

Angle Modulation

→ PM ——— Phase modulation

→ FM ——— frequency modulation

$f(t) \rightarrow \begin{cases} \cos \omega_m t \\ \sin \omega_m t \end{cases}$ } modulating sig

$$V_c(t) = \text{carrier} = A \cos \omega_c t$$

$$\omega_c \gg \omega_m$$

$$= 2\pi f_c$$

→ $f_c : 88 \text{ MHz} - 108 \text{ MHz}$ ——— FM, PM
: $535 \text{ kHz} - 1605 \text{ kHz}$ ——— AM

Phase Modulation : In phase modulation, the phase of carrier is varied in accordance with the instantaneous value of the amplitude of the modulating sig. keeping amplitude & the frequency of the carrier fixed.

$$V_c(t) = A \cos \omega_c t = A \cos \theta(t)$$

$$\theta(t) = \omega_c t$$

$$V_m(t) = f(t)$$

$$V_{pm}(t) = A \cos \theta_i(t)$$

$$\text{where, } \theta_i = \theta + k_p f(t) \\ = \omega_c t + k_p \cdot f(t)$$

$$\boxed{V_{pm}(t) = A \cos (\omega_c t + k_p \cdot f(t))}$$

General expression
of the P.M. signal

$$\text{if } f(t) \rightarrow \text{modulating sig} = V_m \cos \omega_m t$$

$$V_{PM}(t) = A \cos[\omega_c t + m_p \cos \omega_m t]$$

where
 $m_p = k_p \cdot V_m$

modulation ind.

FM Sign.

In frequency modulation the frequency of carrier is varied in accordance with the instantaneous value of the amplitude of modulating sig. keep amplitude & the phase of the carrier fixed.

$$V_c(t) = A \cos \omega_c t = A \cos \theta(t)$$

--- carrier

$$\theta = \omega_c t$$

$f(t)$ - modulating sig

$$\omega_i = \omega_c + k_f \cdot f(t)$$

FM signal

$$V_{FM}(t) = A \cos \theta_i(t)$$

$$\theta_i(t) = \int \omega_i dt$$

$$\theta_i = \int (\omega_c + k_f f(t)) dt = \omega_c t + k_f \int_0^t f(t) dt$$

$$V_{FM}(t) = A \cos \left[\omega_c t + k_f \int_0^t f(t) dt \right]$$

general
 expressi
 of FM sig

if $f(t) = V_m \cos \omega_m t$

$$\int_0^t f(t) dt = \frac{V_m}{\omega_m} \sin \omega_m t$$

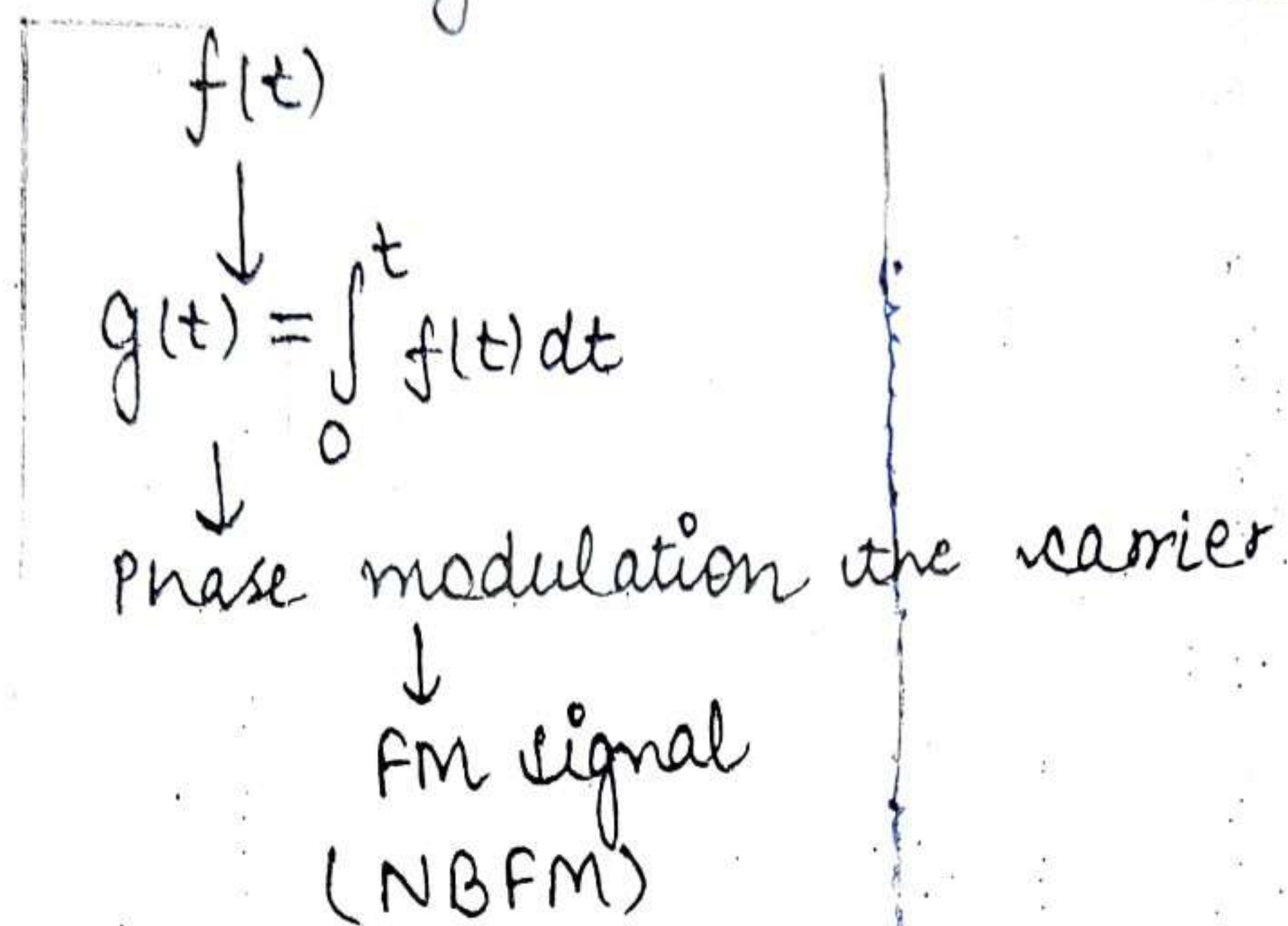
$$V_{FM}(t) = A \cos[\omega_c t + m_f \sin \omega_m t]$$

$$m_f = \frac{k_f V_m}{\omega_m} \quad \text{--- mod. ind.}$$

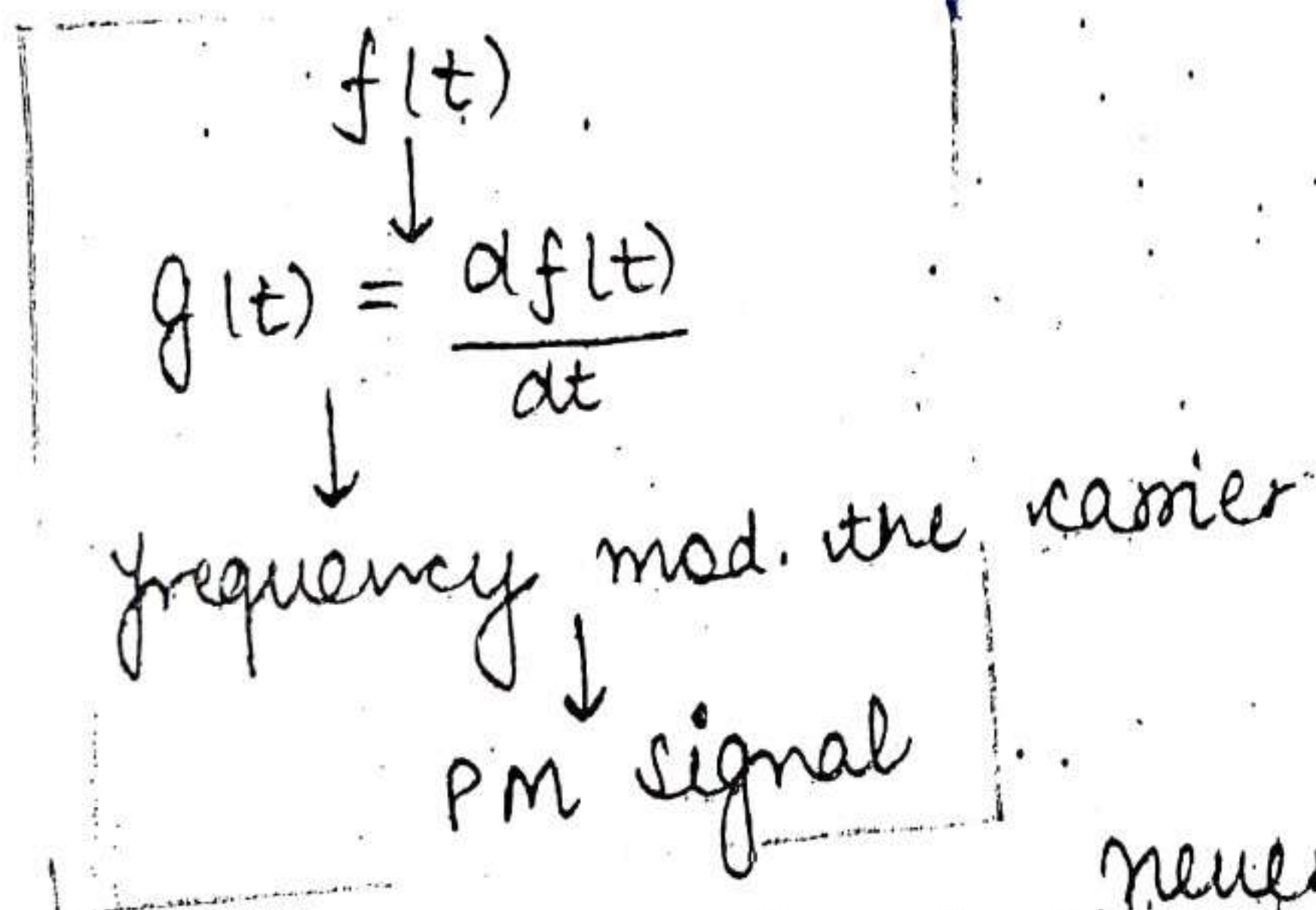
→ The frequency and phase modulation are inter dependent since any variation in the frequency will cause variation in the phase and vice-versa. Therefore we can't have a pure PM signal or pure FM signal.

→ If we integrate the modulating sig and allow the carrier for phase modulation the result is FM signal. This method is always

preferred. Since it reduces the noise in the system.
 → If we differentiate the modulating sig and allow the carrier for frequency modulation, the result is PM signal. this method is never used since the noise in the system increases due to differentiation.



Indirect method
 or Armstrong's method of FM



Power contained in FM signal → never preferred

$$u_{FM}(t) = A \cos \left[\omega_c t + k_f \int_0^t f(t) dt \right]$$

$$= A \cos [\omega_c t + m_f \sin \omega_m t]$$

$$P = \frac{u_{FM}^2(t)}{2} = \frac{A^2 \cos^2 [\omega_c t + k_f \int_0^t f(t) dt]}{2}$$

$$P = \frac{1}{2} A^2$$

FM

$$P_t = P_c \left(1 + \frac{m_a^2}{2} \right) \text{ --- AM}$$

Ideally, head...

→ For AM signal the Total Power required upon the carrier amplitude along with modulation index. For FM or PM the total required for transmission depends purely upon carrier voltage and is independent of modulation index.

frequency component in FM signal →

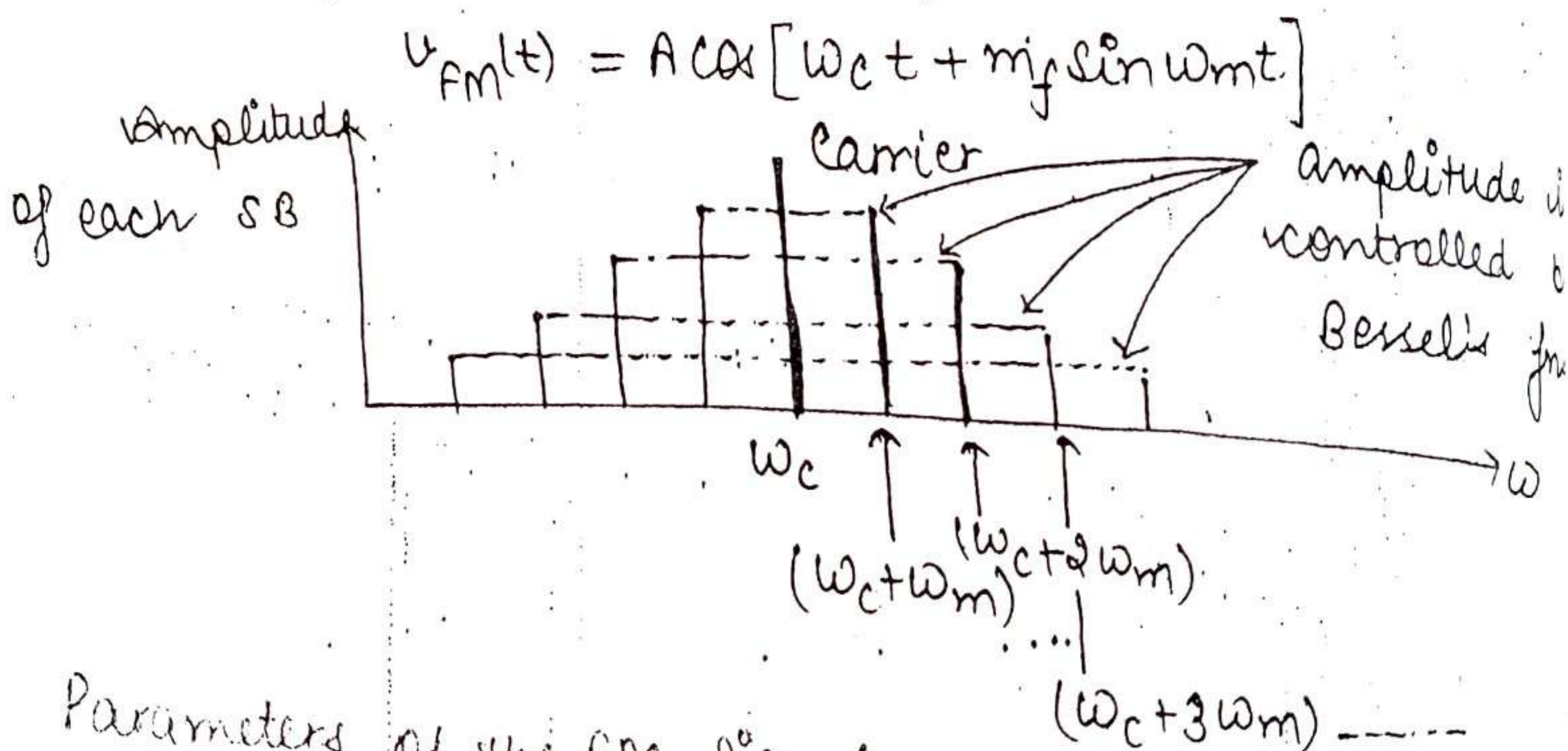
$$V_{FM}(t) = A \cos[\omega_c t + m_f \sin \omega_m t]$$

--- infinite no of terms & the mag. depends upon the Bessel's function

⇒ infinite no of side bands

⇒ infinite Band width for the FM signal

frequency spectrum of FM signal →



Parameters of the FM signal →

1. ω_m ; f_m — modulating frequency
2. m_f — modulation index
3. δ — frequency deviation

$$\delta = m_f \cdot \omega_m \text{ — rad/sec.}$$

$$= m_f \cdot f_m \text{ — Hz}$$

bandwidth →

$$B.W \approx 2(\delta + \omega_m) \quad \text{Carson's rule}$$
$$\approx 2(\delta + f_m) \quad \text{rad/sec}$$

Hz

Case-1. Wide Band FM Signal (WBFM) →

$$m_f \gg 1$$

$$BW \approx 2(\delta + \omega_m)$$

$$\approx 2(m_f \omega_m + \omega_m)$$

$$\approx 2\omega_m(m_f + 1) \quad \{ \text{bcz } m_f \gg 1 \}$$

$$B.W. \approx 2\omega_m m_f$$

$$\approx 2\delta$$

Rad/sec

Used for FM broadcast.

$$\delta_{\max} \approx \pm 75 \text{ KHz}$$

Case-2. Narrow band FM Signal (NBFM) →

$$m_f \ll 1$$

$$BW \approx 2\omega_m(m_f + 1)$$

$$B.W. \approx 2\omega_m \quad \text{Rad/sec.}$$

--- Same as that of AM-DSB system.

--- contains 2 side bands.

--- used as FM radio in mobiles.

Any FM signal contains infinite no of side bands.

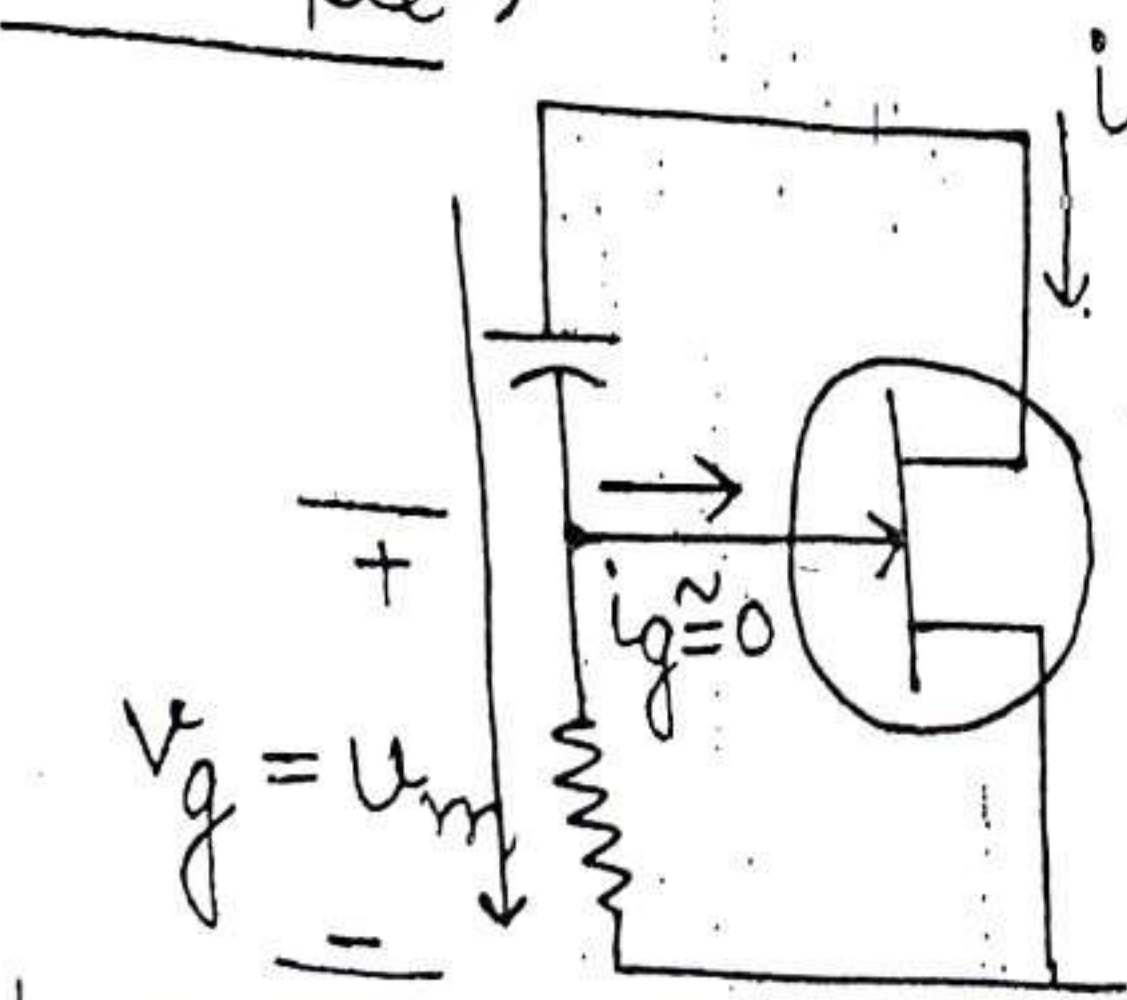
Therefore ideally the B.W. required for the transmission of FM signal is also ∞ , which practically is not available. We then transmit first few pair of sidebands which contribute significant amount of power and we neglect those higher side band which contribute only 2-5% of the total power. Therefore practically the B.W. required is now finite without compromising quality of the sig.

Practically the B.W. required is given by the Carson's rule.

- iii) In wide band FM, No. of side bands are large in no. where as for narrow band we transmit only one pair of side bands.
- iv) ω_m controls the separation b/w two side bands.
- v) the modulation index m_f controls total significant side bands which are being transmitted.
- vi) the frequency deviation δ controls the total width required for the transmission of FM signal.
- vii) The FM system is a high fi-fi (high fidelity) system and therefore such system has the ability to reproduce the sig having same quality which the sig was initially transmitted.
- viii) Generation of WBFM signal $\rightarrow$

Reactance modulator

Principle $\rightarrow$



gate voltage
or
modulating signal

$$g_m = \frac{i}{U_g} \quad \text{--- (i)}$$

$$U_g = \frac{R}{R + \frac{1}{j\omega C}} U$$

$$U_g = \frac{RU}{R - jX_C}$$

$$Z_{out} = \frac{U}{i} = ?$$

$$Z_{out} = \frac{U}{i} \quad \text{--- (ii)}$$

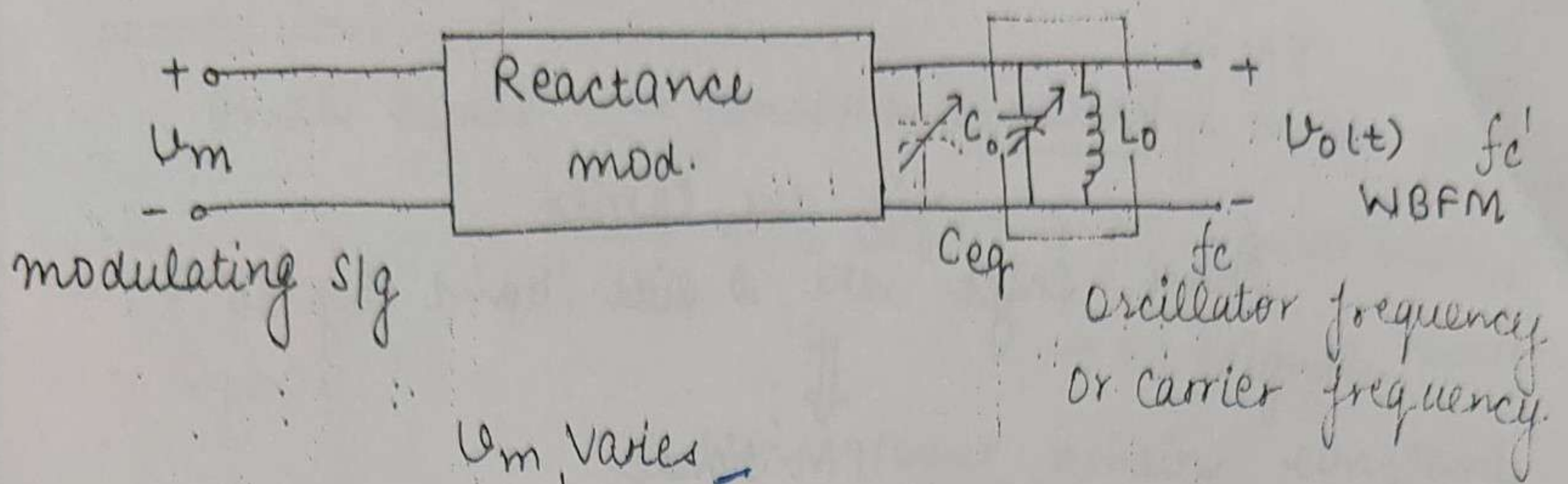
$$g_m = \frac{i(R - jX_C)}{RU}$$

$$\frac{U}{i} = \frac{(R - jX_C)}{R g_m} = Z_{out} \quad \text{--- (iii)}$$

$$Z_{out} \approx \frac{-jX_C}{g_m R} = \frac{1}{+j g_m \omega C R} \approx \frac{1}{j\omega C_{eq}}$$

$$C_{eq} = g_m R C$$

Purely Reactive



V_m varies

$$g_m = \frac{i}{V_m} \text{ varies}$$

$$C_{eq} = g_m R_C \text{ varies}$$

$$C = C_0 + C_{eq}$$

$$= C_0 + g_m R_C \text{ varies}$$

$$f_c' = \frac{1}{2\pi \sqrt{L_0(C_0 + C_{eq})}} \text{ varies} \Rightarrow \text{WB FM signal}$$

1) Using this ckt the frequency of the output signal varies in accordance with instantaneous value of the amplitude of modulating signal. therefore the F.M. signal is generated. Since there may be large variation for the amplitude of the i/p signal, the frequency deviation at the output may also be large and therefore this results in wide band FM signal.

Generation of narrow band FM signal →

- indirect method
- Armstrong's method

Principle →

$V_m(t)$ --- modulating signal

↓ integrate it $\int_0^t V_m(t) dt = g(t)$

↓ Phase modulate the carrier using $g(t)$

↓ NB FM signal

PM →

Option-1,

fix the carrier
Phase Shift the 2 side band by 90°

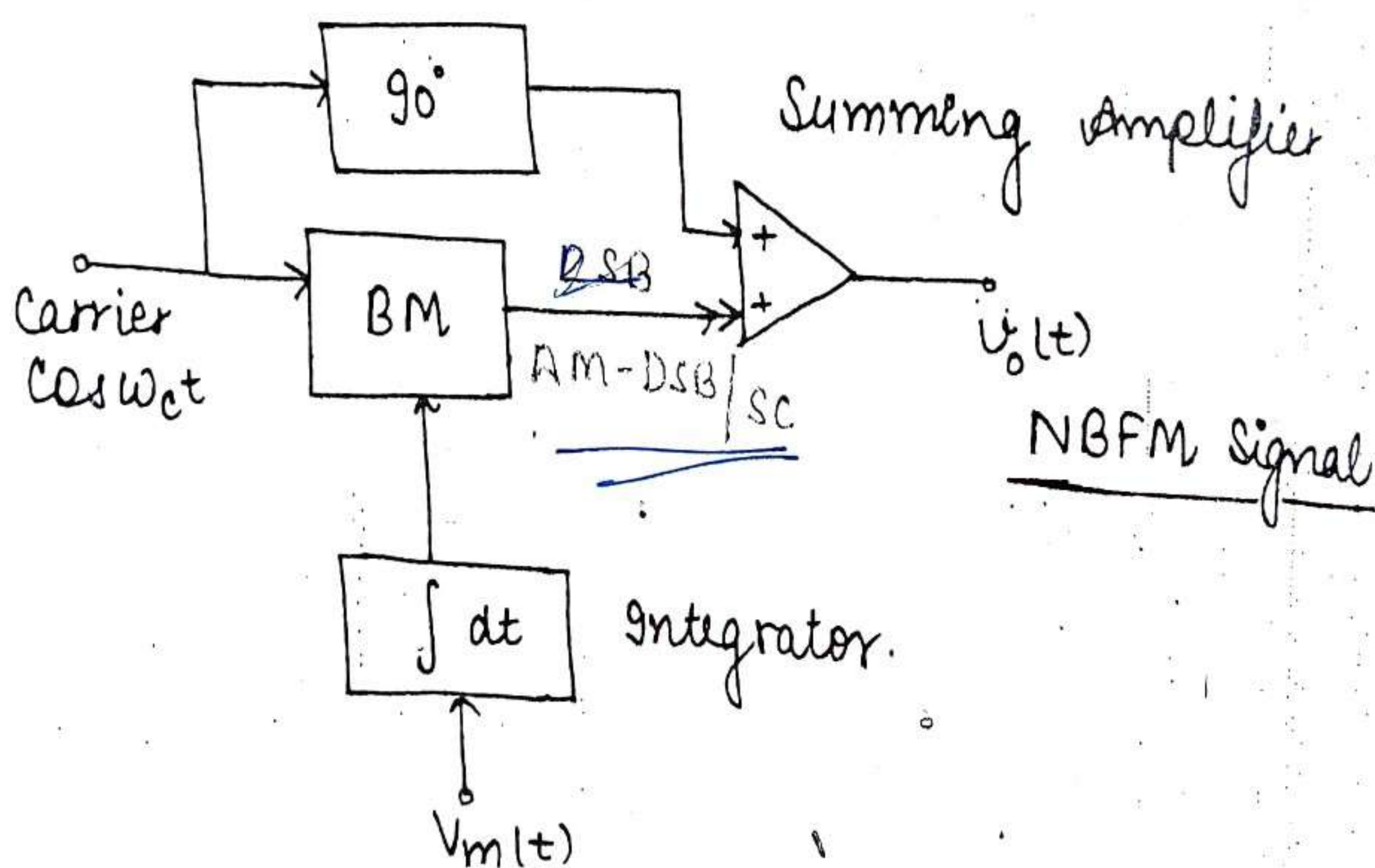
↓
PM signal.

Option-2,

fix the 2 side band
Phase Shift the carrier by 90° .

↓
PM signal

* 2nd option is better, bcz both side band have same frequency and much typical to provide phase of 90° both.



* In the NBFM if we add up frequency modulator, we get WBFM....

→ The generation of NBFM signal has more complex circuitary as compared to generation of wide band FM signal....

→ The wide band FM signal can then also be generated from the narrow band FM signal by passing the sig through series of frequency multipliers.

Pre-emphasis & de-emphasis →

Audio signal → fundamental + large no. of Harmonics

→ Side Band ↑ → f ↑ → amplitude ↓

→ S ↓ (Signal Power)

→ Noise Power remains constant

→ f ↑ → $\left(\frac{S}{N}\right)$ ↓

(offset +90°)
(offset -90°)

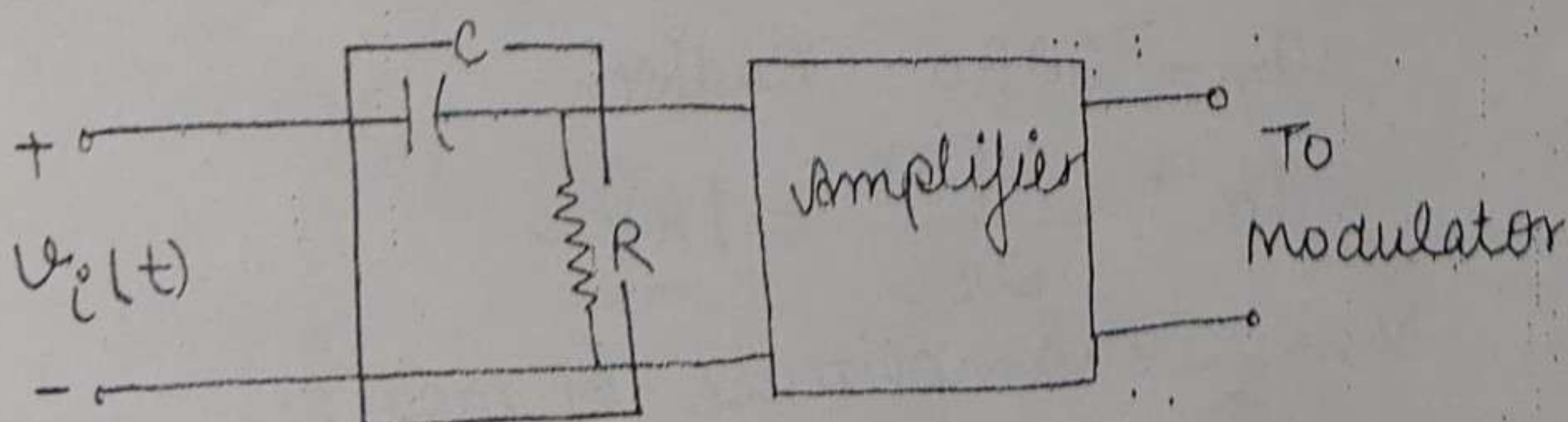
Hence corresponding to high frequency components the signal to noise ratio has low value.

Therefore these high frequency components are boosted or emphasized before the transmission of sig takes place. This is done by using a preemphasis ckt and is used at the transmitting end before the modulation takes place.

Since the relative sig to noise ratio for various frequency component has been disturbed then those high frequency components which were initially boosted are now broad down to the same level so that relative sig to noise ratio for various frequency component is maintained to have same quality of sig.

This is done by using a deemphasis ckt and is used at the receiving end just after demodulation takes place.

Pre-emphasis ckt →

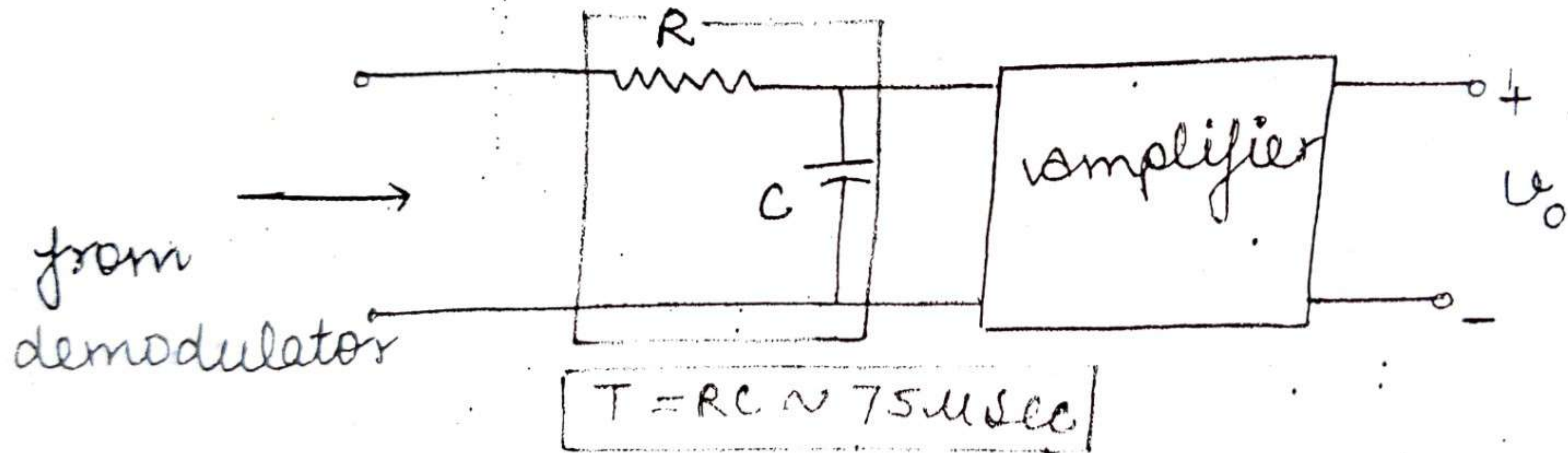


$$T = RC \cong 75 \mu\text{Sec.}$$

Used at the transmitting end before modulator

De-emphasis ckt →

used at the Receiving end just after
Demodulator



* by arranging the value of $R + C$ at Receiver we
get the volume as we like by suppress
the low frequency components.