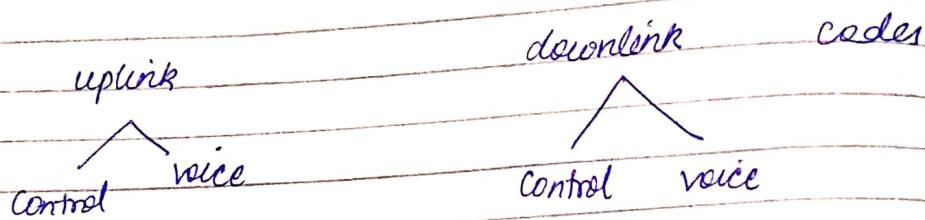
2G1

CDMA

$m$  bits for code (both uplink & downlink)

$2^m$  codes

(all codes will not valid)



let a code  
8 bits.

A  $\rightarrow$  10101100

(code)

B  $\rightarrow$  11001101

data ki bits  
called logical  
bits 0,1.

let A & B be uplink codes  
for S1 station & S2 station

To send 1 bit, S1 has to send

~~example~~ A code A

then each bit in code is  
call sub-bit on chip

i.e. 8 chips need to be transferred to  
send logical bit 1.

To send 0 we will send  $\bar{A}$

similarly,

will send \*

B to send 1 = B

to send 0 =  $\bar{B}$

chip

1

0

1 0 1

SI  
(A)

SL  
(B)

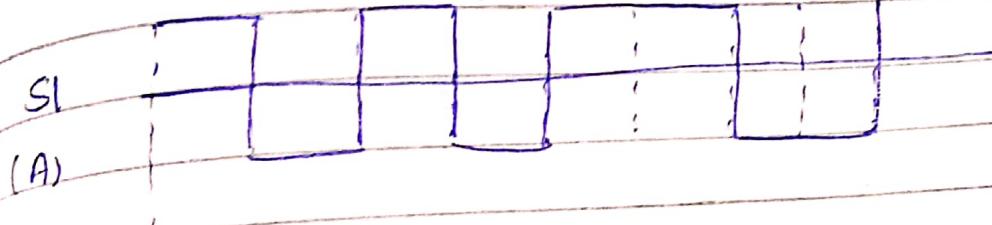
Space  
 $IV + OV$   
 $IV = 0$   
 $OV = 1$

chip

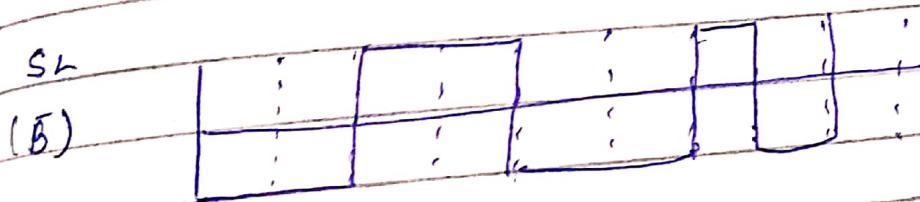
$$1 = +1V$$

$$0 = -1V$$

1 0 1 0 . 1 1 0 0 .

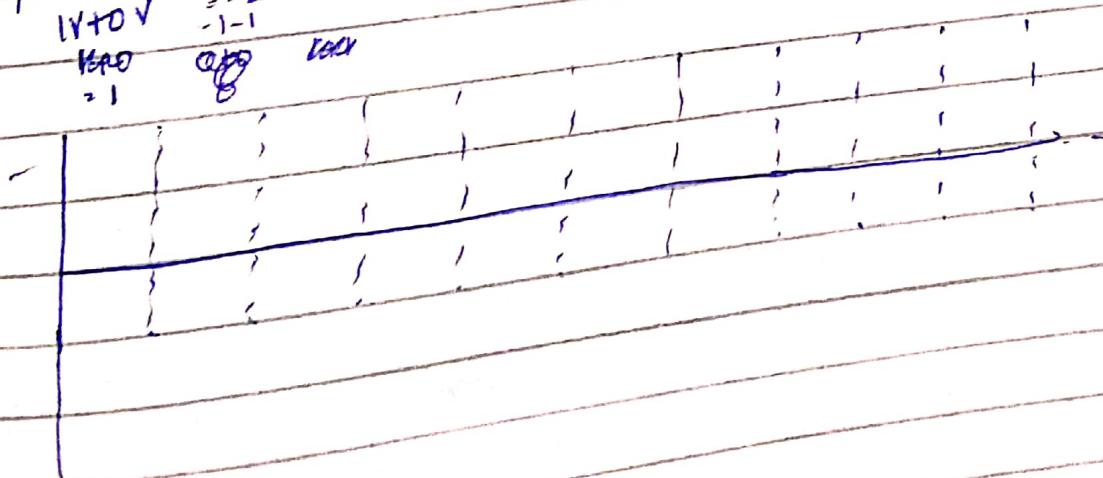


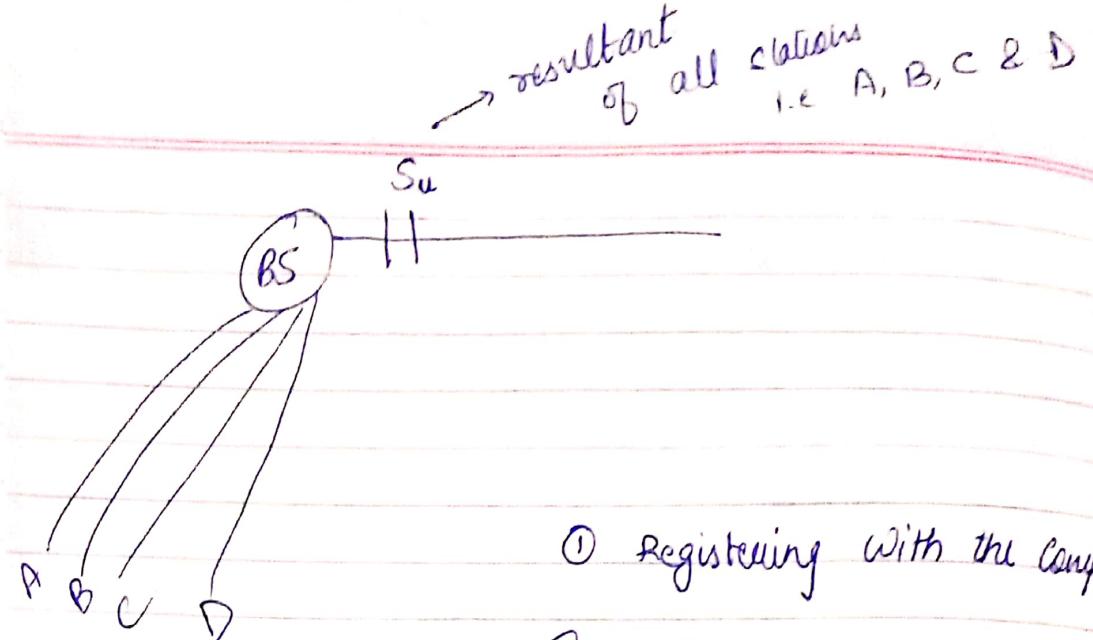
0 0 1 1 0 0 1 0



S1 & S2 will be combined.

Space      we  
WTOV       $\frac{1}{1-1}$   
HAD      off  
=1      rear

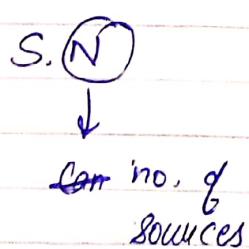
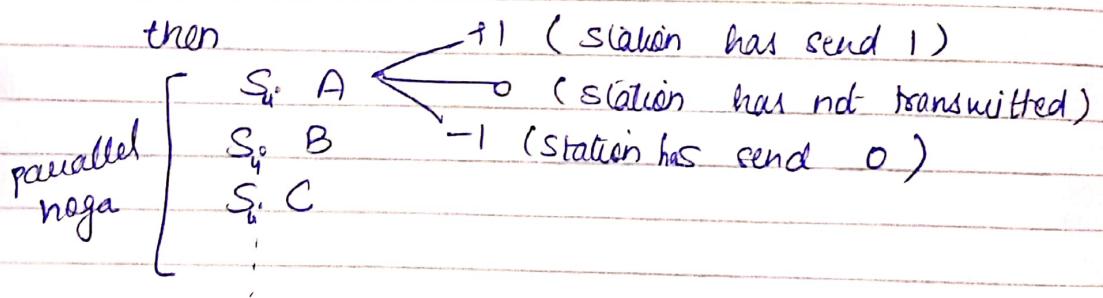




① Registering with the Company

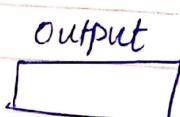
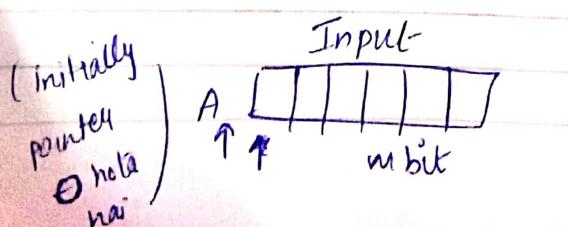
② Switch ON the phone  
7 base-stations around  
you contact you  
and the BS that  
has the highest  
signal strength.

i.e. you send the acceptance  
i.e. you registered with the  
base station.



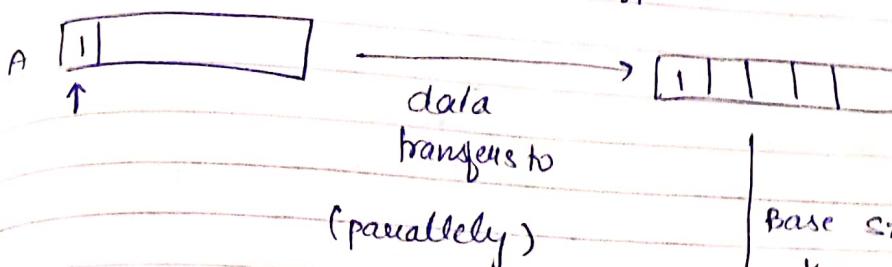
If the stations  
hang unlike the  
buffers hang

input      output  
buffers



DATA STATION A

Station ~~L~~ has send A



Base Station calculates  
Su i.e. uplink.

H | Base Station  
knows A  
to output  
connection  
H ko jana  
hai

Now after

Now H will be receive the data  
that the BS will send but H will  
has his own code so send kahega.

so Sd calculate kahega.  
sb receive ka resultant hoga.  
aur bhej dega

Sd. A  
Sd. B  
Sd. C

m bits

$$C = \{0, \dots, 2^m - 1\}$$

all set  
of codes  
not all codes will be valid

$$C_r \subset C$$

## Orthogonality Property

$$\textcircled{1} \quad S \cdot T = \frac{\sum_{i=0}^{m-1} S_i \cdot T_i}{m}$$

To. Two codes are valid

A code is valid when

Pick a code let  $S$  from  $C$ .

and then two a dot product with  
rest of the codes

and if

$i = \text{rest codes in } C$

$$S \cdot T_i = 0$$

then  $S$  is a valid code

if

$A \bar{A} A \bar{A}$

$1 0 1 0$

(4)

Input

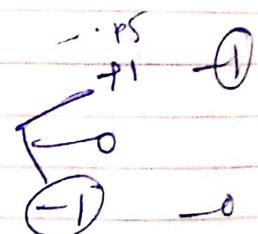


(1)  $\bar{H}$

$H$	$\bar{H}$	$H$	$\bar{H}$
-----	-----------	-----	-----------

$G \bar{G} G \bar{G}$

$Sd.H$



$Sd.H$

$$\begin{matrix} H & + & \bar{H} & + & H & + & \bar{H} \\ G & + & \bar{G} & + & G & + & \bar{G} \end{matrix}$$

6/4/19

## CDMA / FDD

2 conditions

- ① 1.) Send A and  $\bar{A}$
- 2.)

Source      Code      1 bit

A	A	1
B	B	0
C	C	1
D	D	X

$$S = S_0 + S_1 + \dots + S_7$$

sum of voltage of all sources (jo jo kise under  
me register hoga  
hai)

S generate

Now we need to decode A by S.A

+1 aya to buffer me 1

0 aya to buffer me 0.

-1 aya to buffer me kuch ni hoga.

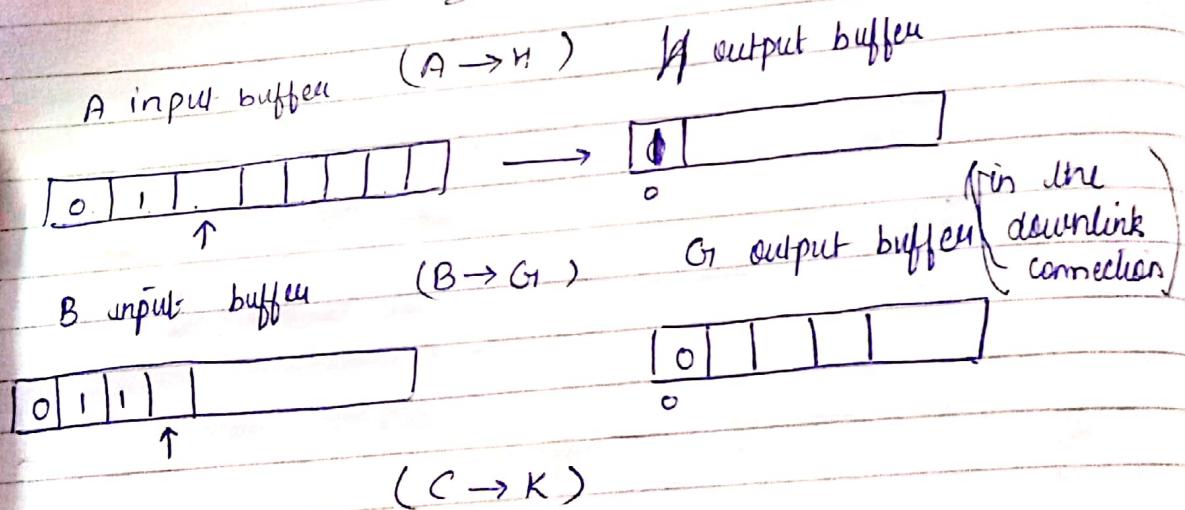
$$\underline{\underline{S}} = A + \bar{B} + C$$

Everyone has uplink buffer / downlink buffer

$$\begin{aligned}S.A &\rightarrow (A + \bar{B} + C) \cdot (A) \\&= A \cdot A + \bar{B} \cdot A + C \cdot A \\&= 1 + 0 + 0 \\&= 1\end{aligned}$$

A ke buffer logic 1 daal denge.

$$\begin{aligned}S.B &= (A + \bar{B} + C) \cdot (B) \\&= A \cdot B + \bar{B} \cdot B + C \cdot B \\&= 0 + (-1) + 0 \\&= -1\end{aligned}$$



Now, we need to transfer first bit to H &  $G_1$ .

& downlink signal banenge.

$$S_d = \underbrace{\bar{H} + \bar{G}_1}_{\text{1 bit code}}$$

downlink Tick dia

$S_d$  ko downlink me transfer karne denge (All stations will store  $S_d$ )

BS ek hai for enry 1 iski parallel

Ab substitutions decode kawenge.  
to sb apna apna kawenge.

A A kuch ni bheja

B

$$S_d \cdot A = 0$$

A downlink buffer  
1 0 1 0 1 ...

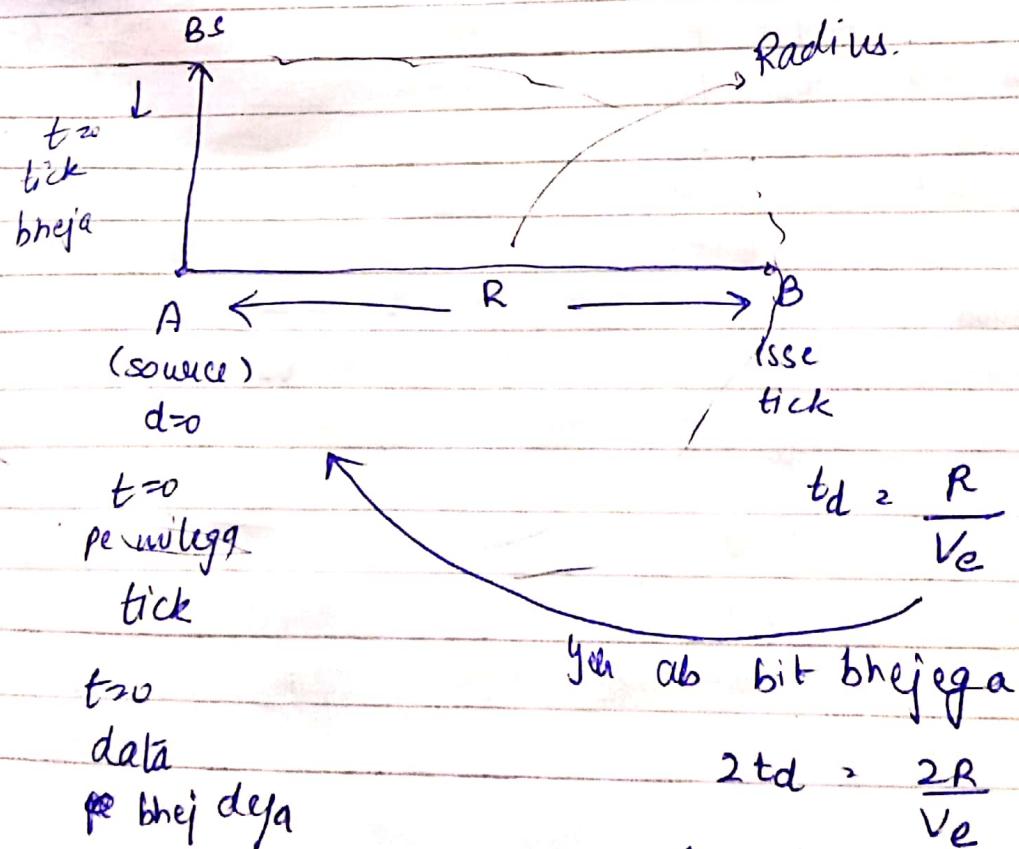
(so A will find ~~me~~ that that iske ~~bad~~ pas koi data receive ni hua )

Z.

$$\begin{aligned} H - S_d \cdot H &= (H + \bar{G}_1) \cdot H \\ &= H \cdot H + \bar{G}_1 \cdot H \\ &= 1 \end{aligned}$$

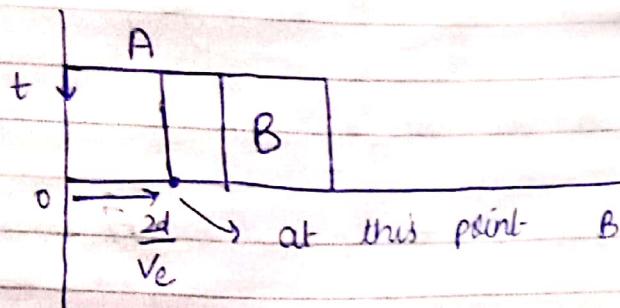
downlink buffer me store hota jaega.

→ There will be a synchronization problem.



(tick jaega data ayega)  
time double

Uplink me hi ~~det~~ problem hoga

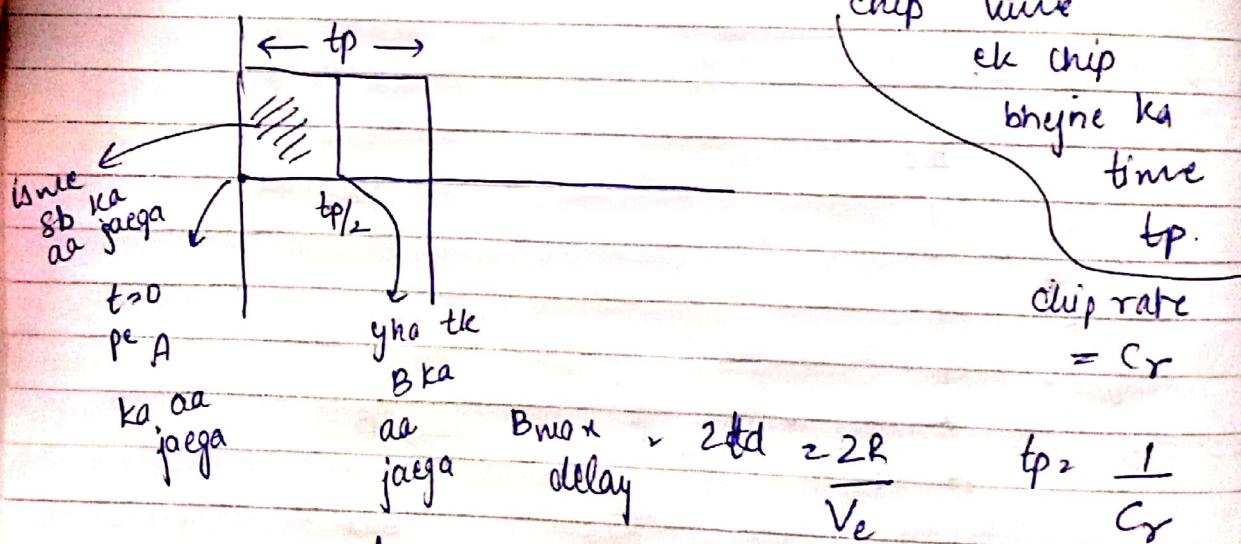


A ne +1 dia base station hoga  
B ka ni aya

to signal  $t_1 - 1 = 0$  ho jaega.  
or

to yeh galat addition hoga,

To solve this problem.



$$2 \frac{td}{Ve} = \frac{1}{Cr}$$

$$\frac{2R}{Ve} = \frac{1}{2Cr}$$

$$\frac{2R}{Ve} = \frac{1}{2Cr}$$

kai hoga.

$$\frac{2R}{Ve} = \frac{1}{2Cr}$$

$$R = \frac{V_e}{4Cr}$$

$$R < \frac{2V_e}{4Cr}$$

↓

Cell radius

$R \uparrow$

$Cr \downarrow$

to increase  
radius

decrease  
chip rate

- TDMA | FDD
  - TDMA | TDD
- ] comparison with CDMA | FDD

CDMA | FDD

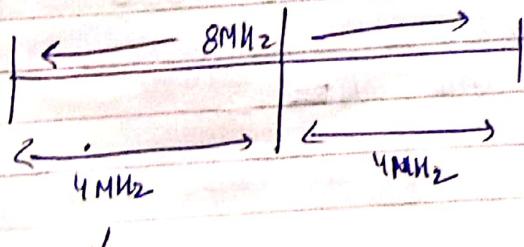
BW = 8 MHz

DAM = 16

→ bit to chips → transmission  
→ chips to mega

for Uplink

CH rate = 10 kbps



Voice channel calculations  
and we will do for uplink.

BW<sub>U</sub> = 4 MHz

BR<sub>U</sub> = 4 MBps

BR<sub>U</sub>



Clip Rate<sub>U</sub> = 4(chip / Band)  $\times 10^6 \times 4$  MCPS  
1 channel ↑ ina bhejega  
(Mega chip per second)

1 station ka  
data rate  
 $10 \times 10^3$

uplink channels  $n_u$   $\rightarrow$

bit Rate =  $n_u \times 10 \times 10^3$  bps  
(for 1 bit)

Each code is of  $m$  chips

Total uplink chip rate =  $n_u \times 10^4 \times m$

Up

Down

1 channel bit rate  $\rightarrow 10 \times 10^3$

1 channel chip rate  $\rightarrow m \times 10^4$

No. of codes

$m =$

$$\frac{10^4}{48}$$

$$\frac{16 \times 10^6}{10^4}$$

$$m = 1600$$

Total, chip codes  $\rightarrow 2^{1600}$   
no. of

( all chips will not be )  
valid

no. of valid chip codes  $\approx 2^{1600}$  ( very very less than )

no. of  
channels created.

$$v \geq 1600$$

surf 1/9

## GSM Networks.

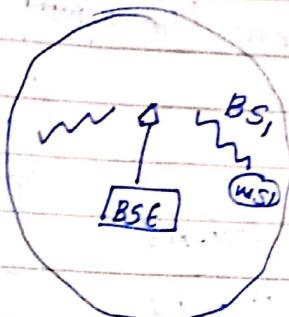
Mobile Service Controller.

MSC 1

BSC,  
Base  
Station  
Controller

BS1

BSCN



BS5

Every service provider has his own MSC

MSC has two database (many more also are)

HLR

VLR

When we approach a service provider  
sim-card will be

It has a number and IMEI

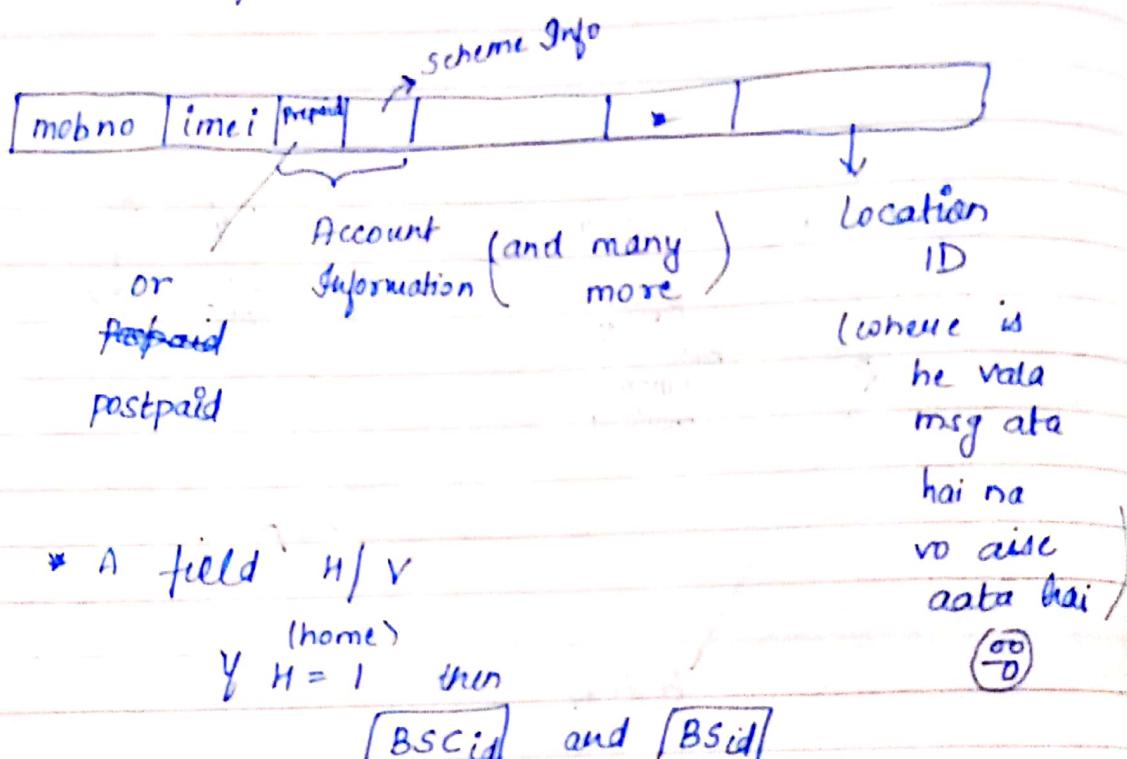
↙  
International  
Mobile  
Combined  
with  
device-no to find

An entry is created in

HLR

theft.

Delhi will have an MSC which will have an entry in HLR.



\* A field H/V

Y  $H=1$  then <sup>(home)</sup>

[BSCid] and [BSid]

also all columns.

or

$H=0$

when in another city.

Mobili will be surrounded with 7 base stations so, need to register to any one base station

(Random Access Channel ke through)

dedicate control

channel se

data location

update hoga.

BSC pas jaega apna identification add

karenge MSC me jaega MSC uss mobile se check karega ki home MSC hi hai

to entry will be

H = 1, BSid BSCid

store kar dega.

phii scheme ke according call ho skta hai ya ni.

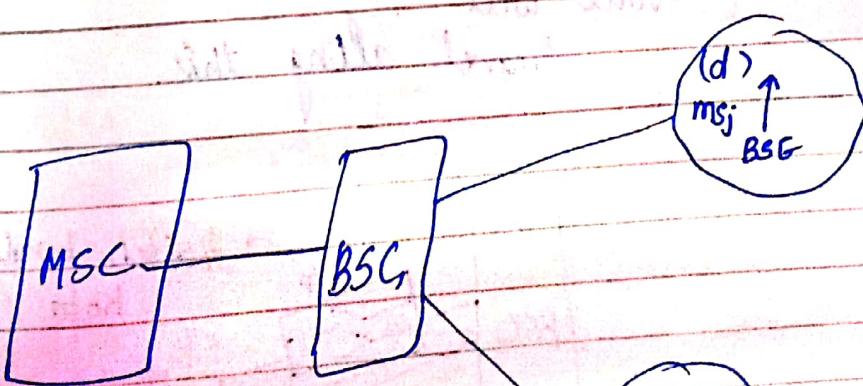
Balance agar hai to destination check kia (uska

same MSC hai ya nhi)

and also phii BSid & BSCid agar same hua to same base station hai.

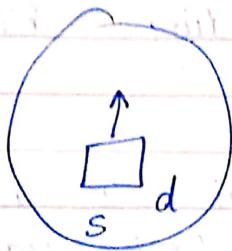
If destination is free to sender & destination ko uplink & downlink channel milega

if destination ko free ni hai to sender receiver will get a paging message that sender is busy.

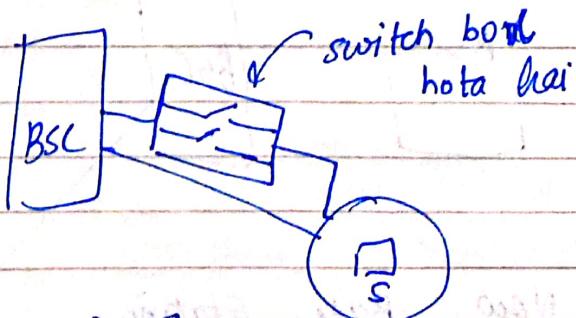
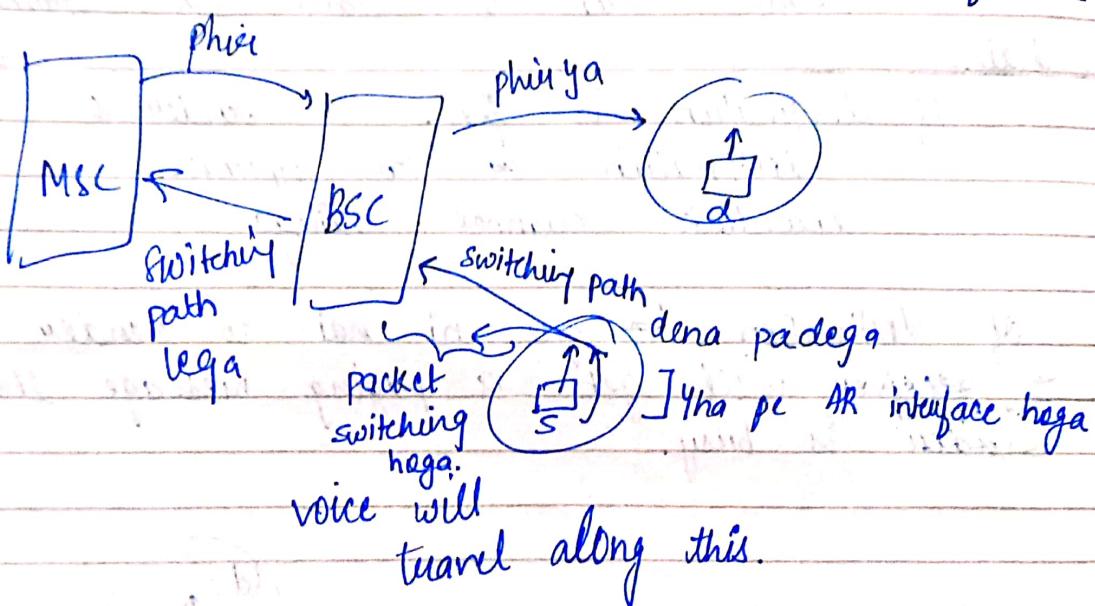


Now Base Station different hai.

To MSC check karega hone MSC  
seme hai pr different BS.  
So, check karega destination free  
hai ki ni



if source  
1 destination  
Same hai  
to. AR  
interface se



switch bord kia switch busy  
hoga —

- jb bhi data bhejna hai jisse BS to BSC  
 (i.e. switch ko reserve karne ka)  
 jb same release ho gye to unreserve  
 krdo. (agari sb reserve karne kehte hai)  
 network busy.

Disconnect ke lie

pehle AR interface will be  
 given back to BS. and switches ko  
 open kau de.

Switches can be controlled by  
 BS & BSC.

if destination is not has not same MSC



mobile number

use pta HLR

me entry

ni milegi

to VLR se

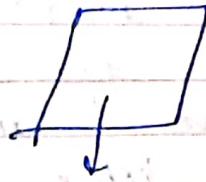
pta laega is

home MSC

VLR

Mid	Call Infor	BSC id	BS	phir to usko message bhejenge.
				↑ to home MSC bhejega.

phir home MSC



VLR

me entry ko

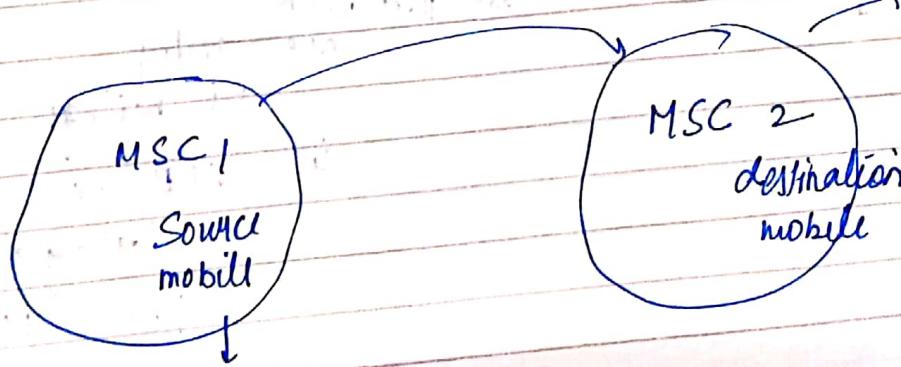
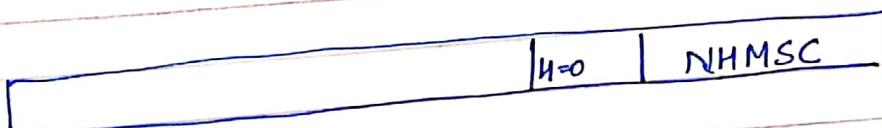
update kरेंगे

i.e.  $H=0$  का

dega aur BSCid & BS  
ki jagah

NH MSC ka

id update  
kरा देंगे



yha  
pe  
VLR me  
to ni  
milega  
kyonki  
NHSC ni

hai  
to VLR

me  
dekhega

no  
se MSC  
HLR  
me check  
karpa

to plz  
lafega Home

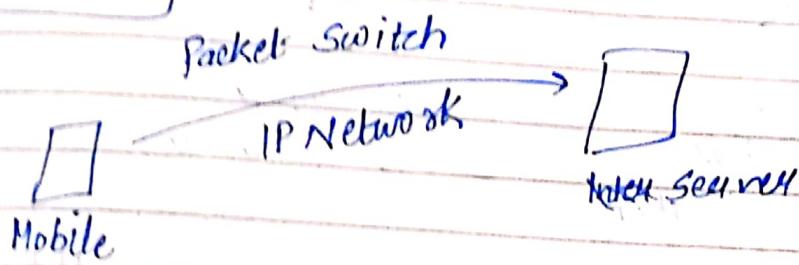
me ni

hai to  
NH MSC ~~dot~~ ki id  
~~dot~~ se data bhejega

Switch Allocation & Deallocation ki ki protocol  
hota hai signalling No. 9  
prototypical. protocol.

Now source & destination are not in  
these Home MSC.

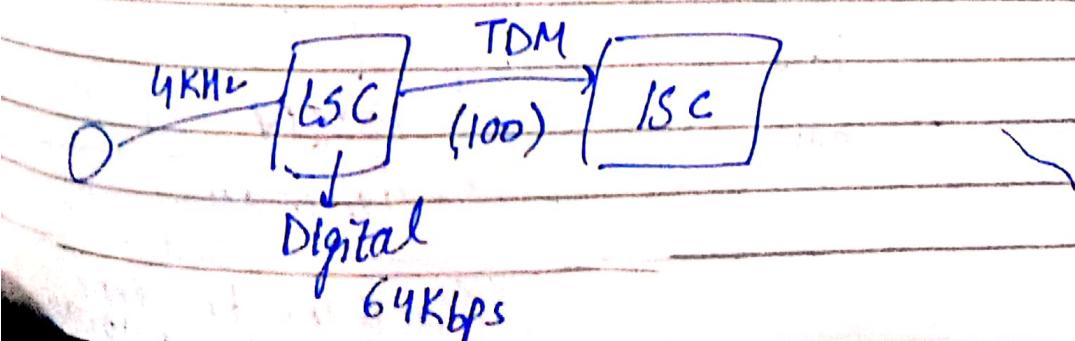
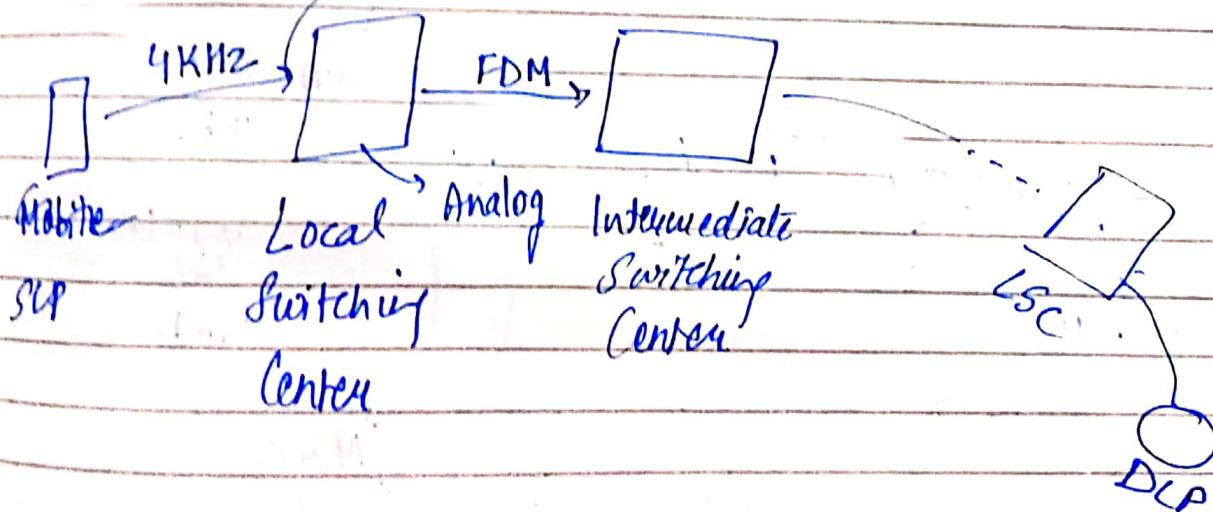
## [GPRS]



Circuit switch me telephone No.  
Packet switch me IP Address.

(landline)

analog switch tha (4KHz)

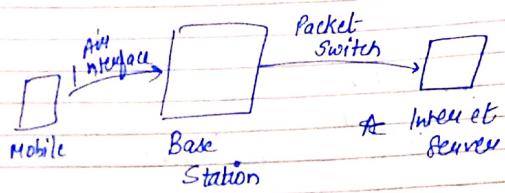


ISDN

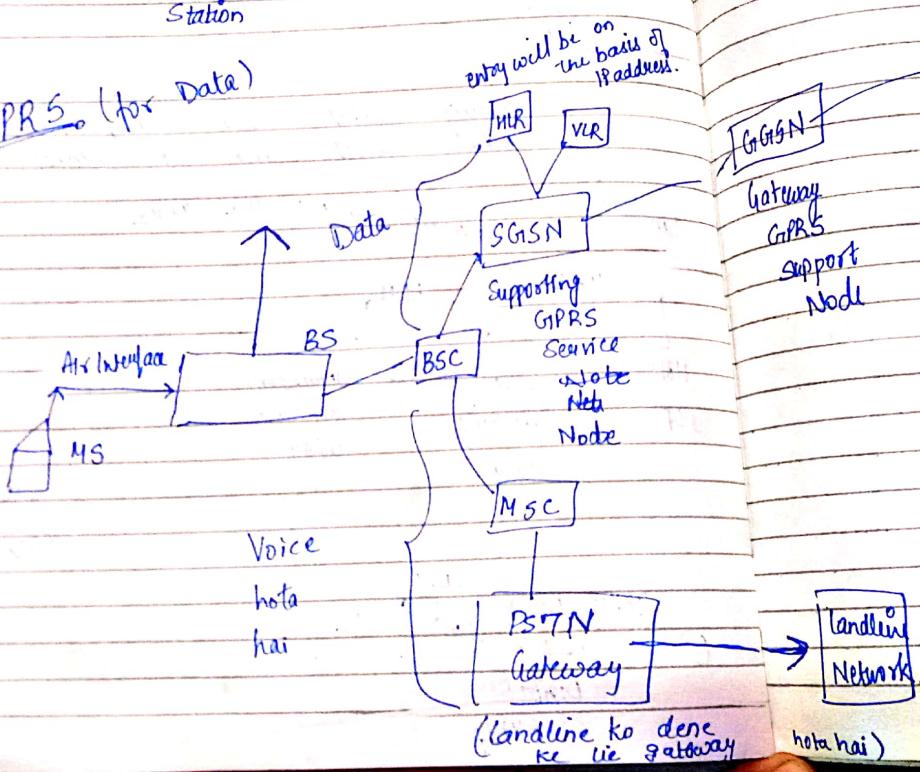
GPRS (Data Network)  
(source will be (mobile) & destination  
Internet servers)

Air

Share the same interface of GSM

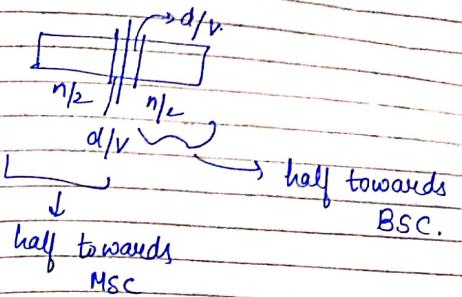


GPRS (for Data)



dt

Now wi will know ki yeh voice &  
hai ya data.



Internet

GGSN  
Gateway  
GPRS  
support  
Node

Landline  
Network

27/2/19

## Multiple Access Schemes

Centralized

Distributed

- FDMA - 1G<sub>1</sub> analog  
(obsolete)
- TDMA - 2G
- CDMA - 2G, 3G, 4G, ...

Digital

FDMA - Frequency Division Multiple Access (European)

→ AMPS (Advanced Mobile Phone Service)

TDMA (Time Div. Multiple Access)

- GSM (European)

CDMA - 2G (IS-95) - USA Std.

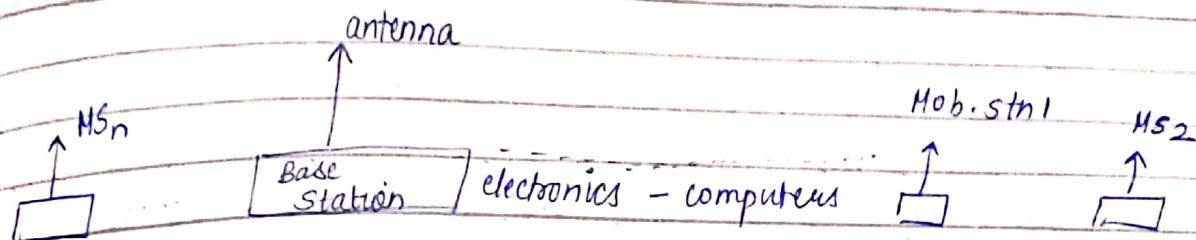
3G — WCDMA (European Standard)

— CDMA (USA Standard)

DISTRIBUTED -

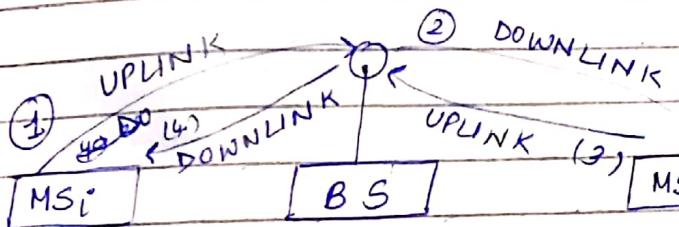
- Distributed Stations : Laptop / smartphones . LAN
- wifi (wireless) - CSMA / CA
  - Ethernet (wired LAN) - CSMA / CD

## ① FDMA (1G analog)



if 2 mobile stations are to communicate ;

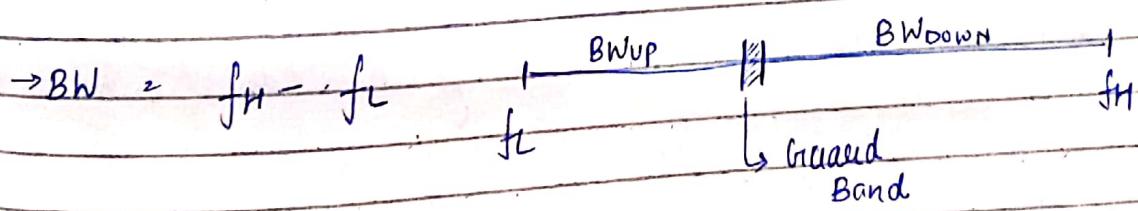
$MS_1$  communicates with Base Station - UPLINK channel  
 BS                  "                  "       $MS_j$  - DOWNLINK          "



Full Duplex - 2 uplink, 2 downlink - to communicate data

↓  
not needed for voice

Each service provider has to get a spectrum Bandwidth



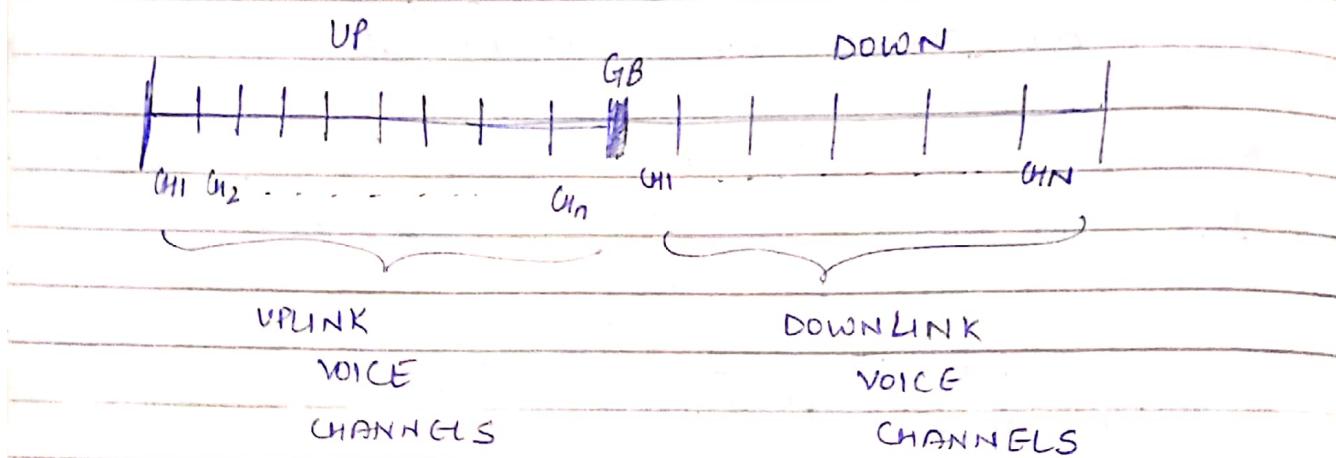
- divided into 2 parts - uplink and downlink
- high frequency may have power loss.
- BS can have power loss ∵ at higher freq. side, there will be downlink communication
- channel at that side

→ In this scheme, 2 can communicate  
This scheme is called FDMA / FDD



Freq. Division Duplexing

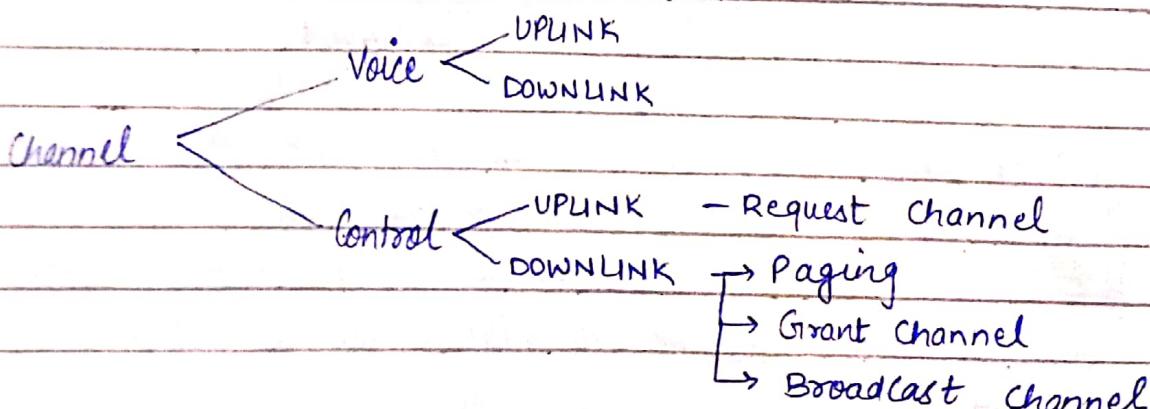
For more entities to communicate, multiple UPLINK and DOWNLINK channels to be created.



### Channel Allocation in FDMA / FDD

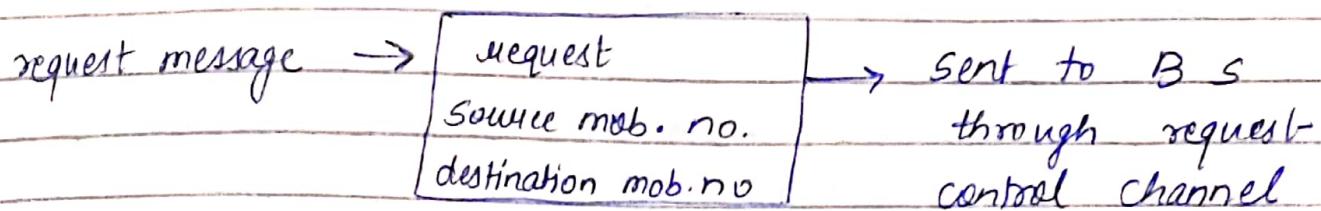
- static
  - You create an uplink/downlink channel
  - provide 1 each channel to n users
  - but some channels will remain idle when not communicating
  - no. of users cannot be increased

- Dynamic
  - request for obtaining a channel sent to BS

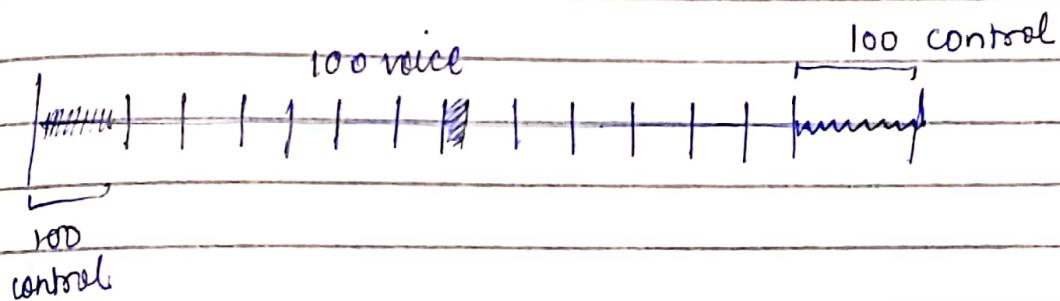


$$\text{uplink channels} = \frac{n}{\text{voice}} + \frac{k}{\text{control}} = \text{downlink channels}$$

- when person A calls, channel A - generates control



Initially, 100 voice channel, 10 control; further divide each control channel to make 10 control channels because freq. requirement is less as compared to voice :



1 - when X dials a number, X chooses channel sends request message to BS.

3 - B.S. checks if Y (destination) is busy, if busy it sends msg to X through PAGING or downlink channel

2 - Now BS checks if X (source) has money to place call, if not tells the source through PAGING that no money.

4 - If destination (Y) is free, BS allocates an UPLINK, downlink channel to the communication and displays

in Grant channel a message that X is allocated this uplink channel, Y is allocated that uplink channel.

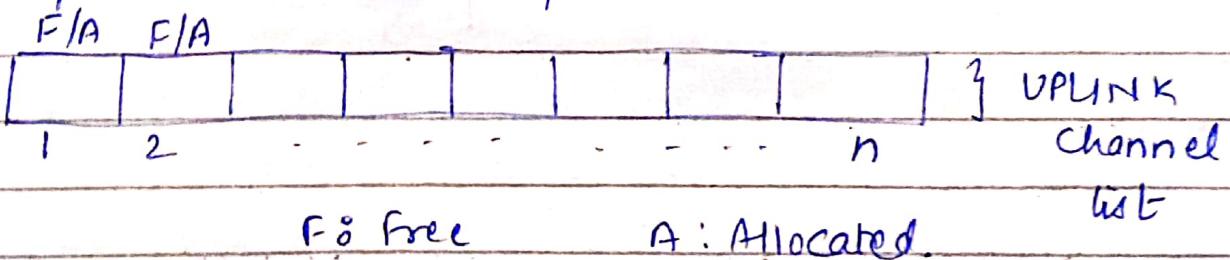
Now, through paging, BS tells source & destination about the communication.

5- If no channel (control) can be allocated then BS tells through PAGING channel - N/W busy

6- When communication is done, free control channels added to FREE list.

When channel allocated, it is added to ALLOCATED list of channels.

→ implementation example

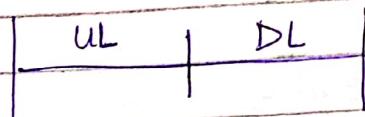


~~FDMA / FDD~~  $\xrightarrow{\text{duplexing}}$  (two sides)

↳ for mobile communication - 1 Generation

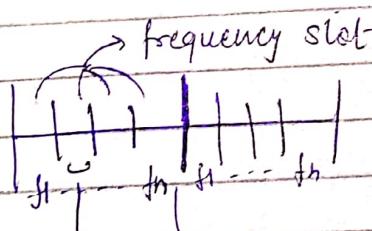


channels will be created in frequency slot.



higher frequency me radiations  
zyada hota hai

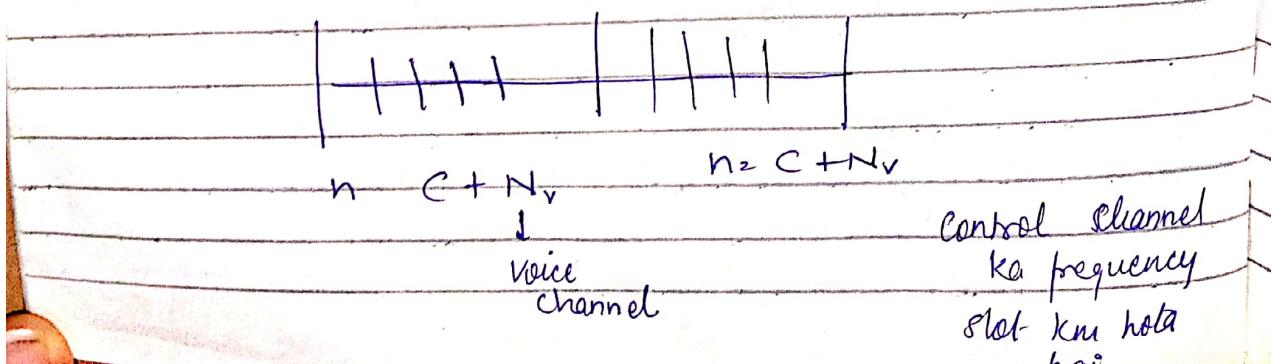
$$Bw_u = Bw_d = \frac{Bwt}{2}$$



within channel       $\Delta_{fg}$  (guard band)

bhi guard band hota hai.

Voice channel  $\rightarrow$  bigger frequency slot



C

K me divide  
kr dia

To total control channels  $C \times K$ ,  
for uplink and downlink.

voice channel  $\rightarrow$  slot-

control channel  $\rightarrow$  mini slot.

control frequency slots

Control Channel

Uplink  
Random Access Channel

downlink

1 broadcast CH  
check karega ki kisika signal zyada

lkon se mobile ko konse channel de rha hai

hai - jiska zyada uske saath associate karega  
(Hands off Mechanism)

kisi ek base station ke saath

(Data transmission me problem hota kar )