

## \* Transmission

↳ **Base band** :- is the transmission of a message as generated is transmitted and the band of the frequency occupied by the message.

↳ **Band pass** :- a wave form has a spectral magnitude that is non-zero for frequencies in some band concentrated about a frequency  $f = \pm f_c$

$f_c$  is much greater than zero.  $\rightarrow$  high frequencies.

\* Band pass is transmission band.

## Amplitude modulation (AM)

is a modulation technique where the Amplitude of a high-frequency sin wave is varied linearly with a time-varying message signal  $m(t)$ . the modulating signal  $m(t)$  contains the intended message or information.

⊙ \* Carrier is modulated by combining it with the modulating signal.

## pure carrier

higher and  
it's pure wave of constant frequency  $f_c$  it requires the modulation of the message on to a carrier signal to transmit it. and carrier signal ~~it~~ doesn't contain information.

\* the need of Hilbert transform at communication system.

- helps us relate the I and Q component.  $\rightarrow$  phase shift ( $\pm \frac{\pi}{2}$ ) degree.
- represent band pass signal as complex signal.  $\rightarrow$  so only effect the phase of the signal it has no effect on the Amplitude at all.

\* Modulation.  $\rightarrow$  definition  
↳ what is need

is the process of imparting the source information on to bandpass signal with a carrier frequency  $f_c$  by the introduction of Amplitude or phase perturbations or both.

why Modulation.

to transfer signals we need to transfer the frequency to higher level

→ what is need of filters in communication system.  
 are used to extract the base band signal in low pass filter.  
 (OR) to extract a specific frequency in bandpass filter.

→ meaning of narrow band signal.  
 is a type of approximations for FM if  $B \ll 1$  and  
 NBFM meaning  $f_c \gg$  band width  $= f_m$  BW  $= 2f_m$

→ the need of frequency domain transform, in communication system.

- simplify the mathematical analysis
- convolution of two signals is simply in multiplication form.
- dealing the signal in frequency domain is more easier.

the difference between SSB and USB

→



the difference between direct method and indirect method.

- in direct method a NBFM signal is first generated and then frequency multiplication is used to increase both carrier frequency & modulation index.

- in indirect method we use resonant circuit to make tuning at certain frequency apply, make

and then I get a linear relationship between  $m(t)$  &  $f_i(t)$  through a specific approximation. but this requires the use of a varactor diode

to change capacitance by changing the voltage ( $m(t)$ ). (this depends on the main diode response).

and therefore (any change in  $m(t)$  voltage causes change in a capacitance value. and hence change in  $f_i(t)$ ).

→ the diff between AM & FM →

→ Band width for each type of modulation

FM → NBFM →  $2f_m$   
 ↳ WBFM →  $\infty$

AM → DSB + C →  $2f_m$   
 ↳ DSB - SC →  $2f_m$   
 ↳ SSB →  $f_m$   
 ↳ USB →  $f_m +$



\* Amplitude modulation.

∴ The amplitude of a carrier signal is varied linearly with a time-varying  $m(t)$ .

$$C(t) = A_c \cdot \cos 2\pi f_c t$$

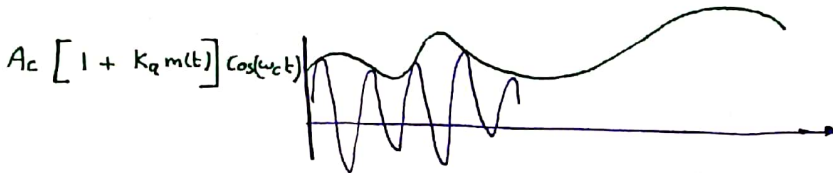
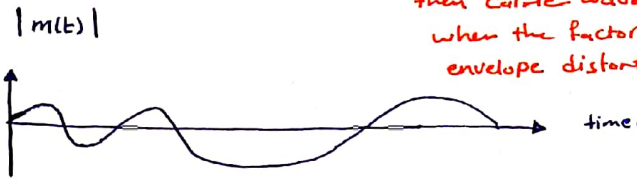
\*  $C(t)$  at AM only Amplitude  $A_c$  allowed to vary

$$S(t) = A_c (1 + k_a m(t)) \cdot \cos 2\pi f_c t$$

\*  $f_c, \theta$  are fixed.

$|k_a m(t)| < 1 \rightarrow$  if  $k_a$  is large to make  $|k_a m(t)| > 1$

then carrier wave becomes over modulated when the factor crosses zero this creates envelope distortion in the modulated signal.



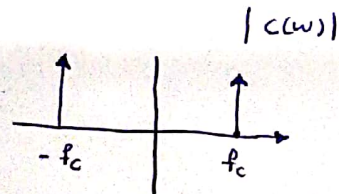
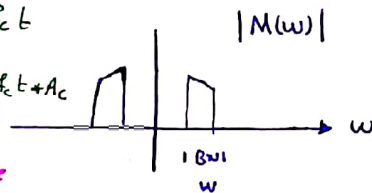
\* Double-sideband AM with carrier.

$$S(t) = A_c (1 + k_a m(t)) \cdot \cos 2\pi f_c t$$

$$S(t) = A_c \cos 2\pi f_c t + k_a m(t) \cos 2\pi f_c t + A_c$$

Carrier.

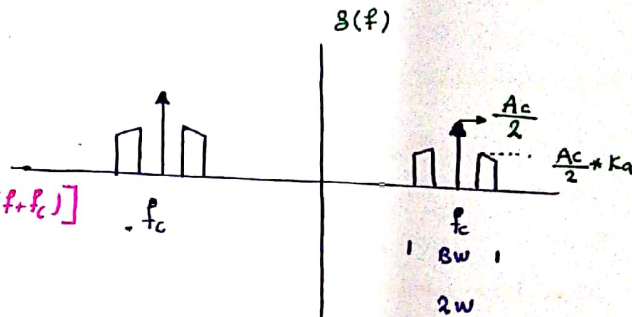
message



$$\frac{A_c}{2} [S(f-f_c) + S(f+f_c)]$$

message.

$$\frac{k_a}{2} A_c [M(f-f_c) + M(f+f_c)]$$



∴  $f_c \gg W$

we have two requirement to have base band signal shape over the envelope of the modulated signal.

$$S(t) = A_c (1 + k_a m(t)) \cdot \cos 2\pi f_c t$$

envelop

for modulated signal  $S(t)$

①  $\rightarrow |k_a m(t)| < 1$  for all  $t \rightarrow$  stays positive.

②  $\rightarrow$  Carrier frequency is much greater than highest frequency in base band signal.  
 $f_c \gg W$

\* Single tone.

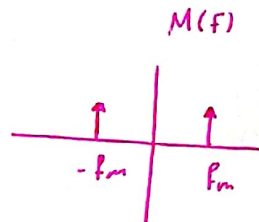
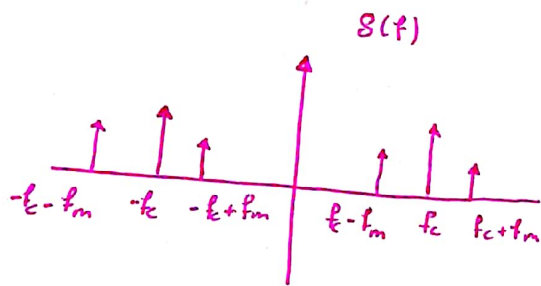
↔ any point in frequency domain it represented a single tone in continuous wave.

$$m(t) = A_m \cdot \cos 2\pi f_m t$$

$$s(t) = A_c [1 + K_a A_m \cos 2\pi f_m t] \cdot \cos 2\pi f_c t.$$

$$s(t) = A_c \cos 2\pi f_c t + \frac{A_c \mu}{2} [\cos 2\pi(f_c - f_m)t + \cos 2\pi(f_c + f_m)t]$$

$$s(f) = \left(\frac{A_c}{2}\right) [s(f-f_c) + s(f+f_c)] + \left(\frac{A_c \mu}{4}\right) [s(f-f_c+f_m) + s(f+f_c-f_m) + s(f-f_c-f_m) + s(f+f_c+f_m)].$$



\* the cause of uses single tone is :-

P\_m is complicated and not known precisely

So, we can find efficiency by using tone modulation.

$$\mu = \frac{A_m}{A_c}$$

$$P_c = \frac{A_c^2}{2} \rightarrow \text{power in carrier}$$

$$P_m = \frac{A_m^2}{2} = \frac{(\mu A_c)^2}{2} \rightarrow \text{power in tone modulation } (m(t))$$

$$P_s = \frac{1}{2} P_m = \frac{(\mu A_c)^2}{4} \rightarrow \text{power in sideband.}$$

$$\left[ \eta = \frac{\mu^2}{2 + \mu^2} \right] \Rightarrow \text{power efficiency.}$$

→ when  $\mu = 1$

$$\eta = 0.33$$

(efficiency is very low)

$$\left[ \eta_c = 0.66 \right] \Rightarrow \text{high}$$

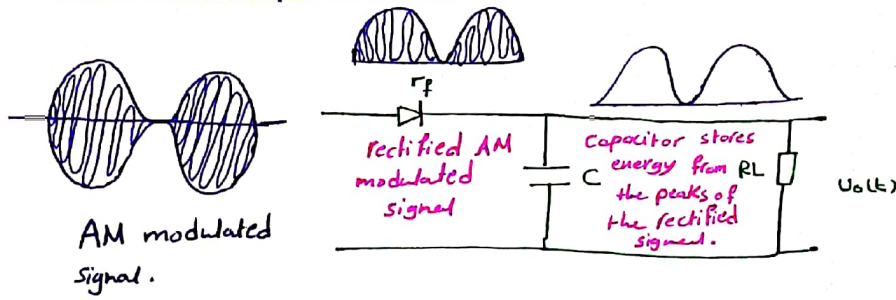
Maximum efficiency is that mean 66.66% power is spent on carrier.

In AM the power in the message is the power in sidebands.



# \* AM Demodulation \*

## AM Envelope Detector Circuit .



[ Key Ideas: Capacitor captures the voltage peaks of rectified wave form ]

← Two condition must be met for an envelope detector to work:-

(1) → Narrow band [ meaning  $f_c \gg$  bandwidth of  $m(t)$  ] .

(2) ⇒  $A_c + m(t) \geq 0$  , Modulation Index  $< 1$  .

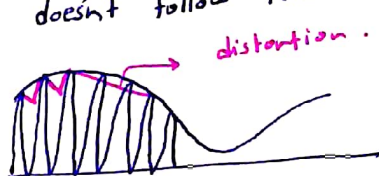
(3) → time constant  $\tau = RC$  too short.

(4) → Design criteria is  $2\pi B < \frac{1}{RC} \ll 2\pi f_c$

← on positive half-cycle of RF input signal (st) the diod is forward biased and the capacitor charges up rapidly to the peak value of RF input signal.

← when RF input falls below the output voltage the the diod becomes reverse-biased and capacitor dis charge slowly  
 $\tau = RC$  too short.

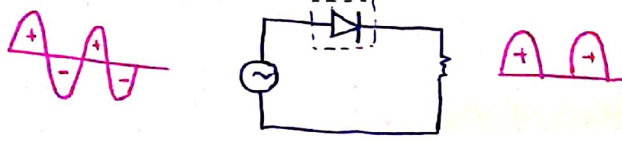
\* if  $RC$  is too large the voltage or the signal at the output doesn't follow the envelop  
 distortion. ☹️



\* Switching Amplitude Modulator.

\* half wave rectifier.

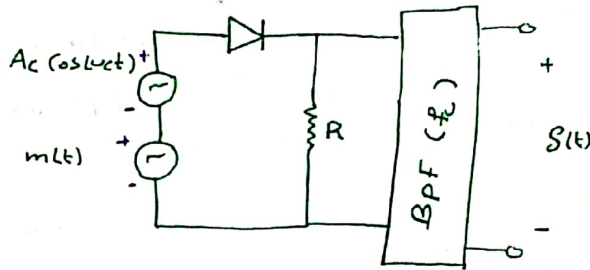
→ non-linear circuit component.



[nonlinearity generates new frequencies]

\* Switching Modulator.

- diode is ideal  $V_D = 0$ ,  $r_D = 0$  no forward bias.
- $c(t)$  is large amplitude.
- $m(t)$  is weak compared with  $c(t)$ .



سوال الفرق!

سواء تم تعديلك rectified او square wave.  
 مربع او حقل لسنه النتيجة  
 خاصة اني بقدر اعبر عن ال square wave.

السؤال! why لا بتا بعمل هيك!؟  
 { Fourier Series }

Because the cos wave is even function, symmetric around x-axis [has double-side band]

So Just we wanted transmit +ve or +ve half cycle

condition

∞ period for square wave = +ve cycle for carrier

← Analysis for circuit!

rectification for  $v_1(t)$

$$v_0(t) = \begin{cases} v_1(t) & , c(t) > 0 \\ 0 & , c(t) < 0 \end{cases}$$

Diode → short circuit.

∵  $m(t)$  negligible.

$$v_0(t) = [A_c \cdot \cos 2\pi f_c t + m(t)] \cdot g_{T_o}(t)$$

→ infinite summation of cos so, BW = ∞

Square wave  
 BW = ∞  
 ↳ non-causal.

$$g_{T_o}(t) = \frac{1}{2} + \frac{2}{\pi} \sum_{n=1}^{\infty} \frac{(-1)^{n-1}}{(2n-1)} \cdot \cos [2\pi f_c t (2n-1)]$$



$$g_{T_0}(t)$$

infinite summation

① فقط

هذه التي يعطين

$n=1$  وهو عند  $f = f_c$

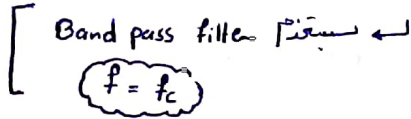
wanted component

$$* s(t) = A_c [1 + k_a m(t)] \cdot \cos 2\pi f_c t$$

$$[V_o(t) = \frac{A_c}{2} [1 + \frac{4}{\pi A_c} m(t)] \cdot \cos(2\pi f_c t)]$$

$\downarrow$   
 $k_a$

component wanted component



an unwanted component

delta's  
(Carrier)  
at even frequencies

version of the message signal  
at odd frequencies.

Infinite number of summation  
ربناك ربع كبره مني

infinite number of version of  
the message signal.

Switching circuit in Modulation

هي التي اجبرتها ان ترسل ال (Carrier)

من خلال وجود term لا Carrier في معادله  $V_o(t)$

AM وبتلك شوية انزل Full carrier وبتلك ال  
power side band

كثير من ال total power

$$s(t) = A_c(1 + k_a m(t)) \cdot \cos(2\pi f_c t)$$

$$\cdot \cos(2\pi f_c t)$$

السبب في AM modulation قديم واول نزع منه لان  
{ switching Modulator ← [ diod circuit ] سبب تقنيا }

\* at Full AM Limitations, AM wave has two main concern.

① Amplitude modulation has **waste full of power** → why?!  
the carrier wave  $c(t)$  is completely independent of the information-bearing signal  $m(t)$ . therefore, the transmission of the carrier wave is waste of power. (66.66% of the carrier)

② Amplitude modulation has **waste full of band width** → why?!  
the upper and lower sideband of an AM wave are uniquely related each other by virtue of symmetry about the carrier frequency.  
Hence, given magnitude and phase spectra of either sideband we can determine the other.

# Double - sideband suppressed carrier.

\* Conventional AM transmits both  $m(t)$  and  $c(t)$ , the its power efficiency is Low.

$$S(t) = A_c (1 + K_a m(t)) \cos(2\pi f_c t)$$

when this term canceled just remain.

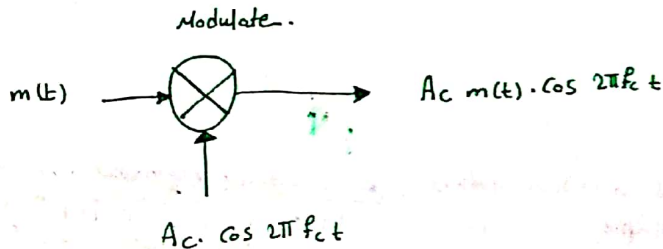
$$[S(t) = A_c m(t) \cdot \cos 2\pi f_c t] \Rightarrow \text{double - sideband suppressed carrier.}$$

then power efficiency becomes  $\eta = 100\%$  Because  $(A_c)^2$  approaches  $\infty$ .

DSB-SC, full AM with carrier.

التشابه  $2W = BW$

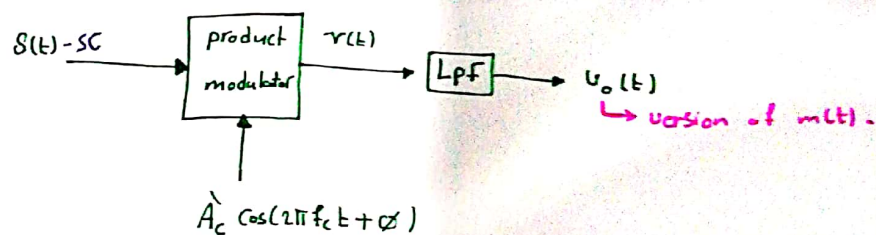
## Modulation



$$S(t) = A_c m(t) \cdot \cos 2\pi f_c t \quad \text{AM. DSB-SC}$$

$$S(f) = \frac{A_c}{2} [M(f - f_c) + M(f + f_c)]$$

## De Modulation



$$\begin{aligned} v(t) &= S(t) A_c \cos(2\pi f_c t + \phi) \\ &= A_c \cos(2\pi f_c t + \phi) \cdot (A_c m(t) \cdot \cos 2\pi f_c t) \\ &= \frac{A_c \cdot A_c}{2} m(t) \cos(4\pi f_c t + \phi) + \frac{A_c A_c}{2} m(t) \cos \phi \end{aligned}$$

LPF

$$v_o(t) = \frac{A_c A_c}{2} m(t) \cos(\phi)$$



AT D. Modulation in DSB-SC we needed **coherent detection (synchronous)**

coherent demodulation or detection is a method to recover the message signal from the received modulated signal that requires a carrier at the receiver.  
 ← this carrier signal must match in frequency and phase to the received signal.

$$v_o(t) = \frac{A_c A_c'}{2} \cos \phi \cdot m(t)$$

at receiver after demodulated

received signal.

modulated signal.

phase shift from carrier signal.

\* quadrature null effect  
 when  $\phi = \pm \frac{\pi}{2}$

قوتہ إذا  
 لانت قلت  
 اتاجب آرز  
 من طرف آ و سائل  
 بصرفی سے زیادہ

نہی ہے نقل سنا!

**Advantage of AM**

← AM not requiring coherent detection Method, Non-coherent Methods can be used which are much simpler to implement. like **Envelope Detector.**

desi words

vs  
 modulated signal without phase discrimination  
 ← **Costas receiver**  
 ← **De modulation DSB-SC**  
 ← **coherent detection.**

زیر = چستی  
 رپر = بدل  
 controlling  
 phase discrimination

\* De-modulation

Full AM → Envelope detector  
 DSB-SC → Costas receiver

ای عبارت سے مراد

## Single side band (SSB) AM

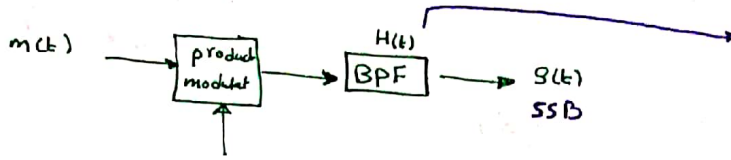
لـ تستخدم كل حثي إنا أنقال في BW  
لـرسل  $m(t)$

DSB-SC is spectrally inefficient because it uses twice the bandwidth of the message.

[SSB addresses that issue.]

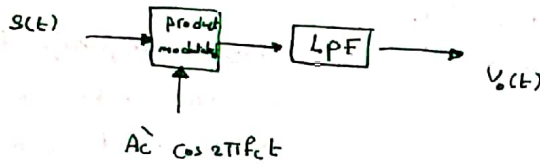
SSB Plan: (2)

Modulated:-



$$[ S(t) = \frac{1}{2} A_c m(t) \cos 2\pi f_c t + \frac{1}{2} A_c \hat{m}(t) \sin 2\pi f_c t ]$$

De-Modulated:-



$$[ V_o(t) = \frac{1}{4} A_c A_c [ M(t) \cdot \cos \phi + \hat{M}(t) \sin \phi ] ]$$

SSB is better

For the design of the band-pass filter to be practically feasible, there must be a certain separation between the two side bands that is wide enough to accommodate the transition band of the band-pass filter.

بـسـتـخدمـهـ المـوجـهـ بارـهـ بـيـكونـهـ سـبـايـ السـرـيـهـ رـهـ بـتـاحـيـهـ مـشـالـهـ

upper  
β lower.

SSB ومـعـنـاهـ فـقـهـ الشـرـطـ الـذيـ يـجـبـ عـمـدـهـ gap بـيـن

Dis adv SSB

1) Designing and implementing the sharp Low/ high (Ideal filter)

Band pass filter is not easy for circuit designer.

2) SSB relies on being able to filter out one side band, this make it not possible for all signal, that have strong component at low frequency.

- SSB valid for audio signal
- SSB not valid for video signal

طـيـبـ هـيـهـ مـشـالـهـ حـالـهـ بـهـ الـكـلـ ؟  
سـكـانـهـ المـشـلـهـ

the solution is Vestigial sideband Modulation (VSB)  
a small portion (trace) of the unnecessary sideband is used

هذا هو الحل

this reduces DC distortion

adv [SSB] & R Comparison between [SSB & VSB]

- Allow a small amount (trace) or vestige, of the unwanted sideband to appear at the output of an SSB.
- The design of the sideband filter is simplified since the need for sharp cutoff at the carrier frequency is eliminated.
- VSB improve low-frequency response.



Helmet في الـ L في Modulation في الـ \*  
transform . filter في الـ

[phase shift Method to Generate SSB AM]

A. selective Filtering Method  
B. phase shift Method.

\* De-modulation of SSB signals  
A. coherent demodulation

B. Envelope detection with a carrier (SSB+C)

(M)

\* distortion in SSB

demodulated  $\rightarrow$  distortion في الـ

[phase distortion] في الـ

valid for audio signal, not for video signal (shift in time)

video signal frequency  $\rightarrow$  80 Hz to 6 KHz

audio signal frequency  $\rightarrow$  20 Hz to 20, KHz

higher frequency than video  
so any shift in time in  
audio signal will be very  
effective.

Adv SSB

- transmit power efficiency
- Bandwidth efficiency
- improved SNR
- Reduction in distortion

Dis adv SSB

- difficulty in alignment and tuning
- complex

Adv VSB

- use simple filter design
- less bandwidth as compared to that DSB SC
- Also efficient as compared to that SSB
- possibility of transmission of low frequency components of modulating signals. like video signals

[TV signal]

\* distortion in VSB happens in De modulation. inside the envelope  
Because  $m(t)$  is non-linear.

\* the distortion can be reduced by

1) - reducing the percentage modulation to reduce  $k_a$ .

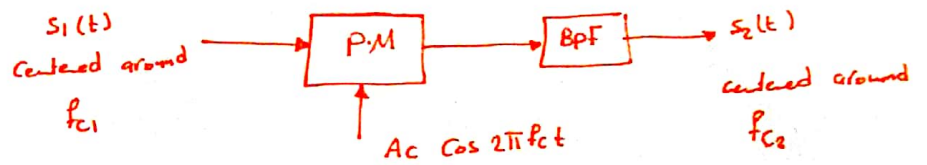
2) - Increasing the width of vestigial side band to reduce  $m'(t)$ .

Frequency Translation OR Frequency Division Multiplexing: FDM

Multiplexing is a technique when a number of independent signals are combined and transmitted in a common channel.

\* these signals are de-multiplexed at the receiver.

Mod :- Shift message frequency range to mutually exclusive high frequency bands



→ . کم از کم دو فرکانس FDM

← useful bandwidth of medium exceeds required bandwidth of channel.

☺ [ Each signal modulated to a different carrier frequency ]  
 ☺ [ Carrier frequencies separated so signals don't overlap ( don't happen trace at near message signal ) .

← سبب عدم وجود trace at near message signal.  
 ← علی الاطلاق الفرم بین سینی  
 ← BW سبب  
 ← إذا كان  $m(t)$  متادی  
 ← نتیج .