

ELEC366

Fundamentals of Telecom. Systems

Summer 2015

Lecture 1

Course outline :

- 1) Review of Signals & Systems
- 2) Amplitude Modulation
- 3) Angle Modulation
- 4) Noise in AM & FM Systems.
- 5) Sampling and Quantization
- 6) Digital Modulation
- 7) Digital Services and Technologies.

Signals & Systems

Signals are time-varying quantities, i.e., they are functions of time, e.g., speech, video, data.

Signals may be ^{time-}discrete or time-continuous digital or analog.

Energy signals:

if the energy of a signal is finite, i.e., if

$$E_g = \int_{-\infty}^{\infty} |g(t)|^2 dt < \infty$$

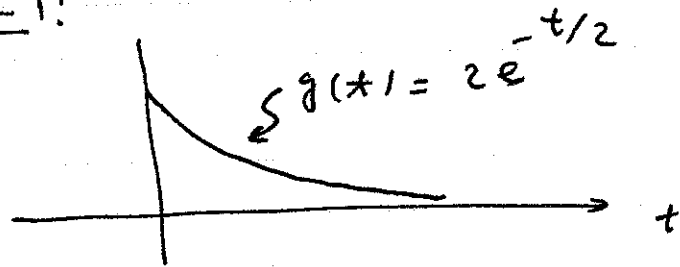
then the signal is called an energy signal.

Power signals: if the Power of a signal:

$$0 < P_g = \lim_{T \rightarrow \infty} \frac{1}{T} \int_{-T/2}^{T/2} |g(t)|^2 dt < \infty$$

then the signal is called a power signal.

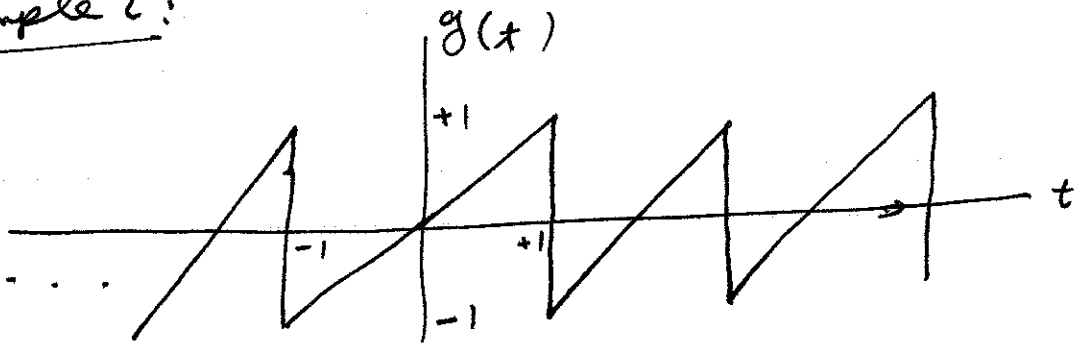
example 1:



$$E_g = \int_0^{\infty} 4e^{-t} dt = 4 \int_0^{\infty} e^{-t} = 4$$

~~Power~~ $P_g = 0$

Example 2:



$E_g = \infty \Rightarrow$ not an energy signal

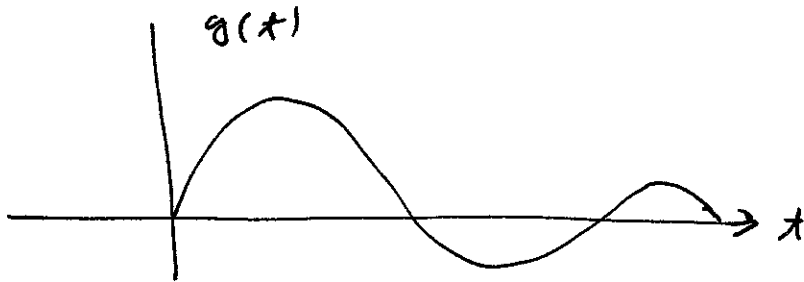
$$P_g = \lim_{T \rightarrow \infty} \frac{1}{T} \int_{-T/2}^{T/2} g^2(t) dt = \frac{1}{2} \int_{-1}^1 g^2(t) dt = \frac{1}{2} \int_{-1}^1 t^2 dt$$

$$P_g = \frac{1}{2} \times \frac{t^3}{3} \Big|_{-1}^1 = \frac{1}{6} (1+1) = \frac{1}{3}$$

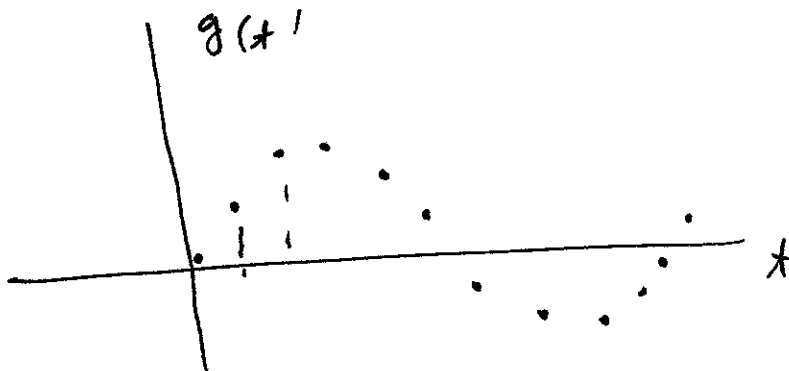
A signal can be Power, energy, non
but not both at the same time.

Continuous-time ↔ Discrete-time

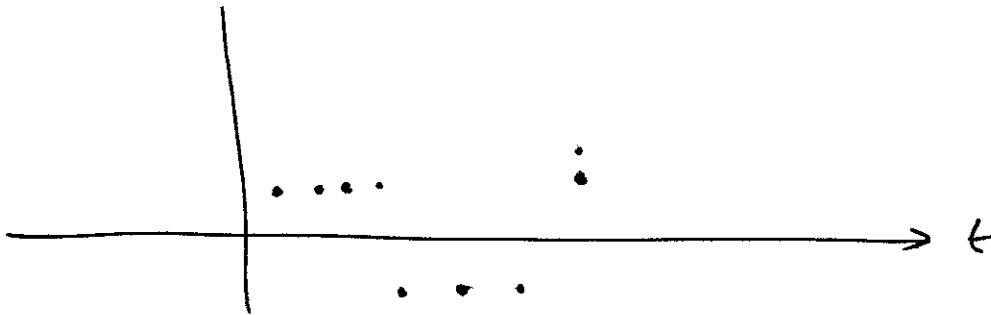
Analog ↔ Digital



Continuous time, Analog (Continuous Value)



discrete-time, Analog.



Digital (Discrete-time, discrete Amplitude)

Periodic \leftrightarrow aperiodic

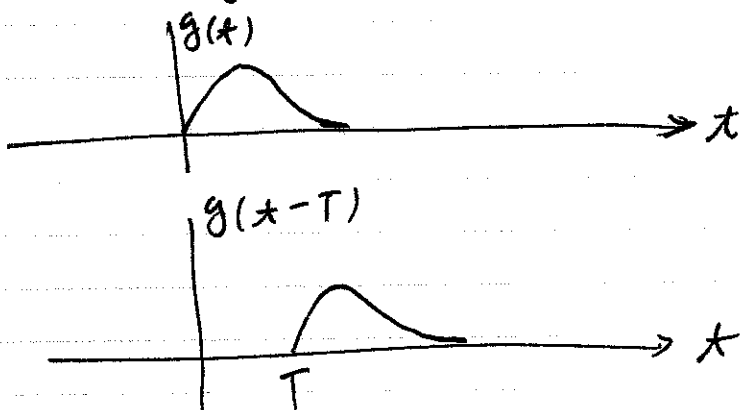
A signal is period with with period T if

$g(x) = g(x + T)$ all x .

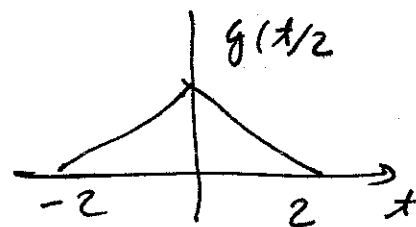
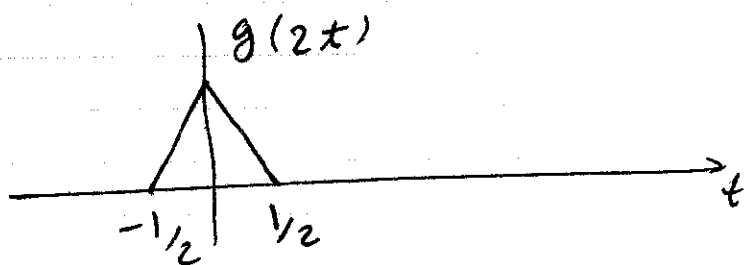
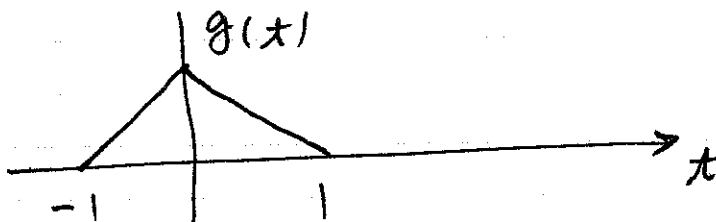
Signal operations

Time - shift

$f(x) = g(x - T)$

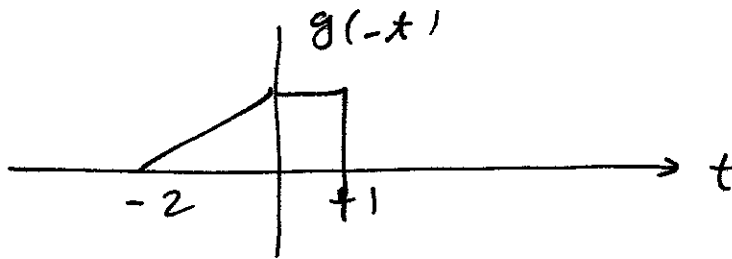
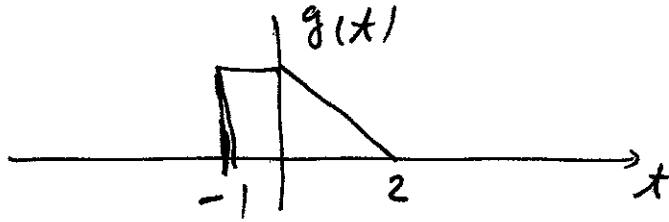


Time - Scaling $f(x) = g(ax)$



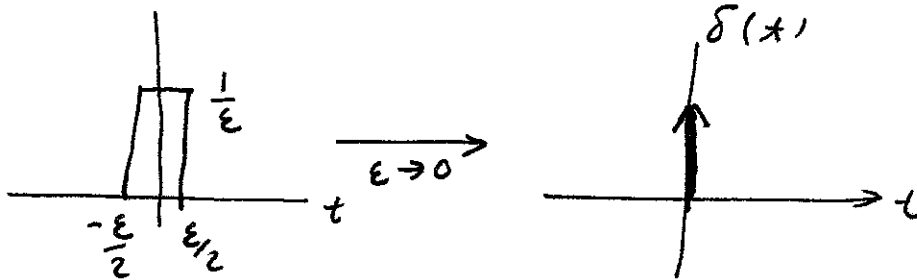
Time-reversal

$$f(x) = g(-x)$$



Generalized Functions

Unit impulse function



$$\delta(x) = 0 \quad x \neq 0$$

$$\int_{-\infty}^{\infty} \delta(x) dt = 1$$

$$f(x) \cdot \delta(x) = f(0) \delta(x)$$

So:

$$\int_{-\infty}^{\infty} f(x) \delta(x) dt = f(0) \int_{-\infty}^{\infty} \delta(x) dt = f(0)$$

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Similarly

$$\int_{-\infty}^{\infty} f(x) \delta(x-T) dx = f(T)$$

this is called the sifting property

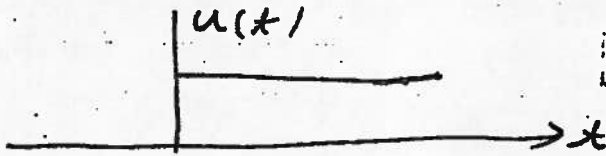
Unit step function

$$u(x) = \begin{cases} 1 & x \geq 0 \\ 0 & x < 0 \end{cases}$$

$$\int_{-\infty}^t \delta(\tau) d\tau = \begin{cases} 0 & t < 0 \\ 1 & t \geq 0 \end{cases} = u(t)$$

or

$$\delta(x) = \frac{d}{dx} u(x)$$



Fourier Series

Trigonometric Fourier Series:

$$g(t) = a_0 + a_1 \cos \omega_0 t + a_2 \cos 2\omega_0 t + \dots \\ + b_1 \sin \omega_0 t + b_2 \sin 2\omega_0 t + \dots$$

or

$$g(t) = a_0 + \sum_{n=1}^{\infty} [a_n \cos n\omega_0 t + b_n \sin n\omega_0 t]$$

where

$$\omega_0 = \frac{2\pi}{T_0}$$

T_0 is the period of $g(t)$.

$$a_0 = \frac{1}{T_0} \int_{t_1}^{t_1 + T_0} g(t) dt$$

$$a_n = \frac{2}{T_0} \int_{t_1}^{t_1 + T_0} g(t) \cos n\omega_0 t dt \quad n = 1, 2, 3, \dots$$

$$b_n = \frac{2}{T_0} \int_{t_1}^{t_1 + T_0} g(t) \sin n\omega_0 t dt \quad n = 1, 2, 3, \dots$$

Compact Trigonometric F. Series

$$a_n \cos n\omega_0 t + b_n \sin n\omega_0 t = C_n \cos(n\omega_0 t + \theta_n)$$

where

$$C_n = \sqrt{a_n^2 + b_n^2}$$

$$\theta_n = \tan^{-1}\left(-\frac{b_n}{a_n}\right)$$

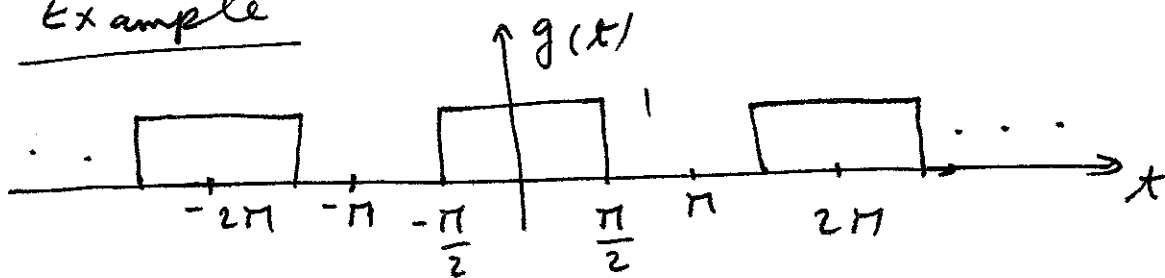
So:

$$g(t) = C_0 + \sum_{n=1}^{\infty} C_n \cos(n\omega_0 t + \theta_n)$$

where

$$C_0 = a_0$$

Example



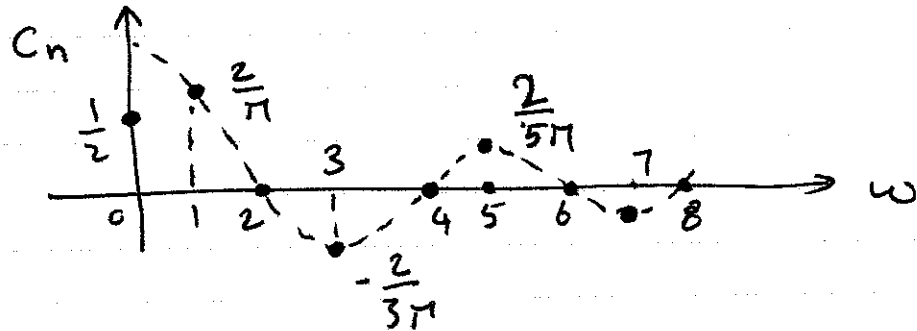
$$T_0 = 2\pi \Rightarrow \omega_0 = \frac{2\pi}{T_0} = 1$$

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

$$a_n = \frac{2}{2\pi} \int_{-\pi/2}^{\pi/2} \cos(n\omega_0 t) dt = \frac{2}{n\pi} \sin\left(\frac{n\pi}{2}\right)$$

$$b_n = \frac{2}{2\pi} \int_{-\pi/2}^{\pi/2} \sin(n\omega_0 t) dt = 0$$

$$g(x) = \frac{1}{2} + \frac{2}{\pi} \left(\cos x - \frac{1}{3} \cos 3x + \frac{1}{5} \cos 5x - \dots \right)$$



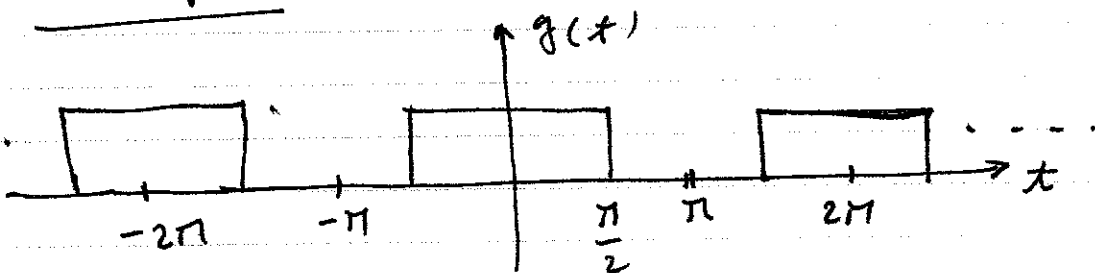
Exponential Fourier Series

$$g(x) = \sum_{n=-\infty}^{\infty} D_n e^{jn\omega_0 t}$$

where

$$D_n = \frac{1}{T_0} \int_{T_0} g(x) e^{-jn\omega_0 t} dt$$

Example:

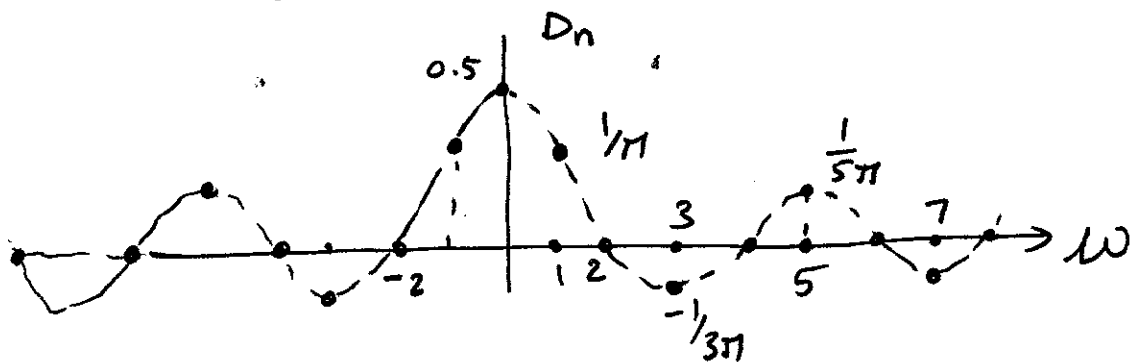


$$g(x) = \sum_{n=-\infty}^{\infty} D_n e^{jn\omega_0 t}$$

$$D_n = \frac{1}{T_0} \int_{-T_0/2}^{T_0/2} g(x) e^{-jn\omega_0 t} dt = \frac{1}{2\pi} \int_{-\pi/2}^{\pi/2} e^{-jn\omega_0 t} dt$$

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$$D_n = \frac{2}{n\omega_0 T_0} \sin\left(\frac{n\omega_0 T_0}{4}\right) = \frac{1}{n\pi} \sin\left(\frac{n\pi}{2}\right)$$



Parseval's Theorem

$$g(x) = C_0 + \sum_{n=1}^{\infty} C_n \cos(n\omega_0 t + \theta_n)$$

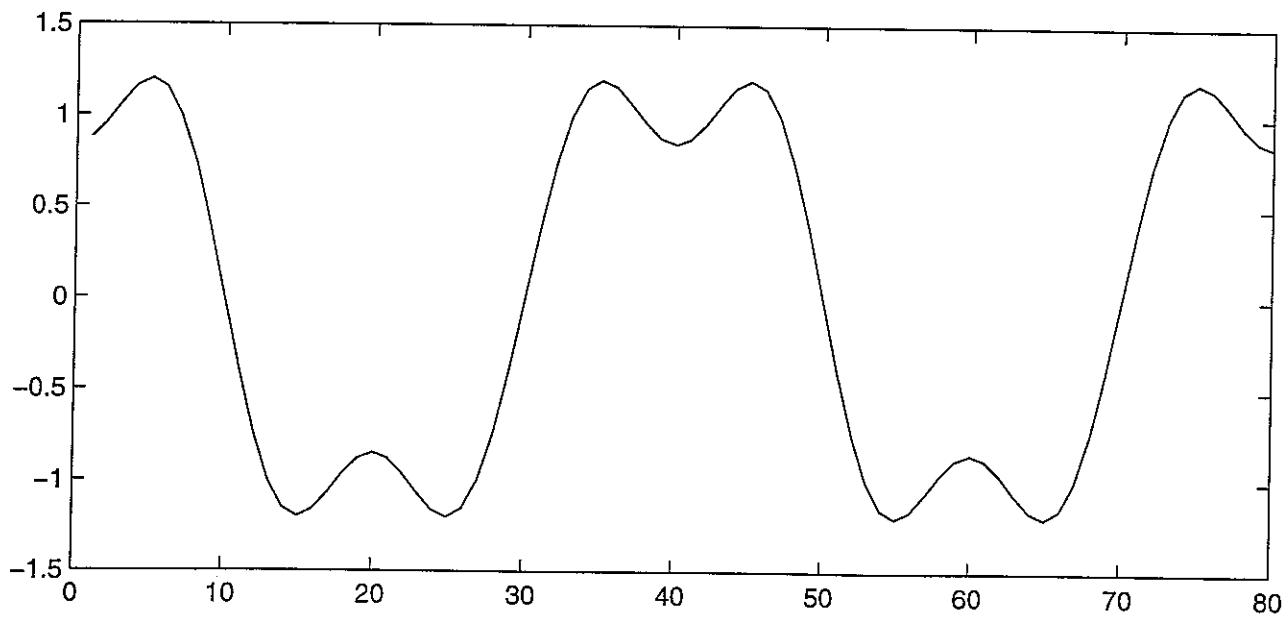
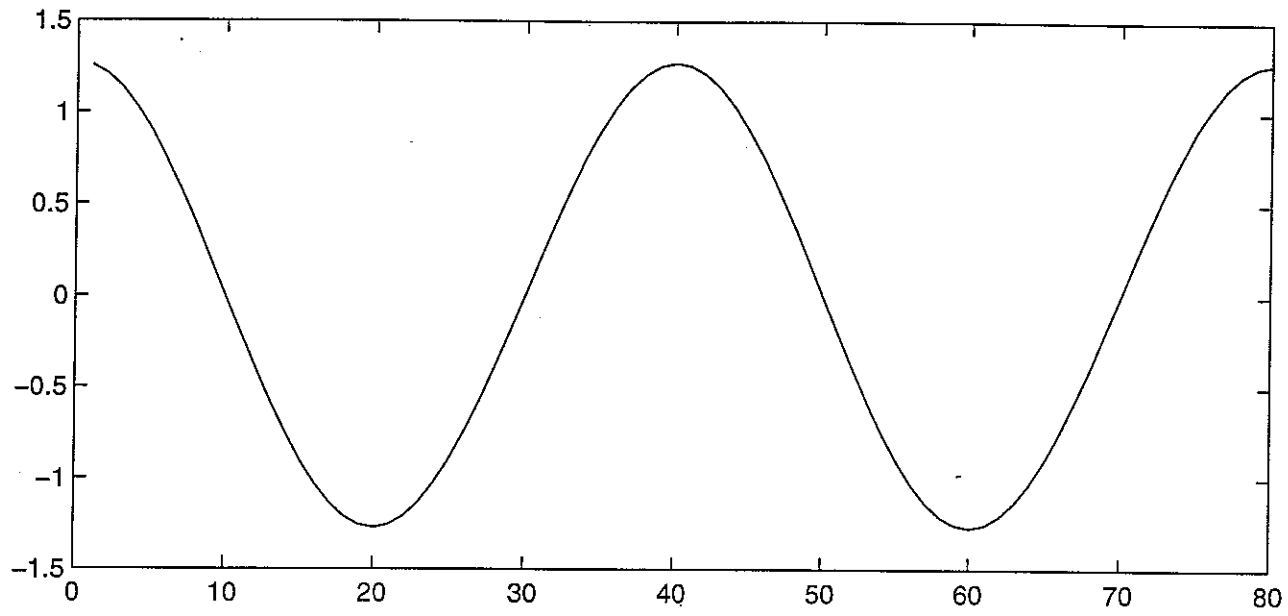
$$P_g = C_0^2 + \sum_{n=1}^{\infty} \frac{C_n^2}{2} = C_0^2 + \frac{1}{2} \sum_{n=1}^{\infty} C_n^2 = a_0^2 + \frac{1}{2} \sum (a_n^2 + b_n^2)$$

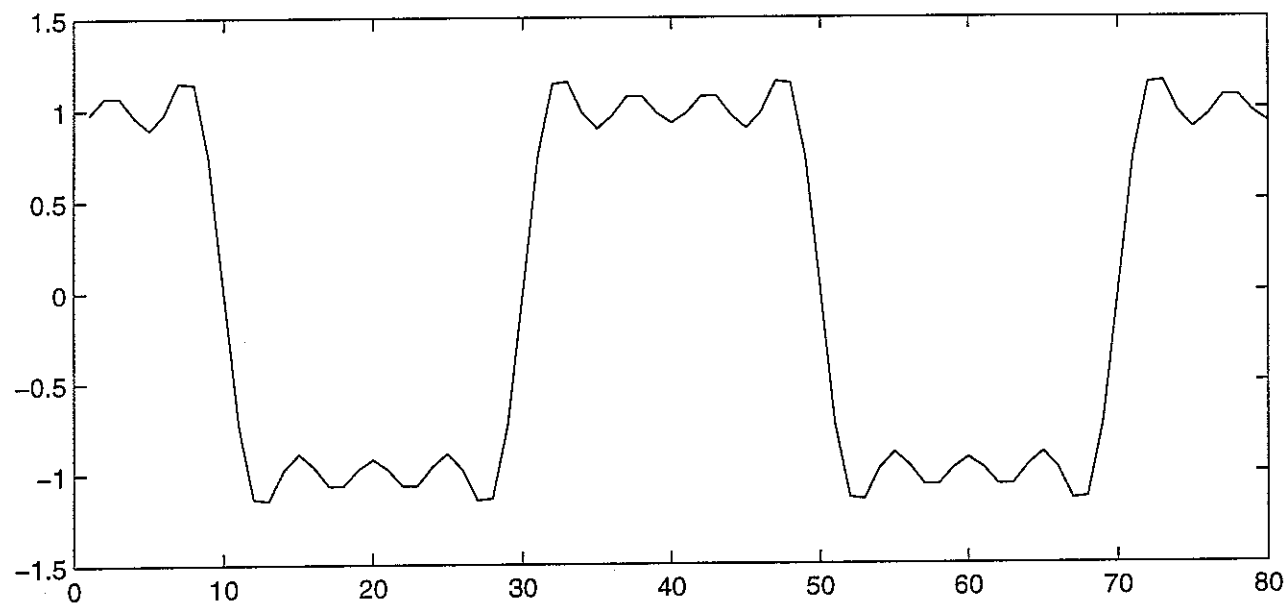
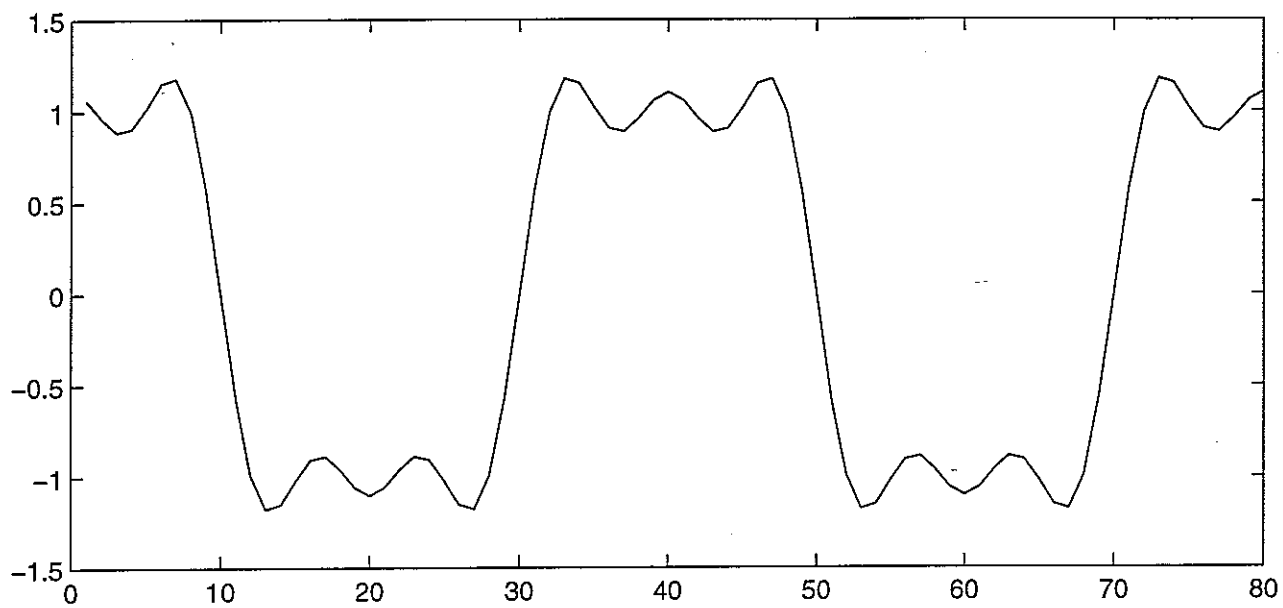
Similarly, for

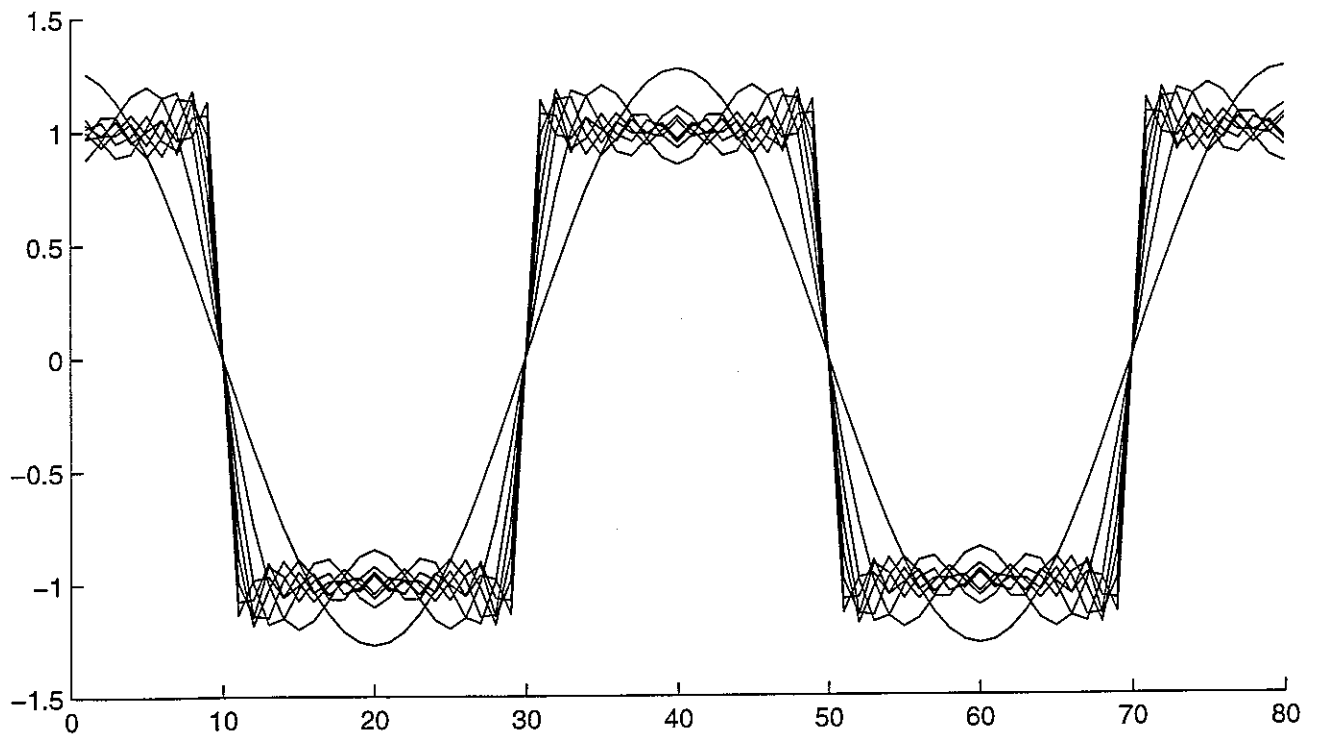
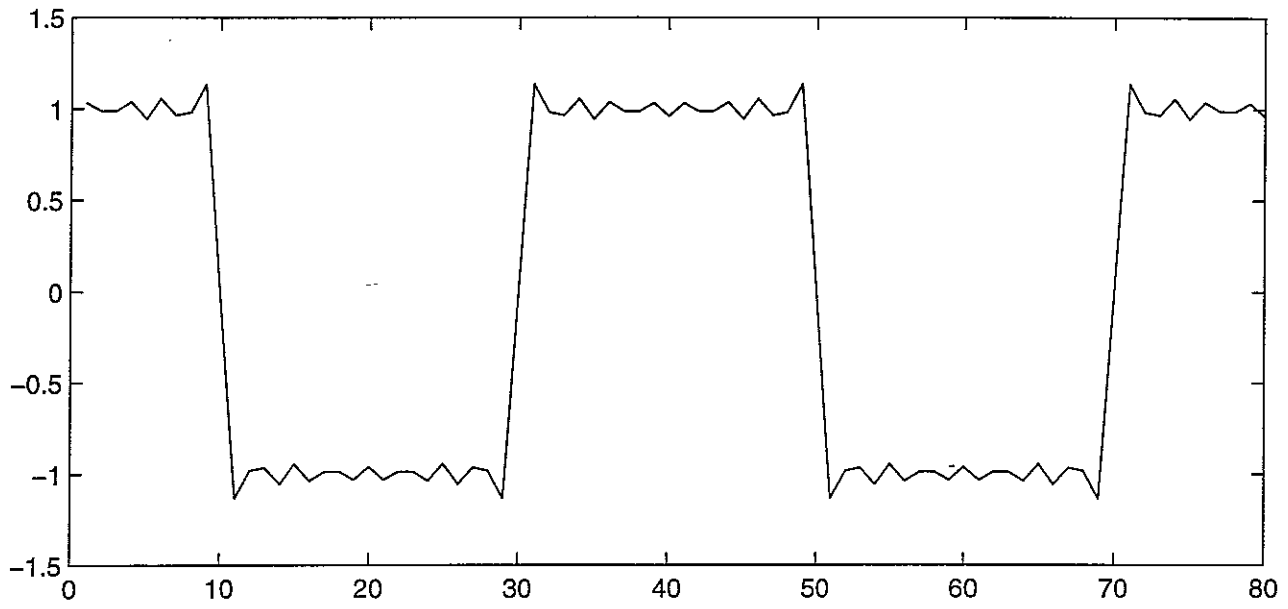
$$g(x) = D_0 + \sum_{\substack{n=-\infty \\ n \neq 0}}^{\infty} D_n e^{jn\omega_0 t}$$

$$P_g = \sum_{n=-\infty}^{\infty} |D_n|^2 = |D_0|^2 + \sum_{\substack{n=-\infty \\ n \neq 0}}^{\infty} |D_n|^2$$

```
function sq=sq(length,waves,period)
clf
S=period/2;
temp=1;
c=4/pi;
for k=1:length
x(k)=0;
end
for i=1:waves
j=2*i-1;
for k=1:length
x(k)=x(k)+c*temp*cos(pi*k*j/S)/j;
end
temp=temp*(-1);
hold
plot(x);
%hold
end
```







Fourier Transforms:

Fourier Transforms are used for aperiodic signals. One can derive Fourier Transform relationships by considering an aperiodic function as a periodic function with period ∞ (as $T_0 \rightarrow \infty$).

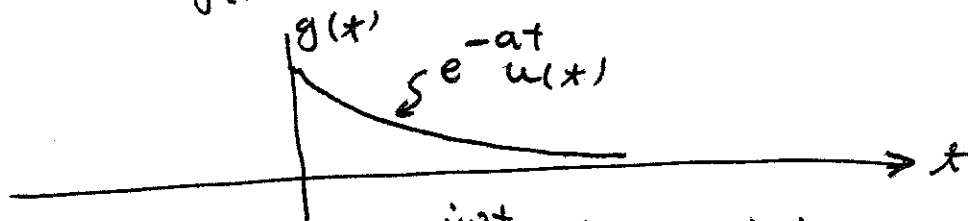
$$g(t) \iff G(\omega)$$

$$G(\omega) = \int_{-\infty}^{\infty} g(t) e^{-j\omega t} dt$$

$$g(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} G(\omega) e^{j\omega t} d\omega$$

example:

$$g(t) = e^{-at} u(t)$$



$$G(\omega) = \int_{-\infty}^{\infty} e^{-at} u(t) e^{j\omega t} dt = \int_0^{\infty} e^{-(a+j\omega)t} dt$$

$$G(\omega) = \frac{-1}{a+j\omega} e^{-(a+j\omega)t} \Big|_0^{\infty} = \frac{1}{a+j\omega} \quad a > 0$$

$$|G(\omega)| = \frac{1}{\sqrt{a^2 + \omega^2}}$$

$$\angle G(\omega) = -\tan^{-1}\left(\frac{\omega}{a}\right)$$

Existence of F. T.:

$$\int_{-\infty}^{\infty} |g(x)| dx < \infty .$$

This is a sufficient but not necessary Condition.

Linearity of F. T.

$$\text{if } g_1(x) \Leftrightarrow G_1(\omega)$$

and

$$g_2(x) \Leftrightarrow G_2(\omega)$$

the

$$a_1 g_1(x) + a_2 g_2(x) \Leftrightarrow a_1 G_1(\omega) + a_2 G_2(\omega)$$

F. T. of some Basic functions:

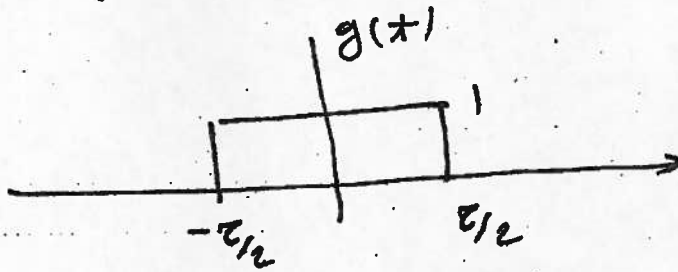
$$g(x) = \delta(x)$$

$$\mathcal{F}[\delta(x)] = \int_{-\infty}^{\infty} \delta(x) e^{-j\omega x} dx = 1 \quad \text{all } \omega$$



X Lecture 2:

gate function:

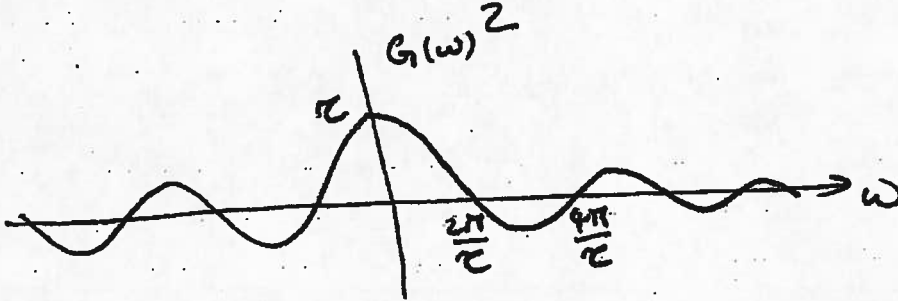


$$g(t) = \begin{cases} 0 & |t| > \frac{\tau}{2} \\ 1 & |t| < \frac{\tau}{2} \end{cases} = \text{rect}\left(\frac{t}{\tau}\right)$$

$$G(\omega) = \int_{-\tau/2}^{\tau/2} 1 \times e^{-j\omega t} dt = \frac{-1}{j\omega} e^{-j\omega t} \Big|_{-\tau/2}^{\tau/2}$$

$$G(\omega) = \frac{e^{j\omega \frac{\tau}{2}} - e^{-j\omega \frac{\tau}{2}}}{j\omega} = \frac{2 \sin\left(\frac{\omega \tau}{2}\right)}{\omega}$$

$$G(\omega) = \tau \frac{\sin\left(\frac{\omega \tau}{2}\right)}{\frac{\omega \tau}{2}} = \tau \text{sinc}\left(\frac{\omega \tau}{2}\right)$$



$$\tau \rightarrow \infty \Rightarrow g(t) \rightarrow 1 \quad \& \quad G(\omega) \rightarrow \delta(\omega)$$

$$\tau \rightarrow 0 \Rightarrow \frac{g(t)}{\tau} \rightarrow \delta(t) \quad \& \quad \frac{G(\omega)}{\tau} \rightarrow 1$$

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$$\mathcal{F}^{-1}[\delta(\omega)] = \frac{1}{2\pi} \int_{-\infty}^{\infty} \delta(\omega) e^{j\omega t} d\omega = \frac{1}{2\pi}$$

$$1 \iff 2\pi \delta(\omega)$$

$$\mathcal{F}^{-1}[\delta(\omega - \omega_0)] = \frac{1}{2\pi} \int_{-\infty}^{\infty} \delta(\omega - \omega_0) e^{j\omega t} d\omega$$
$$= \frac{1}{2\pi} e^{j\omega_0 t}$$

$$\frac{1}{2\pi} e^{j\omega_0 t} \iff \delta(\omega - \omega_0)$$

or

$$e^{j\omega_0 t} \iff 2\pi \delta(\omega - \omega_0)$$

Similarly

$$e^{-j\omega_0 t} \iff 2\pi \delta(\omega + \omega_0)$$

$$\cos \omega_0 t = \frac{1}{2} e^{j\omega_0 t} + \frac{1}{2} e^{-j\omega_0 t}$$

So

$$\mathcal{F}[\cos \omega_0 t] = \pi \delta(\omega + \omega_0) + \pi \delta(\omega - \omega_0)$$

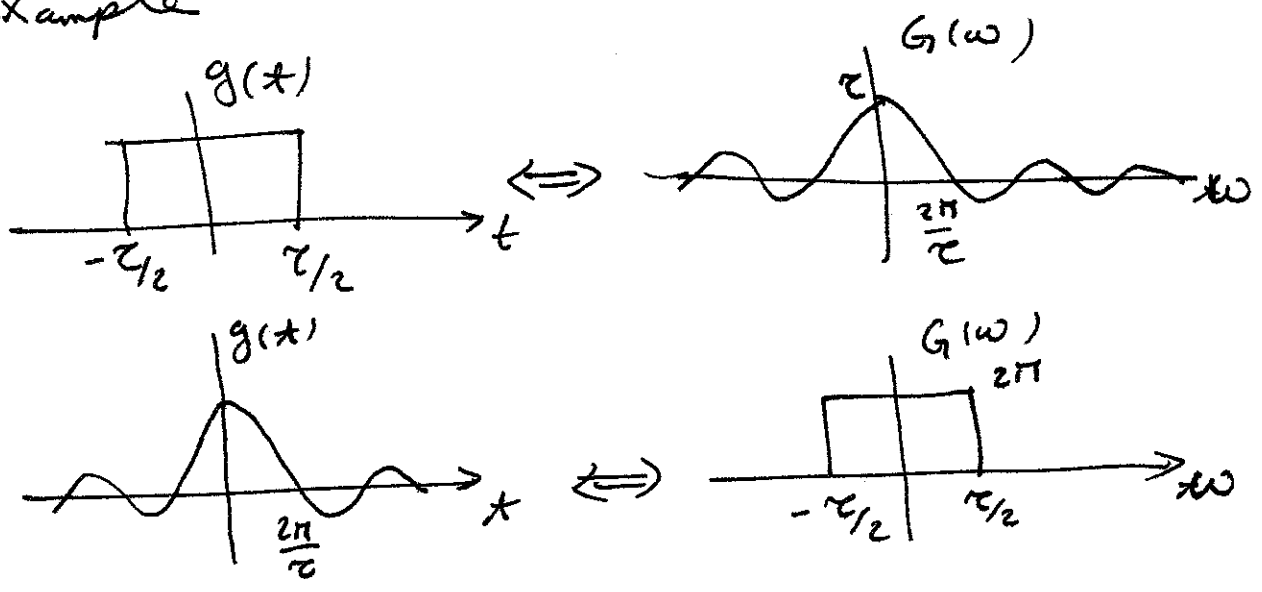
Some properties of F.T.

- 1) $g(t-t_0) \Leftrightarrow e^{-j\omega t_0} G(\omega)$
- 2) $g(t)e^{j\omega_0 t} \Leftrightarrow G(\omega - \omega_0)$
 $g(t)\cos\omega_0 t = g(t)\left[\frac{1}{2}e^{j\omega_0 t} + \frac{1}{2}e^{-j\omega_0 t}\right] \Leftrightarrow \frac{1}{2}[G(\omega + \omega_0) + G(\omega - \omega_0)]$
- 3) if $g(t) \Leftrightarrow G(\omega)$

then

$$G(t) \Leftrightarrow 2\pi g(-\omega)$$

example



4) Scaling Property

if $g(t) \Leftrightarrow G(\omega)$
 then $g(at) \Leftrightarrow \frac{1}{|a|} G\left(\frac{\omega}{a}\right)$

$$5) \quad g(x) \cos \omega_0 t = \frac{1}{2} g(x) e^{j\omega_0 t} + \frac{1}{2} g(x) e^{-j\omega_0 t}$$

$$\mathcal{F}[g(x) \cos \omega_0 t] = \mathcal{F}[G(\omega - \omega_0) + G(\omega + \omega_0)]$$

6) Convolution

$$\mathcal{F}[g(x) * f(x)] = G(\omega) F(\omega)$$

where

$$g(x) * f(x) = \int_{-\infty}^{\infty} g(\tau) f(x - \tau) d\tau.$$

7) Derivation

$$\text{if } g(x) \Leftrightarrow G(\omega)$$

$$\text{then } \frac{dg(x)}{dt} \Leftrightarrow j\omega G(\omega)$$

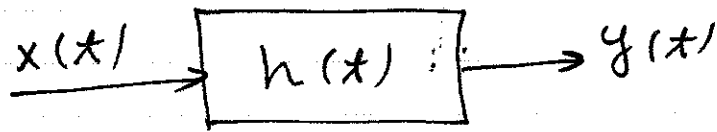
8) Integration

$$\int_{-\infty}^x g(\tau) d\tau \Leftrightarrow \frac{G(\omega)}{j\omega} + \pi G(0) \delta(\omega)$$

9) Parseval's theorem

$$E_g = \int_{-\infty}^{\infty} |g(x)|^2 dx = \frac{1}{2\pi} \int_{-\infty}^{\infty} |G(\omega)|^2 d\omega$$

Signal Transmission through a L.T.I. System



where $h(t)$ is the unit impulse response of the system, i.e.,

$$\mathcal{L}[\delta(t)] = h(t)$$

then

$$y(t) = x(t) * h(t) = \int_{-\infty}^{\infty} x(\tau) h(t-\tau) d\tau$$

and

$$Y(\omega) = X(\omega) H(\omega)$$

where $H(\omega)$ is the F.T. of $h(t)$ and is called the transfer function of the system.

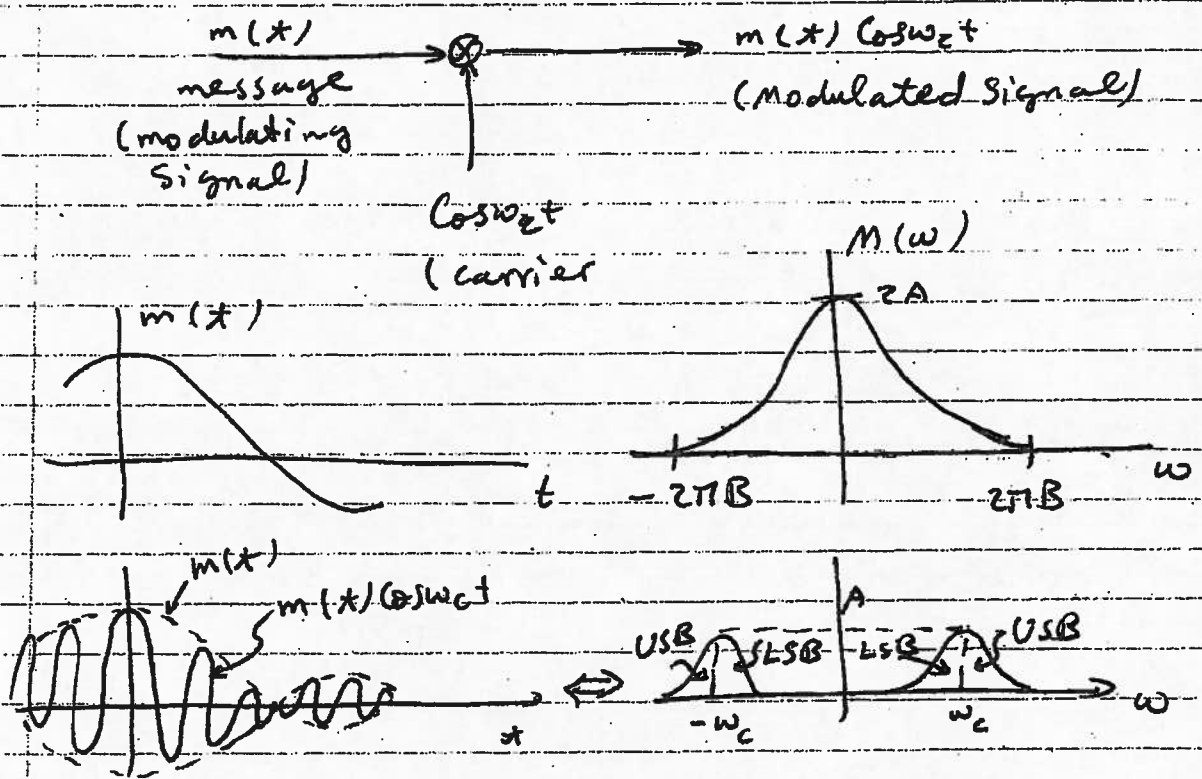
Modulation:

- Why modulation? - ease of transmission
- multiplexing.

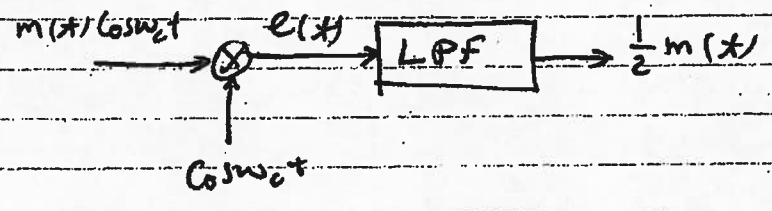
Amplitude Modulation: DSB-SC

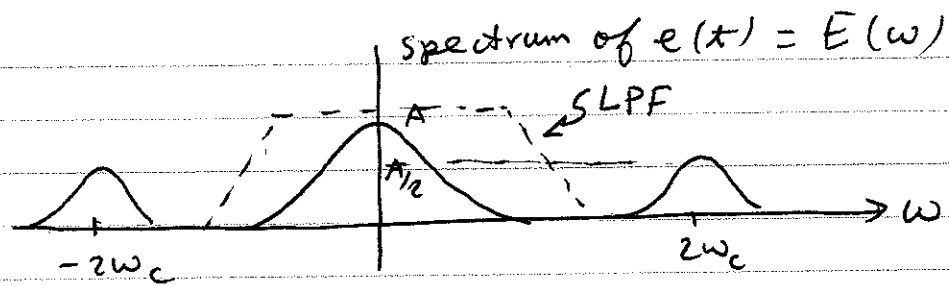
$$m(t) \iff M(\omega)$$

$$m(t) \cdot \cos \omega_c t \iff \frac{1}{2} [M(\omega + \omega_c) + M(\omega - \omega_c)]$$



Demodulation:





$$e(t) = m(t) \cos^2 \omega_c t = \frac{1}{2} [m(t) + m(t) \cos 2\omega_c t]$$

$$\text{LPF} \rightarrow \frac{1}{2} m(t)$$

$$E(\omega) = \frac{1}{4} [M(\omega) + M(\omega + 2\omega_c) + M(\omega - 2\omega_c) + M(\omega)]$$

$$= \frac{1}{2} M(\omega) + \frac{1}{4} [M(\omega + 2\omega_c) + M(\omega - 2\omega_c)]$$

This is called Synchronous or Coherent detection.

Modulation Techniques:

1- Multiplier Modulator

using direct multiplication, e.g., through

an amplifier whose gain is controlled

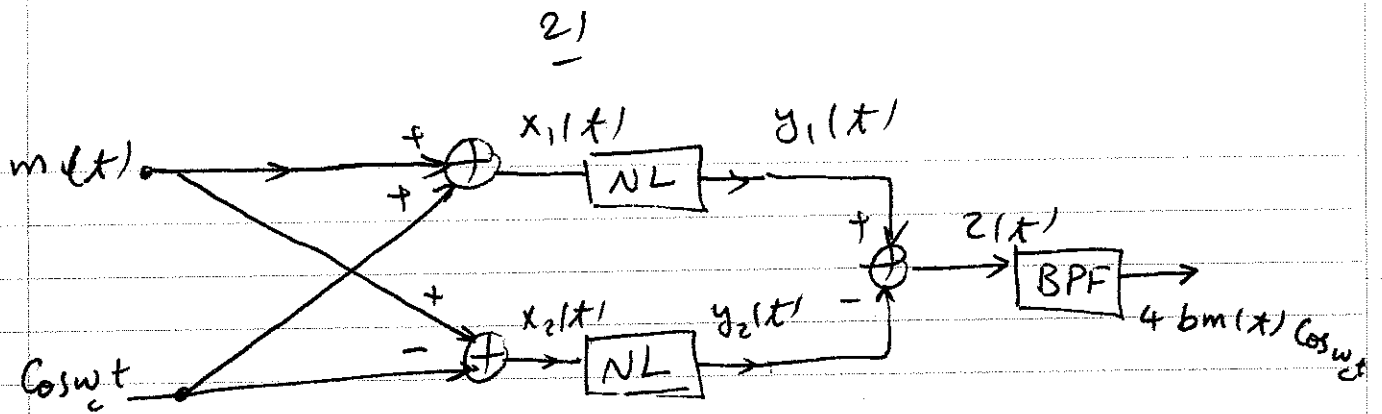
by $m(t)$ and amplifying $\cos \omega_c t$.

It is difficult to maintain linearity,

therefore, expensive to implement.

2) Non-linearity modulators:

$$y(t) = ax(t) + bx^2(t)$$



$$x_1(t) = m(t) + \cos \omega_c t$$

$$x_2(t) = m(t) - \cos \omega_c t$$

$$y_1(t) = a[m(t) + \cos \omega_c t] + b[m(t) + \cos \omega_c t]^2$$

$$y_2(t) = a[m(t) - \cos \omega_c t] + b[m(t) - \cos \omega_c t]^2$$

$$z(t) = y_1(t) - y_2(t) = 2am(t) + \underbrace{4bm(t)\cos \omega_c t}$$

$$z(t) \xrightarrow{\text{LPF}} 4bm(t)\cos \omega_c t$$

3) Switching Modulator

A periodic wave, e.g., a square wave

can be represented as:

$$g(t) = \sum_{n=0}^{\infty} C_n \cos(n\omega_c t + \phi_n)$$

So

$$m(t)g(t) = \sum_{n=0}^{\infty} C_n m(t) \cos(n\omega_c t + \phi_n)$$

X Lecture 3:

For example:

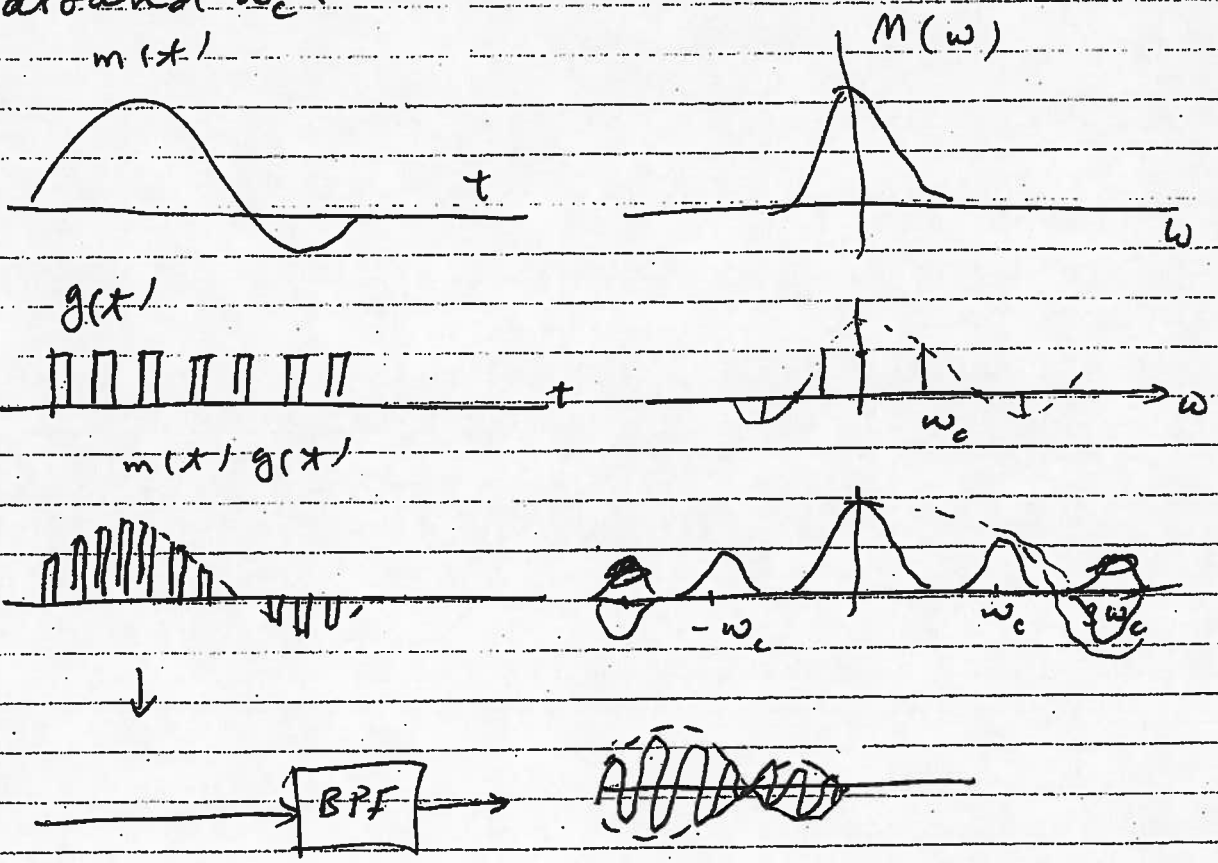
$$g(x) = \frac{1}{2} + \frac{2}{\pi} \left(\cos \omega_c t - \frac{1}{3} \cos 3\omega_c t + \frac{1}{5} \cos 5\omega_c t - \dots \right)$$

is the representation for a square wave

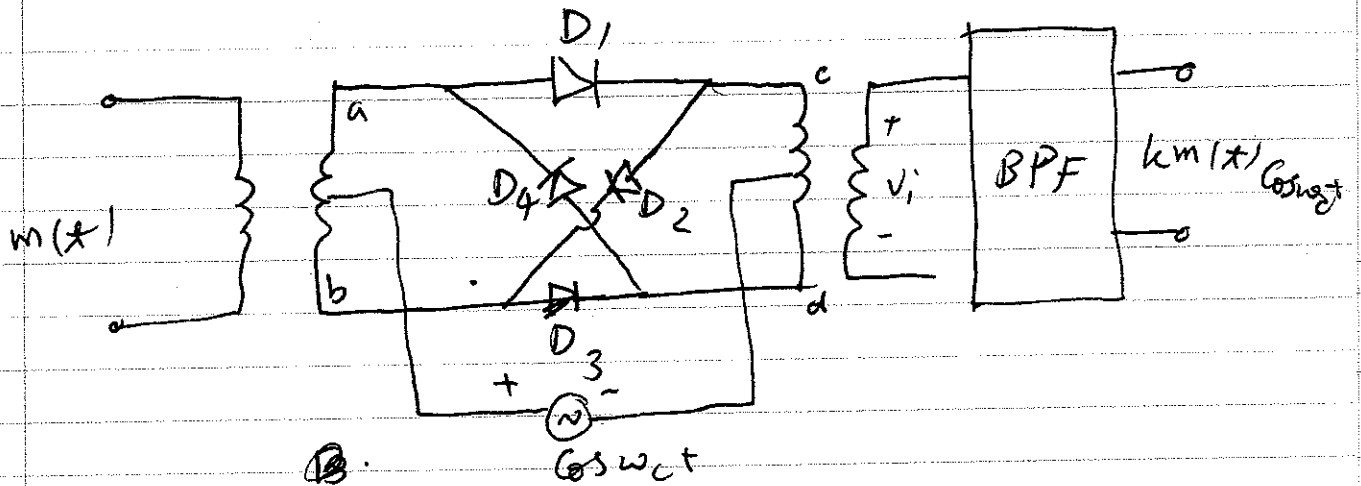
so

$$m(x)g(x) = \frac{1}{2}m(x) + \frac{2}{\pi}m(x)\cos\omega_c t - \frac{2}{3\pi}m(x)\cos 3\omega_c t + \dots$$

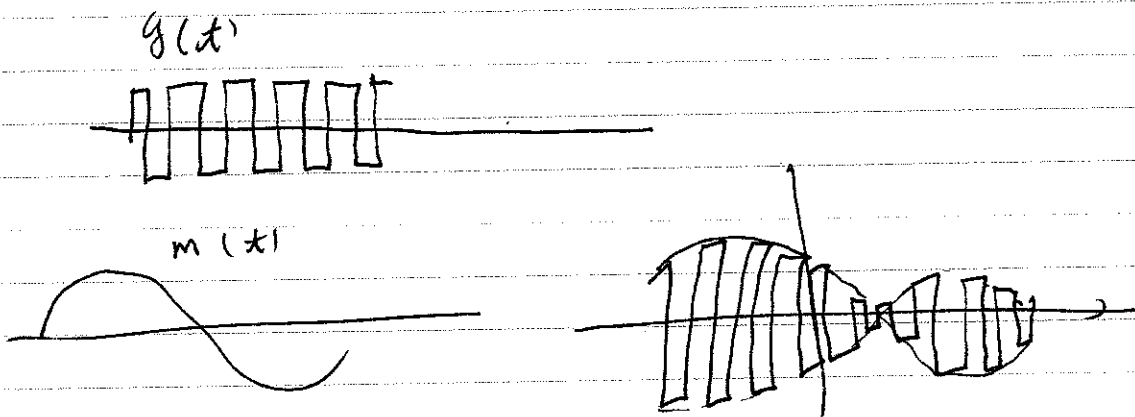
we can separate $\frac{2}{\pi}m(x)\cos\omega_c t$ by a BPF centered around ω_c .



Ring Modulator



Double Balanced Modulator.



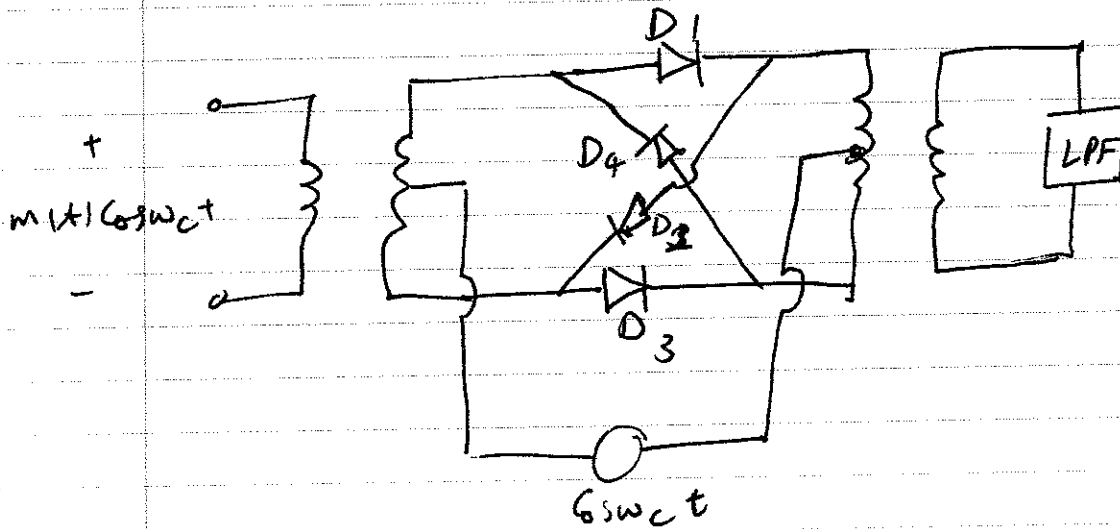
$$g(t) = \frac{4}{\pi} \left(\cos \omega_c t - \frac{1}{3} \cos 3\omega_c t + \frac{1}{5} \cos 5\omega_c t + \dots \right)$$

$$g(t)m(t) = \frac{4}{\pi} \left[m(t) \cos \omega_c t - \frac{1}{3} m(t) \cos 3\omega_c t + \dots \right]$$

Demodulation of DSB-SC signals:

Any of the modulator circuits discussed can be used for demodulation (followed by a LPF)

For example, we can use the Ring-Modulator;



The input to the LPF is

$$m(t) \cos \omega_c t \cdot g(t) = m(t) \cos \omega_c t \left[\frac{4}{\pi} (\cos \omega_c t - \frac{1}{3} \cos 3\omega_c t + \dots) \right]$$
$$= \frac{2}{\pi} m(t) + \frac{2}{\pi} m(t) \cos 2\omega_c t + \text{other terms}$$

after LPF we get $\frac{2}{\pi} m(t)$

AM (or DSB-LC)

need for carrier acquisition:

To avoid phase recovery (carrier Acquisition) we add unmodulated carrier

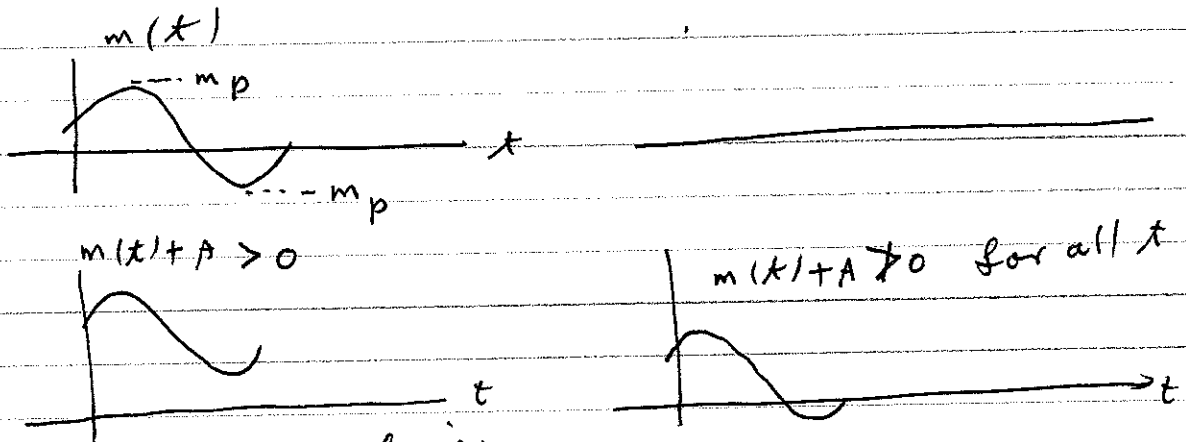
$$\begin{aligned} \phi_{AM}(t) &= A \cos \omega_c t + m(t) \cos \omega_c t \\ &= [A + m(t)] \cos \omega_c t \end{aligned}$$

$$\begin{aligned} \phi_{AM}(\omega) &= \frac{1}{2} [M(\omega + \omega_c) + M(\omega - \omega_c) + \pi A \delta(\omega + \omega_c) \\ &\quad + \pi A \delta(\omega - \omega_c)] \end{aligned}$$

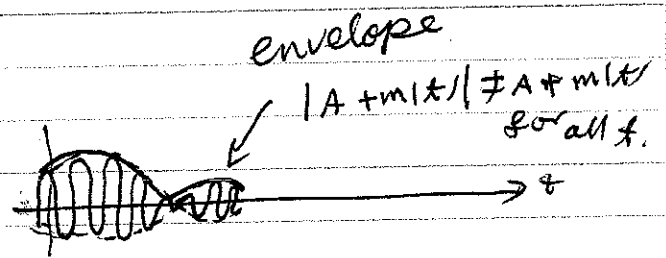
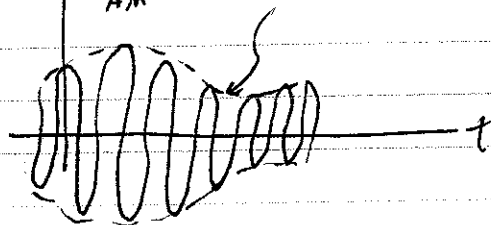
In order to be able to recover $m(t)$ from

$\phi_{AM}(t)$ easily, we need $A + m(t) > 0$ all t

Then we can use envelope detector:



then $\phi_{AM}(t)$ envelope is: $m(t) + A$



$$A + m(x) \geq 0 \quad \text{all } x$$

So

$$A \geq m_p$$

where $-m_p$ is the minimum message amplitude;

$$\mu = \frac{m_p}{A} \equiv \text{modulation index.}$$

So, we need

$$0 \leq \mu \leq 1$$

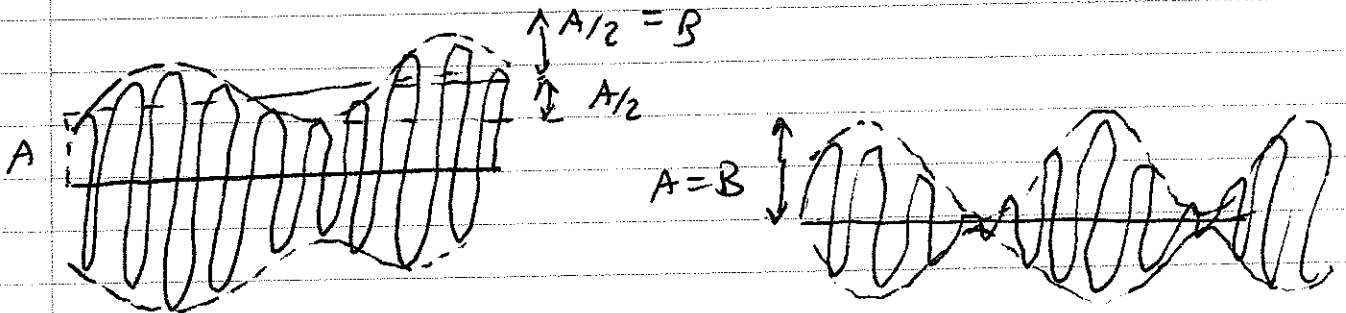
in order to be able to use envelope ~~demodulation~~ detector.

when $A < m_p$ then $\mu > 1$ (overmodulation),
then we cannot use envelope detector.

example:

Sketch $\phi_{AM}(x)$ for modulation indexes

$\mu = 0.5$ and $\mu = 1$ when $m(x) = B \cos \omega_m t$



$$\mu = 0.5 = \frac{m_p}{A} = \frac{B}{A} \Rightarrow A = 2B$$

$$\mu = 1 = \frac{m_p}{A} = \frac{B}{A} \Rightarrow B = A$$

$A = B$

Handwritten signature

Sideband and carrier Power

$$\phi_{AM}(t) = \underbrace{A \cos \omega_c t}_{\text{unmodulated (Pure) Carrier}} + \underbrace{m(t) \cos \omega_c t}_{\text{modulated information (sidebands)}}$$

$$P_c = \frac{A^2}{2} \quad \text{and} \quad P_s = \frac{1}{2} \overline{m^2(t)}$$

efficiency of modulation:

$$\eta = \frac{\text{useful power}}{\text{Total Power}} = \frac{P_s}{P_c + P_s} = \frac{\overline{m^2(t)}}{A^2 + \overline{m^2(t)}} \times 100\%$$

For a sinusoidal message (tone)

$$m(t) = \mu A \cos \omega_c t \Rightarrow \overline{m^2(t)} = \frac{(\mu A)^2}{2}$$

$$\eta = \frac{\mu^2}{2 + \mu^2} 100\%$$

the maximum of η is achieved with $\mu = 1$

(since $0 \leq \mu \leq 1$), i.e.,

$$\eta_{\max} = 33\frac{1}{3}\%$$

Example: Determine the efficiency of modulation and the per cent of power in the sidelobes when a) $\mu = 0.5$ b) $\mu = 0.3$

$$a) \quad \eta = \frac{\mu^2}{2 + \mu^2} \times 100\% = \frac{(0.5)^2}{2 + (0.5)^2} \times 100\% = 11.11\%$$

So 11% of power is in the sidelobes, i.e., is useful power.

$$b) \quad \eta = \frac{(0.3)^2}{2 + 0.3^2} \times 100\% = 4.3\%$$

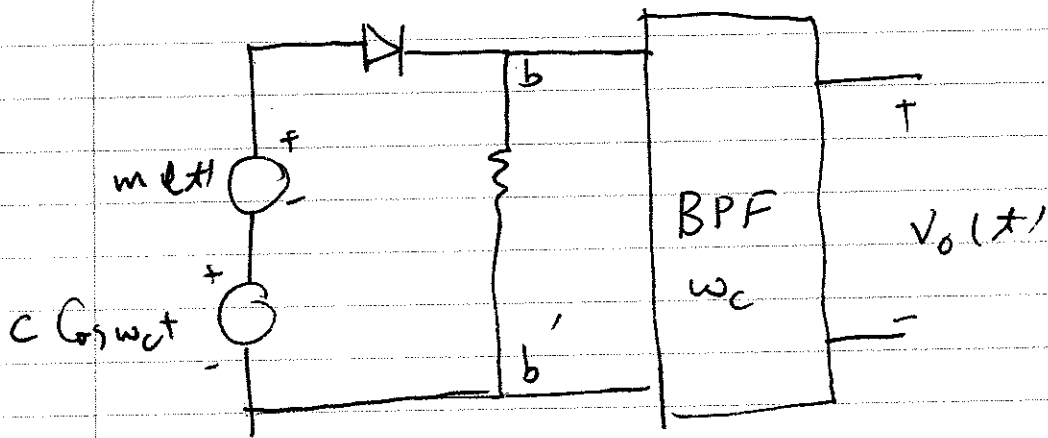
only 4.3% of the power is in the sidebands.

Generation of AM Signals

The same as DSB-SC but with input as $A + m(t)$ instead of $m(t)$.

However, since there is no need to suppress the carrier, the circuit does not need to be balanced. So, a single diode can be used.

The following circuit is an example:



$$V_{bb'} = [c \cos \omega_c t + m(t)] w(t)$$

where $w(t)$ is a square wave, since

diode shorts and opens whenever

the sinusoidal term is positive/negative,

respectively ($c \gg m(t)$)

So:

$$V_{bb'}(t) = [c \cos \omega_c t + m(t)] \left[\frac{1}{2} + \frac{2}{\pi} \left(\cos \omega_c t - \frac{1}{3} \cos 3\omega_c t + \dots \right) \right]$$

$$= \underbrace{\frac{c}{2} \cos \omega_c t + \frac{2}{\pi} m(t) \cos \omega_c t}_{\text{desired AM signal}} + \underbrace{\text{other terms}}_{\text{to be filtered out by BPF}}$$

desired AM signal

to be
filtered
out by
BPF

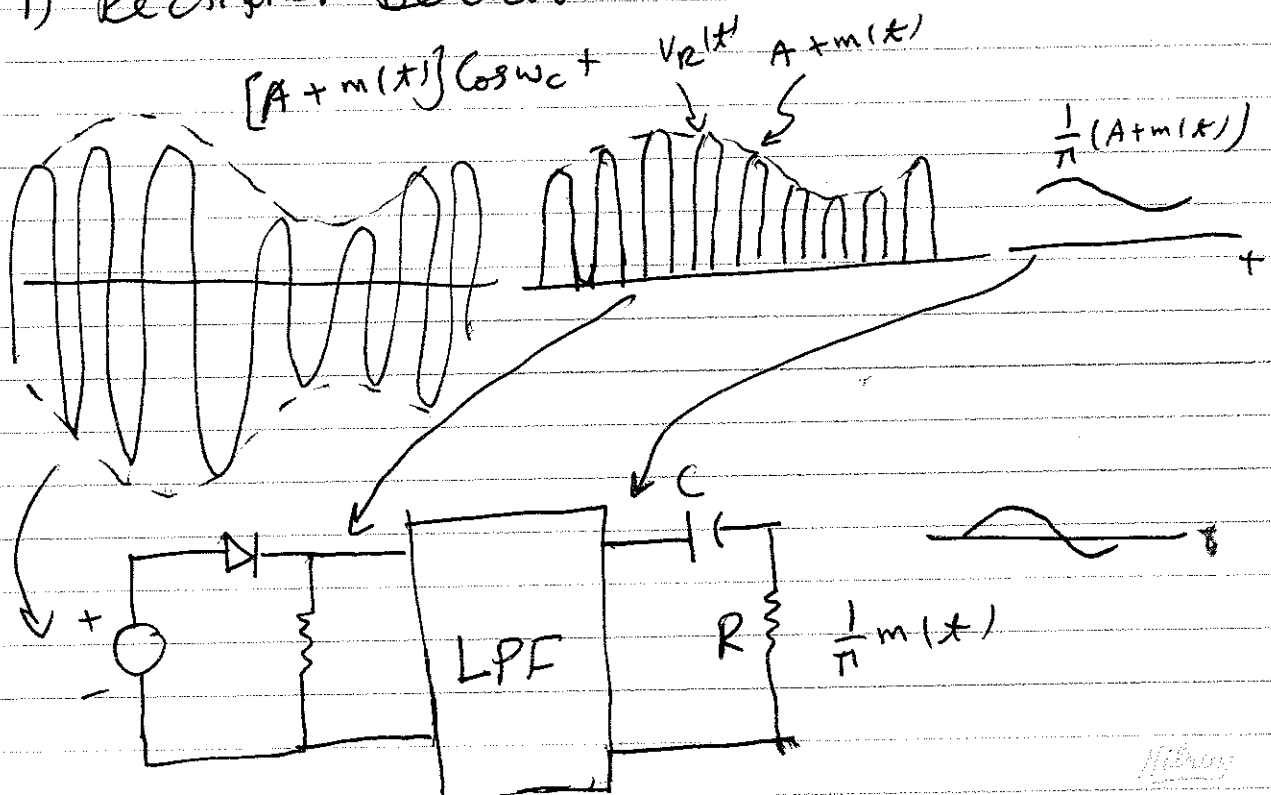
Demodulation of AM Signals:

Coherent demodulation is possible, however, defeats the purpose of having extra carrier added.

So, non-coherent demodulation is preferred. Examples of non-coherent AM demodulators are:

- 1) Rectifier Detector
- 2) Envelope Detector.

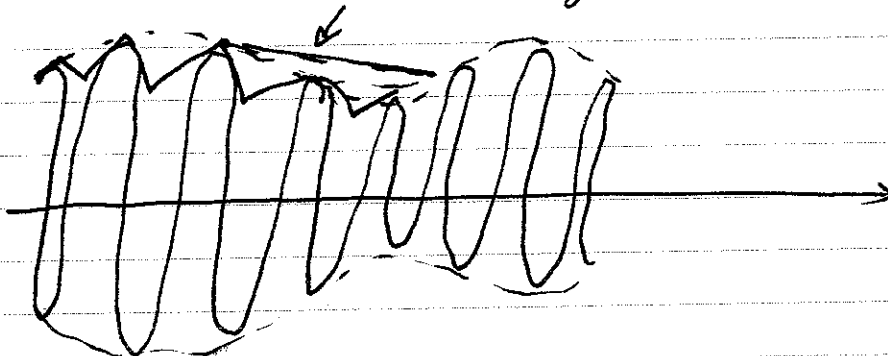
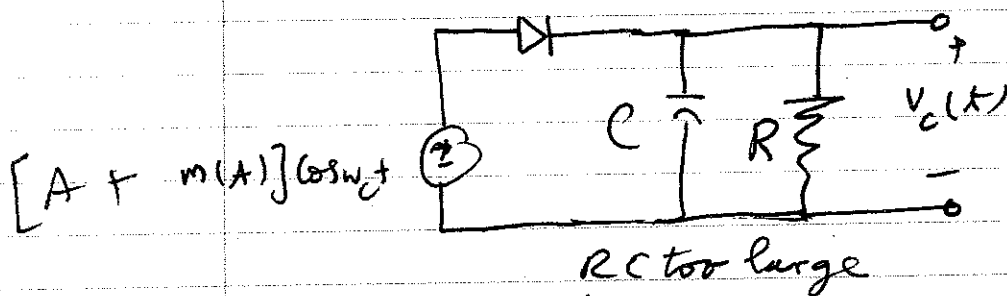
1) Rectifier Detector



$$\begin{aligned}
 V_R &= \{ [A + m(x)] \cos \omega_c t \} w(x) \\
 &= [A + m(x)] \cos \omega_c t \left[\frac{1}{2} + \frac{2}{\pi} (\cos \omega_c t - \frac{1}{3} \cos 3\omega_c t + \frac{1}{5} \cos 5\omega_c t) \right] \\
 &= \frac{1}{\pi} [A + m(x)] + \text{other terms at higher frequencies.}
 \end{aligned}$$

2) Envelope Detector.

Here again, we have a half-wave rectifier followed by a LPF. However, here the purpose of the LPF is to follow the envelope of the waveform and is designed taking into consideration the modulation index m .



The operation is as follows:

during positive cycle of $\phi_{AM}(t)$, diode conducts and C charges to the max value of the input signal. As signal falls below this value, C starts discharging, but diode reverses preventing it to discharge fast (C discharges through R and the diode). The same happens in the next cycle, ...

To see the relationship between the parameters of filter and the modulation index note that,

$$V_c = E e^{-t/RC}$$

where E is the peak value at $t = 0$.

since $RC \gg \frac{1}{\omega_c}$

then the capacitor voltage can be approximated by a straight line. Using Taylor series,

we get:

$$V_c \approx E \left(1 - \frac{t}{RC} \right)$$

The slope of discharge is $-\frac{E}{RC}$

In order for the capacitor to follow the envelope $E(t)$, the magnitude of slope of RC discharge must be greater than the magnitude of slope of the envelope $E(t)$

$$\left| \frac{dV_c}{dt} \right| = \frac{E}{RC} \geq \left| \frac{dE}{dt} \right|$$

But, for a tone : $E = A[1 + \mu \cos \omega_m t]$

$$\frac{dE}{dt} = \frac{d}{dt} \{ A[1 + \mu \cos \omega_m t] \}$$

$$\text{or } \frac{dE}{dt} = -\mu A \omega_m \sin \omega_m t$$

Hence, we need

$$\frac{A[1 + \mu \cos \omega_m t]}{RC} \geq \mu A \omega_m \sin \omega_m t \quad \text{for all } t.$$

or

$$RC \leq \frac{1 + \mu \cos \omega_m t}{\mu \omega_m \sin \omega_m t}$$

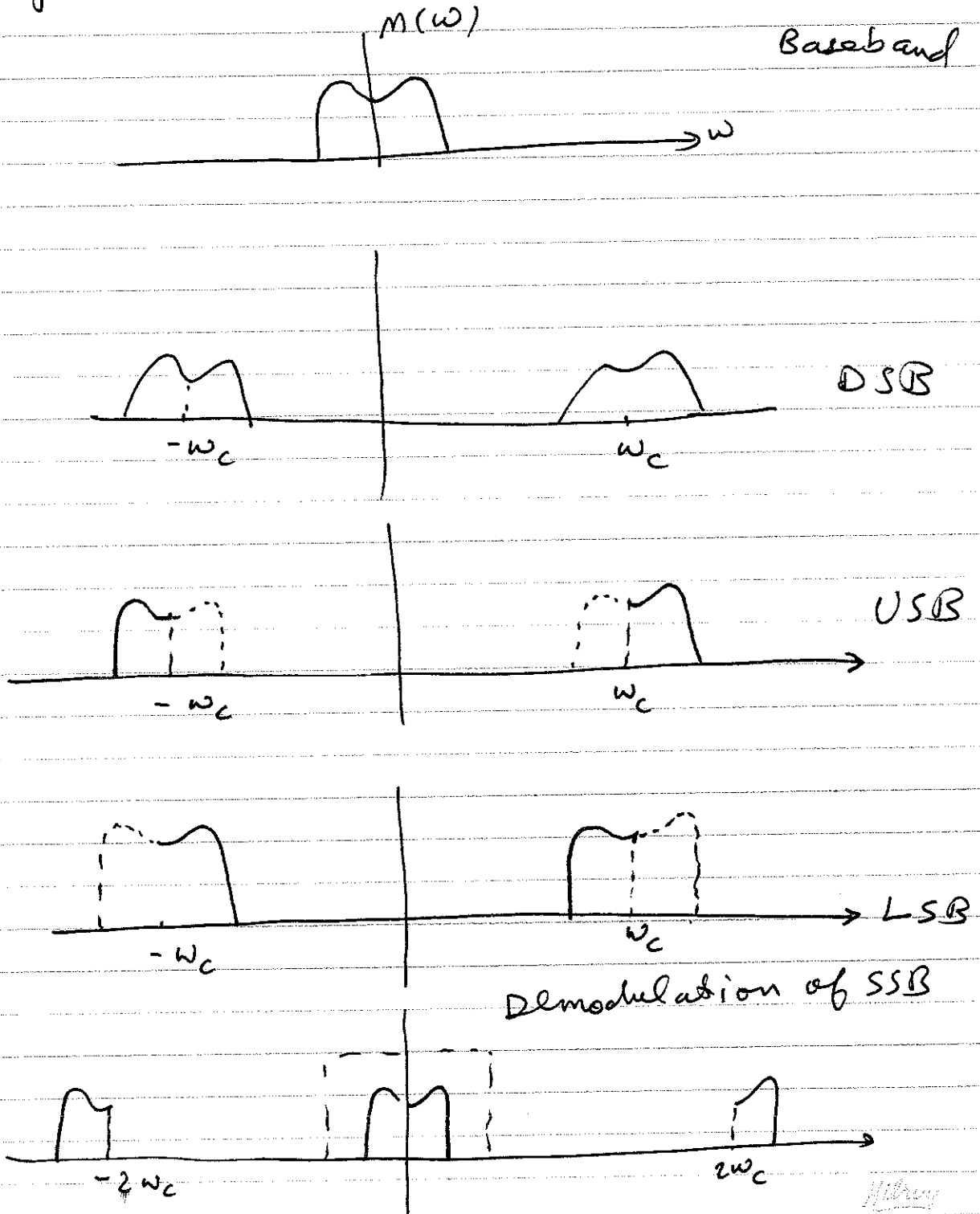
The worst case happens when the right hand side is minimum

$$\frac{d}{dt} \left[\frac{1 + \mu \cos \omega_m t}{\mu \omega_m \sin \omega_m t} \right] = 0 \Rightarrow \cos \omega_m t = -\mu$$

So, we required that:

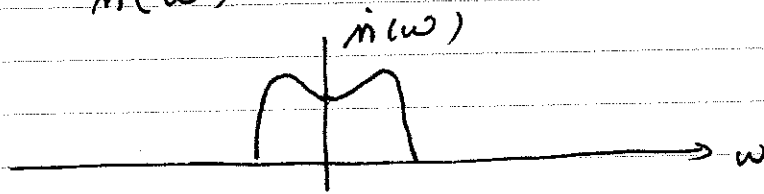
$$RC \leq \frac{1}{\omega_m} \left(\frac{\sqrt{1-\mu^2}}{\mu} \right)$$

Single side band (SSB) Modulation

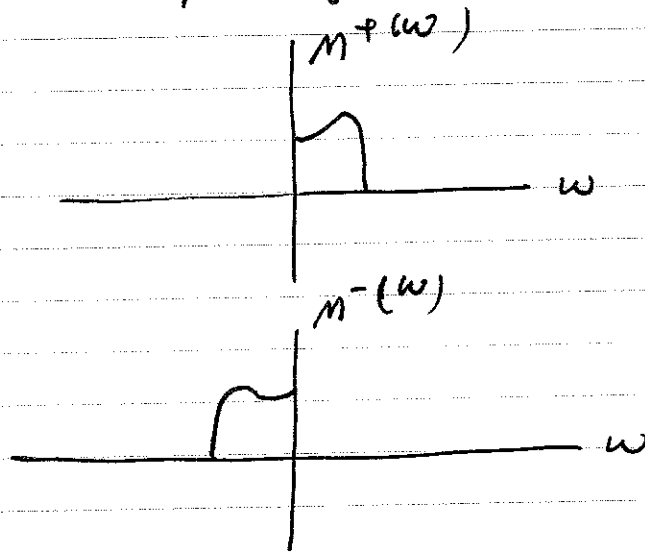


Time-domain representation of SSB

Consider $m(\omega)$ as,

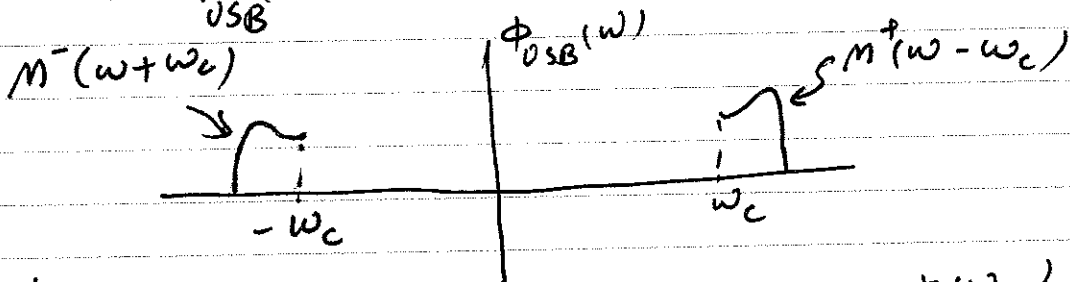


Define $M^+(\omega)$ and $M^-(\omega)$ as the positive and negative frequency portions of $M(\omega)$



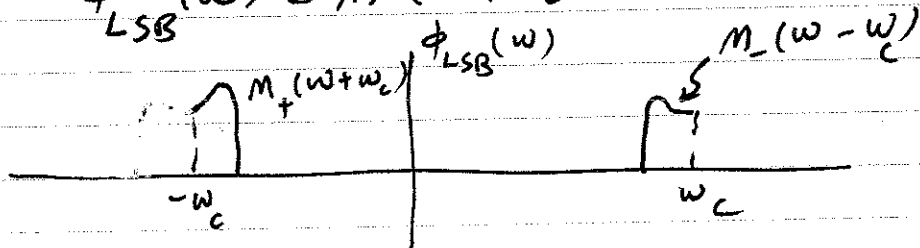
Then USB can be represented as

$$\Phi_{USB}(\omega) = M^+(\omega - \omega_c) + M^-(\omega + \omega_c)$$



and

$$\Phi_{LSB}(\omega) = M^+(\omega + \omega_c) + M^-(\omega - \omega_c)$$



But

$$M_+(w) = M(w) u(w)$$

where

$$u(w) = \begin{cases} 1 & w > 0 \\ 0 & w \leq 0 \end{cases}$$

note that

$$u(w) = \frac{1}{2} [1 + \text{sgn}(w)]$$

So:

$$M_+(w) = \frac{1}{2} M(w) + \frac{1}{2} M(w) \text{sgn}(w)$$

$$m_+(x) = \frac{1}{2} m(x) + \frac{1}{2} \mathcal{F}^{-1} [M(w) \text{sgn}(w)]$$

$$\mathcal{F}^{-1} [M(w) \text{sgn}(w)] = m(x) * \mathcal{F}^{-1} [\text{sgn}(w)]$$

$$\mathcal{F}^{-1} [\text{sgn}(w)] = \frac{j}{\pi t}$$

So:

$$m_+(x) = \frac{1}{2} m(x) + \frac{j}{2\pi} \int_{-\infty}^{\infty} \frac{m(d)}{x-d} dd$$

X Lecture 4:

$$\text{let } m_h(x) = \frac{1}{\pi} \int_{-\infty}^{\infty} \frac{m(d)}{x-d} dd$$

this is called the Hilbert Transform of $m(x)$

Then

$$m_+(x) = \frac{1}{2} [m(x) + j m_h(x)]$$

Similarly, it can be shown that

$$m_-(x) = \frac{1}{2} [m(x) - j m_h(x)]$$

$$\Phi_{\text{USB}}(\omega) = M_+(\omega - \omega_c) + M_-(\omega + \omega_c)$$

implies that

$$\phi_{\text{USB}}(x) = m_+(x) e^{+j\omega_c t} + m_-(x) e^{-j\omega_c t}$$

$$\phi_{\text{USB}}(x) = \frac{1}{2} [m(x) + jm_h(x)] e^{j\omega_c t} + \frac{1}{2} [m(x) - jm_h(x)] e^{-j\omega_c t}$$

$$\phi_{\text{USB}}(x) = m(x) \cos \omega_c t - m_h(x) \sin \omega_c t$$

Similarly,

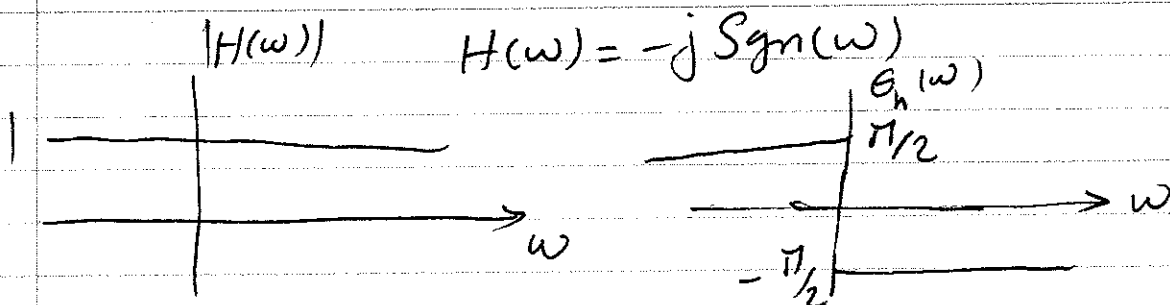
$$\phi_{\text{LSB}}(x) = m(x) \cos \omega_c t + m_h(x) \sin \omega_c t$$

$$M_h(\omega) = \mathcal{F}^{\text{FT}} [m_h(x)] = \mathcal{F}^{\text{FT}} [m(x) * \frac{1}{\pi t}] = -M(\omega) j \text{Sgn}(\omega)$$

$$M_h(\omega) = -j \text{Sgn}(\omega) M(\omega) = M(\omega) H(\omega)$$

So, finding $m_h(x)$ is ~~to~~ equivalent

to passing $m(x)$ through a filter with transfer function



Since

$$H(\omega) = -j \operatorname{sgn}(\omega) = \begin{cases} -j = 1 \cdot e^{-j\pi/2} & \omega > 0 \\ j = 1 \cdot e^{j\pi/2} & \omega < 0 \end{cases}$$

Example: Tone Modulation

$$m(x) = \cos \omega_m t$$

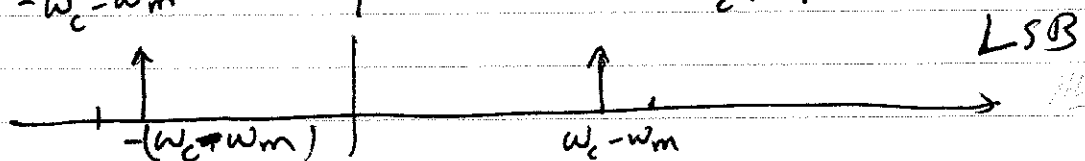
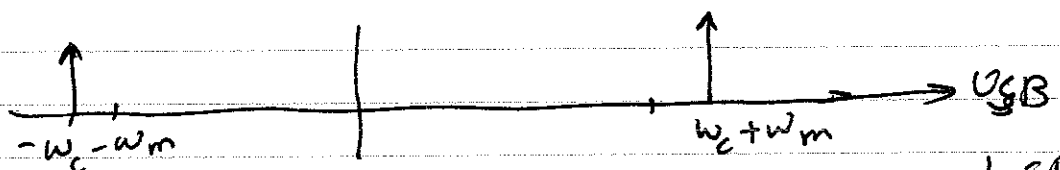
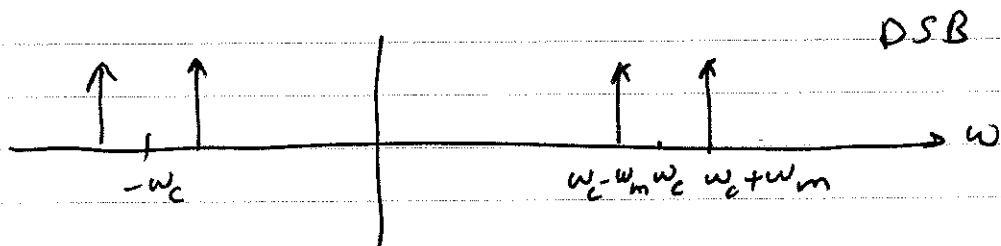
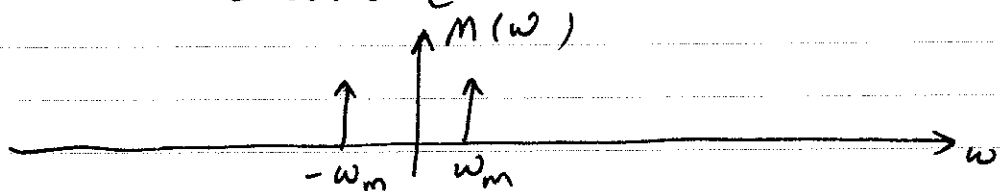
$$m_n(x) = \cos(\omega_m t - \frac{\pi}{2}) = \sin \omega_m t$$

So:

$$\begin{aligned} \phi_{\text{USB}}(t) &= \cos \omega_m t \cos \omega_c t - \sin \omega_m t \sin \omega_c t \\ &= \cos(\omega_c - \omega_m) t \end{aligned}$$

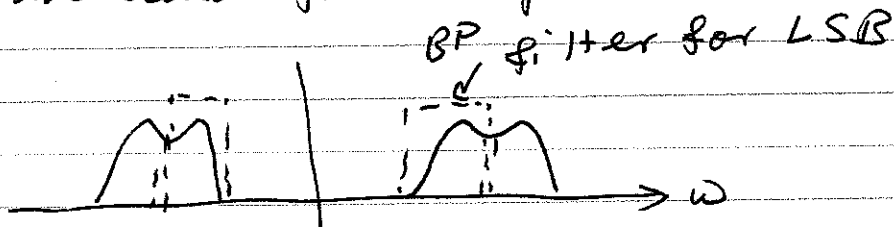
and

$$\begin{aligned} \phi_{\text{LSB}}(t) &= \cos \omega_m t \cos \omega_c t + \sin \omega_m t \sin \omega_c t \\ &= \cos(\omega_c + \omega_m) t \end{aligned}$$



Generation of SSB Signals:

1) Sideband filtering

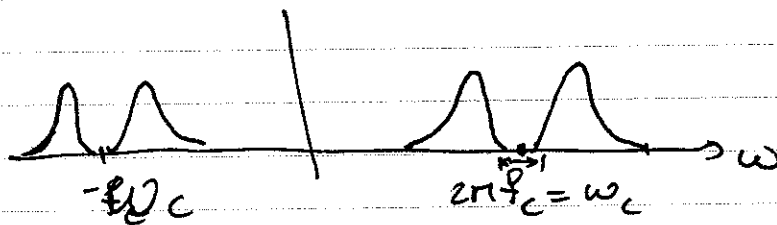


- difficulty of realizing sharp filters

⇒ necessity of having guardband between sidebands ⇒ need to have no components around zero

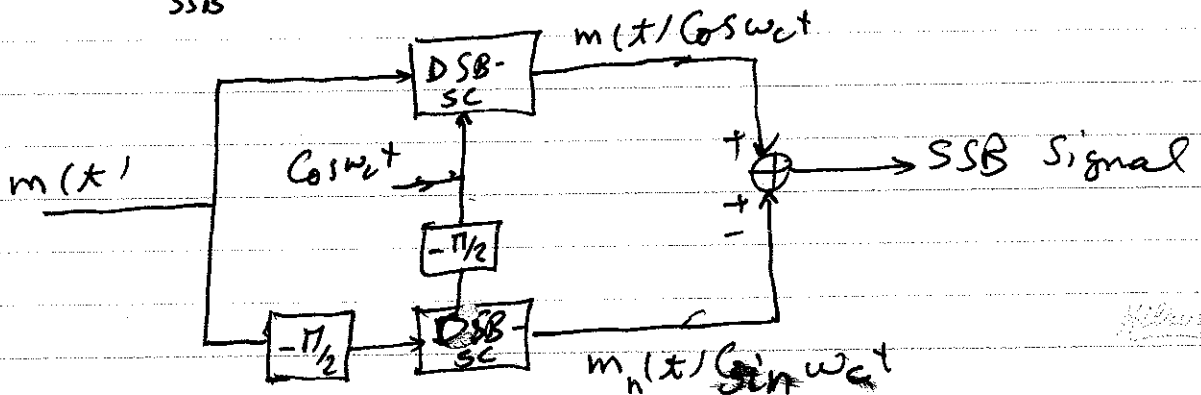
example of voice 0 to 3000 Hz. unimpaired

⇒ so, we have 0 Hz. Guardband.



2) Phase-shift method

$$\phi_{SSB}(t) = m(t) \cos \omega_c t \mp m_n(t) \sin \omega_c t$$



4)

Demodulation of SSB

$$\Phi_{SSB}(t) = m(t) \cos \omega_c t \pm m_n(t) \sin \omega_c t$$

$$\Phi_{SSB}(t) \cos \omega_c t = \frac{1}{2} m(t) [1 + \cos 2\omega_c t] \pm \frac{1}{2} m_n(t) \sin 2\omega_c t$$

after LPF we get $\frac{1}{2} m(t)$

Envelope Detection of SSB

$$\Phi_{SSB+c}(t) = A \cos \omega_c t + [m(t) \cos \omega_c t \mp m_n(t) \sin \omega_c t]$$

$$= [A + m(t)] \cos \omega_c t + m_n(t) \sin \omega_c t$$

$$= E(t) \cos(\omega_c t + \theta)$$

$$E(t) = \left\{ (A + m(t))^2 + m_n^2(t) \right\}^{1/2}$$

$$= A \left[1 + \frac{2m(t)}{A} + \frac{m^2(t)}{A^2} + \frac{m_n^2(t)}{A^2} \right]$$

if $A \gg |m(t)|$ then $\frac{m^2(t)}{A^2}$ and $\frac{m_n^2(t)}{A^2}$

can be ignored and

$$E(t) \approx A \left[1 + \frac{m(t)}{A} \right] = A + m(t)$$

if $\frac{m(t)}{A} \ll 1$ since

Carrier Acquisition

$$\phi(t) = m(t) \cos \omega_c t$$

DSB-SC

DSB-SC

$$e(t) = m(t) \cos \omega_c t + \cos [(\omega_c + \Delta\omega)t + \delta]$$

$$= m(t) \{ \cos [(\Delta\omega)t + \delta] + \cos [(2\omega_c + \Delta\omega)t + \delta] \}$$

if $\Delta\omega = 0$ and $\delta = 0$

i.e., no frequency or phase offset,

then the output of the LPF is:

$$e_o(t) = m(t)$$

otherwise

$$e_o(t) = m(t) \cos [(\Delta\omega)t + \delta]$$

if $\Delta\omega = 0$

$$\text{then } e_o(t) = m(t) \cos \delta$$

i.e., a constant attenuation. $\int e_o(t)$ is max. when $\delta = 0$ and 0 when $\delta = \frac{\pi}{2}$.when $\delta = 0$ & $\Delta\omega \neq 0$

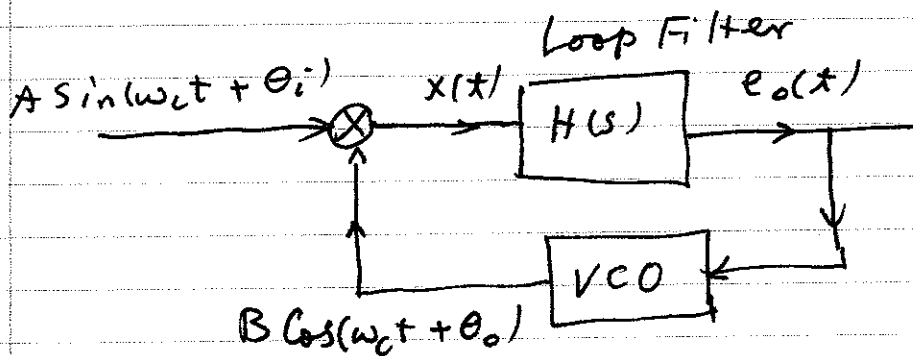
$$e_o(t) = m(t) \cos(\Delta\omega)t$$

i.e., the output is distorted by being

distorted

modulated with a low frequency sinusoid.

Phase Locked Loop (PLL)



VCO : Voltage Controlled Oscillator

has the frequency

$$\omega(t) = \omega_c + c e_o(t)$$

where ω_c is the free running frequency.

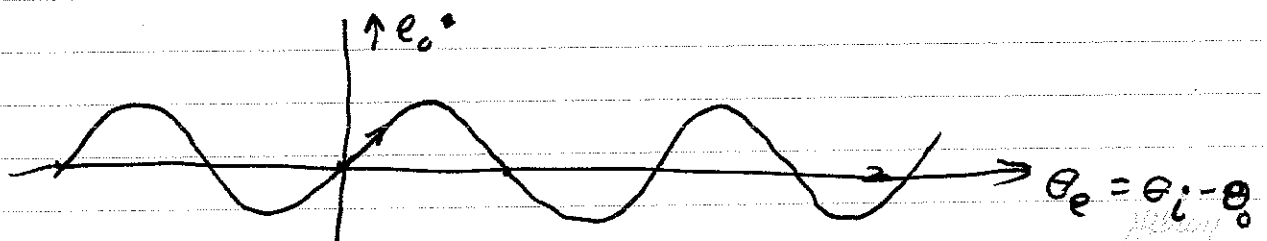
$$x(t) = A B \cos(\omega_c t + \theta_i) \cos(\omega_c t + \theta_o)$$

$$= \frac{AB}{2} [\sin(\theta_i - \theta_o) + \sin(2\omega_c t + \theta_i + \theta_o)]$$

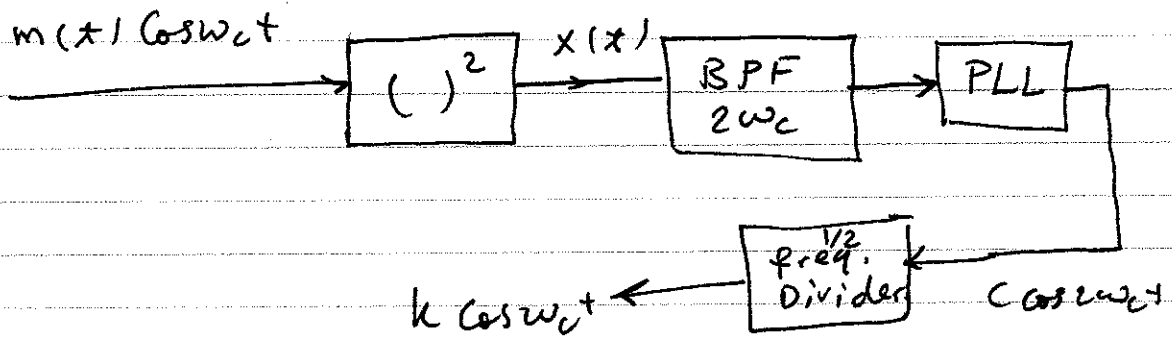
$$e_o(t) = \frac{AB}{2} \sin(\theta_i - \theta_o) \quad \text{or } \sin$$

or

$$e_o(t) = \frac{AB}{2} \sin \theta_e \quad \text{where } \theta_e = \theta_i - \theta_o$$



Carrier Acquisition in DSB-SC



$$x(t) = [m(t) \cos \omega_c t]^2 = \frac{1}{2} m^2(t) + \frac{1}{2} m^2(t) \cos 2\omega_c t$$

let $\frac{1}{2} m^2(t) = k + \phi(t)$

where k is the average value of $\frac{1}{2} m^2(t)$

then:

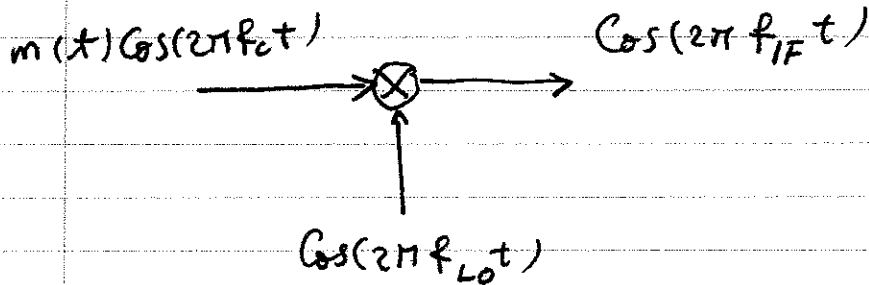
$$x(t) = \frac{1}{2} m^2(t) + \underbrace{k \cos 2\omega_c t + \phi(t) \cos 2\omega_c t}$$

A narrow BPF filters out $\frac{1}{2} m^2(t)$ and

most of $\phi(t) \cos 2\omega_c t$ and gives $k \cos 2\omega_c t$

Superheterodyne AM receiver

$f_{IF} = 455 \text{ kHz.}$

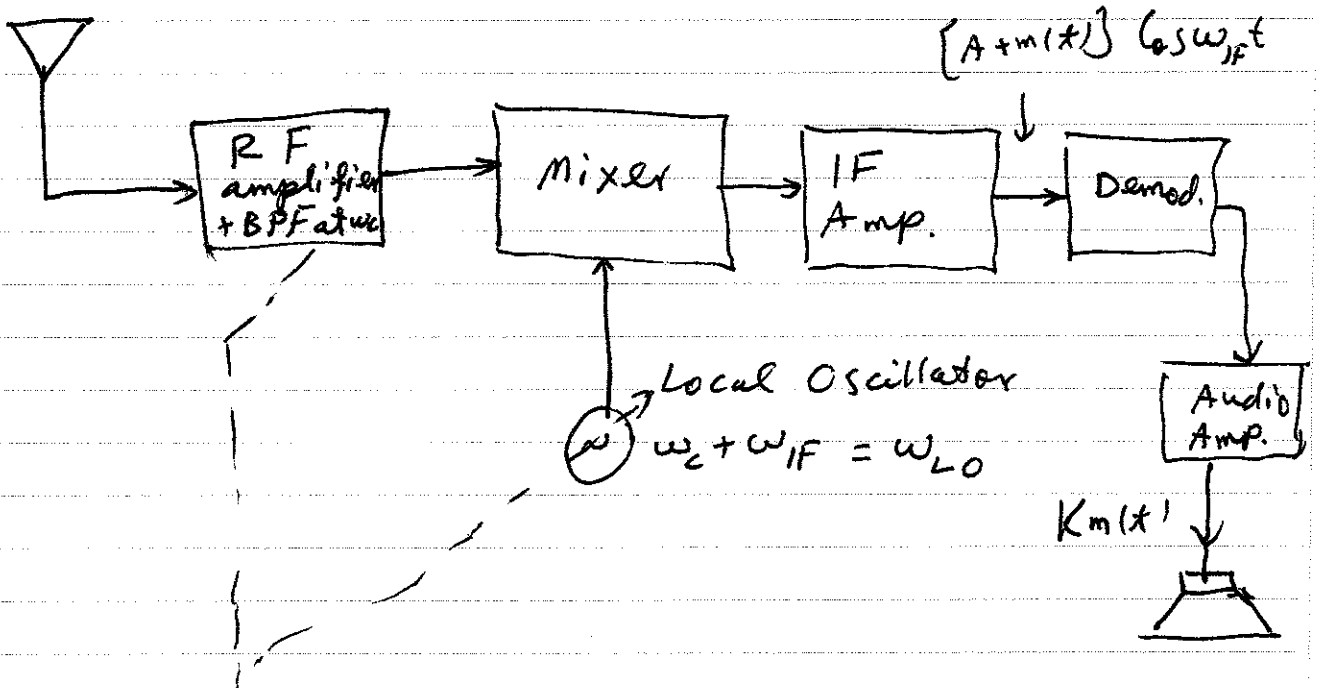


$f_{IF} = f_{LO} - f_c$ or $f_{LO} = f_c + f_{IF}$

The reason for super heterodyning :

Difficulty of designing 10KHz filters

(also AMPs) at all high frequencies.



The broadcast BW for AM is

from 550 to 1600 kHz. \Rightarrow 105

So you can have f_{LO} ~~from~~ stations

from 95 kHz to 2055 kHz

or

from

1005 kHz to 2055 kHz ✓

Image station:

Assume we like to receive a station at

$$f_c = 1000 \text{ kHz.}$$

we use

$$f_{LO} = 1455 \text{ kHz}$$

since

$$f_{LO} - f_c = 455$$

but at the same time we get also

~~$$f_{LO} = 1455 \text{ kHz}$$~~

$$f_{c'} = 1910 \text{ kHz}$$

since

$$f_{c'} - f_{LO} = 1910 - 1455 = 455 \text{ kHz.}$$

Image rejection is done by RF filter. *Illness*

x Lecture 5:

Angle Modulation

$$\varphi(t) = A \cos(\omega_c t + \theta)$$

or in general

$$\varphi(t) = A \cos \theta(t)$$

where $\theta(t)$ is total phase or the generalized angle.

Instantaneous frequency is defined as

$$\omega_i(t) = \frac{d\theta(t)}{dt} \quad \text{or} \quad f_i(t) = \frac{1}{2\pi} \frac{d\theta(t)}{dt}$$

or

$$\theta(t) = \int_{-\infty}^t \omega_i(\alpha) d\alpha$$

For AM we have:

$$\varphi(t) = A_m(t) \cos[\omega_c t]$$

So $\theta(t) = \omega_c t$ and $\omega_i(t) = \omega_c = \text{constant}$

For Phase Modulation (PM), we have

$$\theta(t) = \omega_c t + k_p m(t)$$

that is, phase changes with (in proportion to) the message, $m(t)$.

That is

$$\varphi_{PM}(t) = A \cos[\omega_c t + k_p m(t)]$$

Then, for PM, the frequency is:

$$\omega_i(t) = \frac{d\theta(t)}{dt} = \omega_c + k_p \dot{m}(t)$$

So, the frequency varies with the derivative of the message.

For frequency modulation, FM, we have

$$\omega_i(t) = \omega_c + k_f m(t)$$

that is, here, frequency varies (is modulated) by the message; The phase is:

$$\begin{aligned} \theta(t) &= \int_{-\infty}^t [\omega_c + k_f m(\alpha)] d\alpha \\ &= \omega_c t + k_f \int_{-\infty}^t m(\alpha) d\alpha \end{aligned}$$

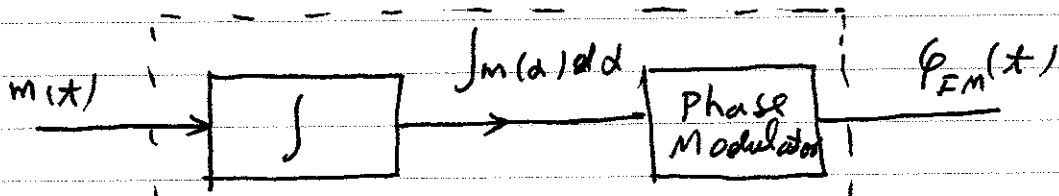
and, therefore

$$\varphi_{FM}(t) = A \cos\left[\omega_c t + k_f \int_{-\infty}^t m(\alpha) d\alpha\right]$$

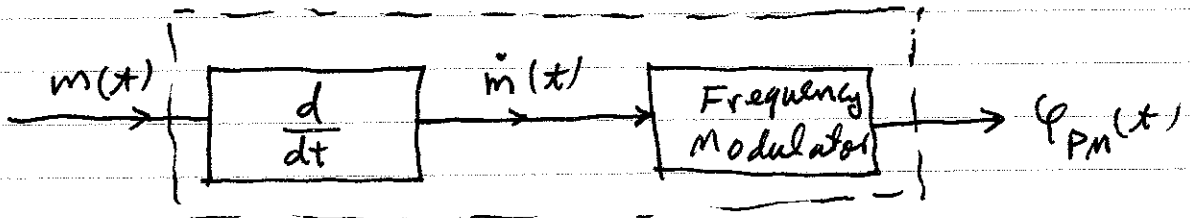
Note that FM and PM are quite similar:

FM is PM with the modulating signal being the integral of the message and PM is FM with the modulating signal being derivative

of the message.



generating FM using PM modulator

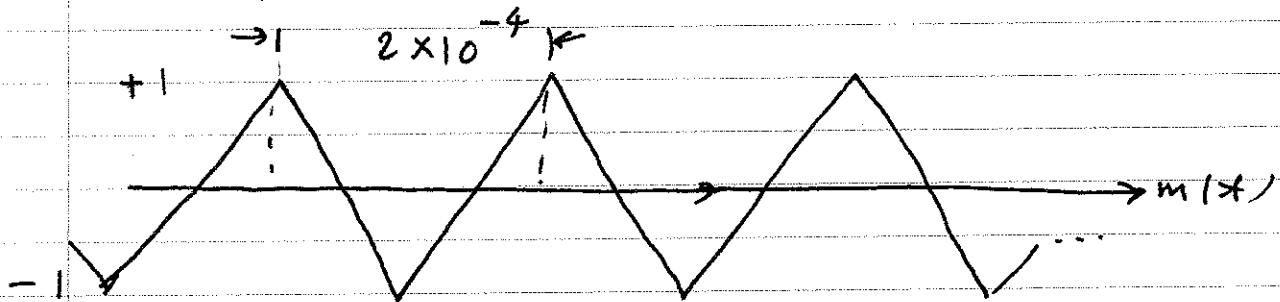


generating PM signal using FM modulator

Example:

Sketch FM and PM wave for the modulating signal $m(t)$ shown below. ~~$k_f = 2\pi \times 10^5$~~ $k_f = 2\pi \times 10^5$

and $k_p = 10\pi$ and $f_c = 100 \text{ MHz}$



$$f_0 = \frac{1}{2 \times 10^{-4}} = 5 \times 10^3 \text{ or } \omega_0 = 10\pi \times 10^3$$

For FM:

$$\omega_i = \omega_c + k_f m(t)$$

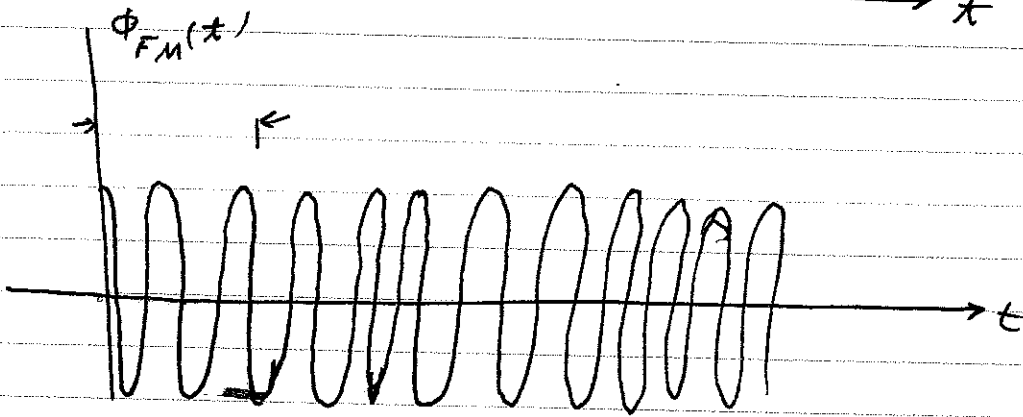
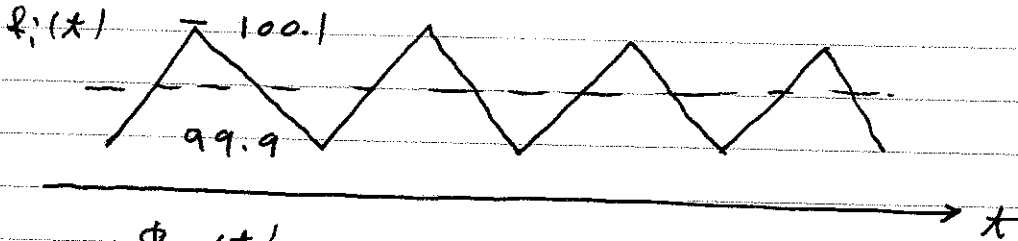
or

$$f_i = f_c + \frac{k_f}{2\pi} m(t)$$

$$f_i = 10^8 + 10^5 m(t)$$

$$f_{i, \min} = 10^8 - 10^5 = 99.9 \text{ MHz}$$

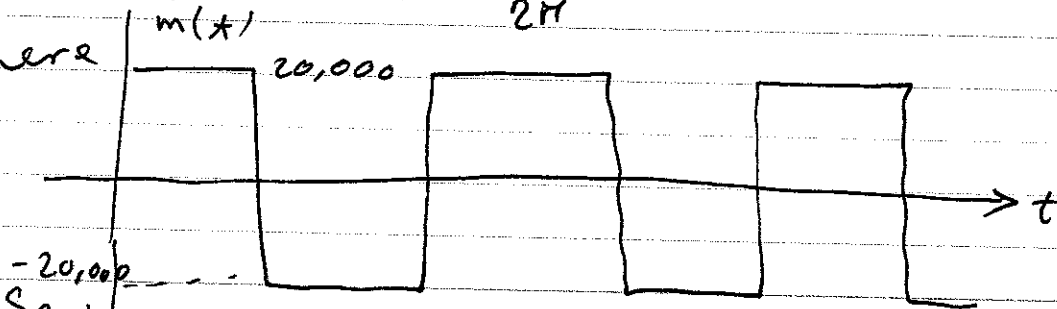
$$f_{i, \max} = 10^8 + 10^5 = 100.1 \text{ MHz}$$



for PM

$$f_i(t) = f_c + \frac{k_p}{2\pi} \dot{m}(t)$$

where

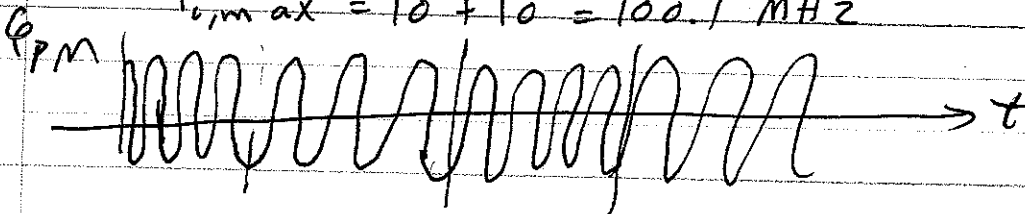


So:

$$f_i(t) = 10^5 + 5 \dot{m}(t)$$

$$f_{i, \min} = 10^8 - 10^5 = 99.9 \text{ MHz}$$

$$f_{i, \max} = 10^8 + 10^5 = 100.1 \text{ MHz}$$



Bandwidth requirement of FM signals:

$$\varphi_{FM}(t) = A \cos \left[\omega_c t + k_f \int_{-\infty}^t m(\alpha) d\alpha \right]$$

$$\text{let } \int_{-\infty}^t m(\alpha) d\alpha = a(t)$$

then

$$\varphi_{FM}(t) = A \cos [\omega_c t + k_f a(t)]$$

to make presentation easier let:

$$\varphi_{FM}(t) = \text{Re} A e^{j[\omega_c t + k_f a(t)]}$$

$$A e^{j[\omega_c t + k_f a(t)]} = A e^{j\omega_c t} \left[1 + j k_f a(t) - \frac{k_f^2}{2!} a^2(t) + \dots \right]$$

So

$$\varphi_{FM}(t) = A \left[\cos \omega_c t - k_f a(t) \sin \omega_c t + \frac{k_f^2}{2!} a^2(t) \cos \omega_c t + \frac{k_f^3}{3!} a^3(t) \sin \omega_c t + \dots \right]$$

So, the spectrum of FM signal has the carrier
plus the spectra of $a(t)$, $a^2(t)$, $a^3(t)$, ...
centered
around ω_c .

Narrowband FM (NBFM)

$$\phi_{FM}(t) = A \cos[\omega_c t + k_f a(t)] = A \cos \omega_c t \cos(k_f a(t)) - A \sin \omega_c t \sin(k_f a(t))$$

if $|k_f a(t)| \ll 1$

then

$$\sin(k_f a(t)) \approx k_f a(t)$$

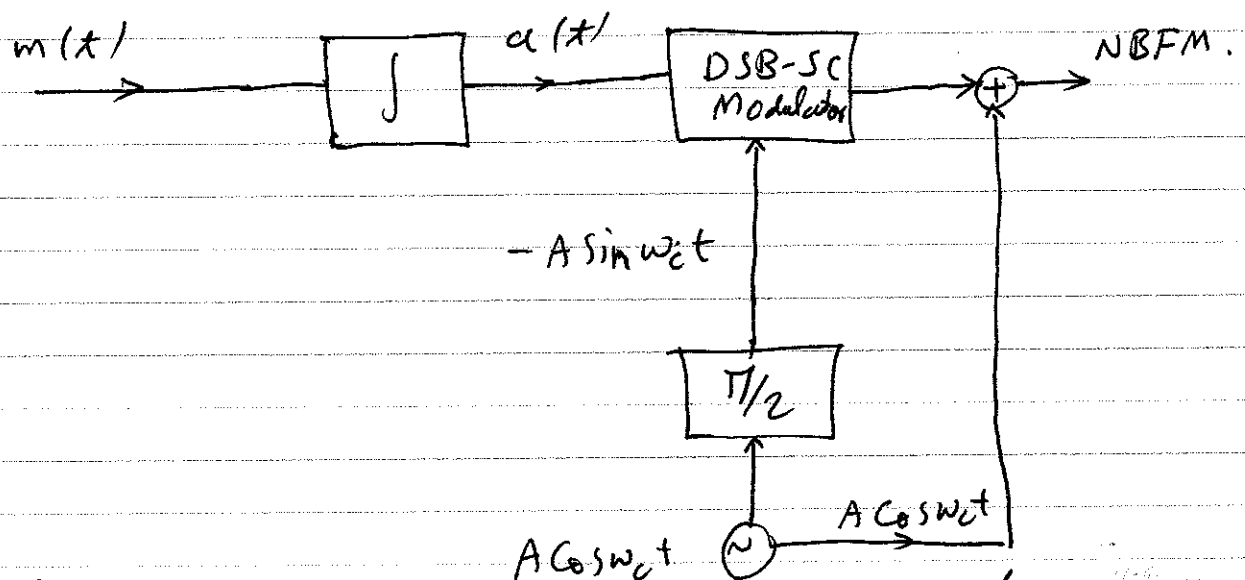
and

$$\cos(k_f a(t)) \approx 1$$

and

$$\phi_{FM}(t) \approx A [\cos \omega_c t - k_f a(t) \sin \omega_c t]$$

and modulation looks like (but, is not!) AM modulation.



Here :

$$BW_{NBFM} = 2B$$

where B is the BW of $m(t)$

For FM, in general, i.e., for wide Band FM it is difficult to find the effective required Bandwidth. So, we find it for a tone being modulated, i.e., when

$$m(t) = \alpha \cos \omega_m t$$

then

$$\alpha(t) = \frac{\alpha}{\omega_m} \sin \omega_m t$$

then

$$\begin{aligned} \varphi_{FM}(t) &= A \cos \left[\omega_c t + \frac{\alpha k_f}{\omega_m} \sin \omega_m t \right] \\ &= \text{Re } A e^{j\omega_c t} \cdot e^{j \frac{\alpha k_f}{\omega_m} \sin \omega_m t} \end{aligned}$$

we can denote

$$\beta = \frac{\alpha k_f}{\omega_m}$$

this is called the modulation index.

Note that αk_f is the peak frequency deviation of FM signal, since

$$\omega_i(t) = \omega_c t + k_f \alpha \cos \omega_m t$$

So

$$\beta = \frac{\alpha k_f}{\omega_m} = \frac{\Delta \omega}{\omega_m} = \frac{\Delta f}{f_m}$$

Now

$$\varphi_{F_m}(x) = \operatorname{Re} A e^{j\omega_c t} \left[e^{j\beta \sin \omega_m t} \right]$$

$e^{j\beta \sin \omega_m t}$ is periodic with period $\frac{2\pi}{\omega_m}$

Since letting $t \rightarrow t + \frac{2\pi}{\omega_m}$, we get

$$e^{j\beta \sin \left(t + \frac{2\pi}{\omega_m} \right) \omega_m} = e^{j\beta \sin \omega_m t}$$

So, $e^{j\beta \sin \omega_m t}$ can be represented as

a Fourier Series:

$$e^{j\beta \sin \omega_m t} = \sum_{n=-\infty}^{\infty} C_n e^{jn\omega_m t}$$

where

$$C_n = \frac{\omega_m}{2\pi} \int_{-\pi/\omega_m}^{\pi/\omega_m} e^{j\beta \sin \omega_m t} \cdot e^{-jn\omega_m t} dt$$

let $x = \omega_m t$ to get

$$C_n = \frac{1}{2\pi} \int_{-\pi}^{\pi} e^{j(\beta \sin x - nx)} dx$$

The integral is called the Bessel Function of first kind and n th order and is denoted as $J_n(\beta)$.

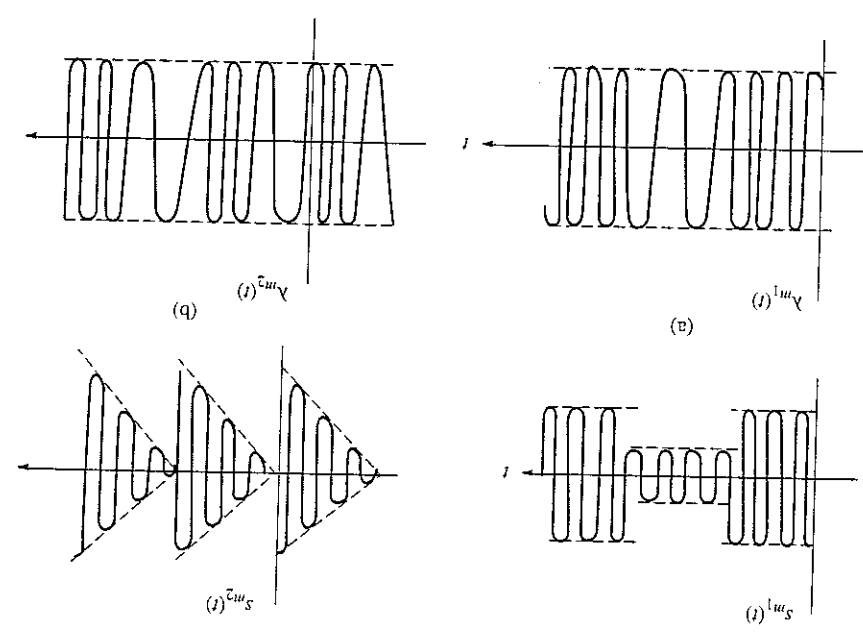
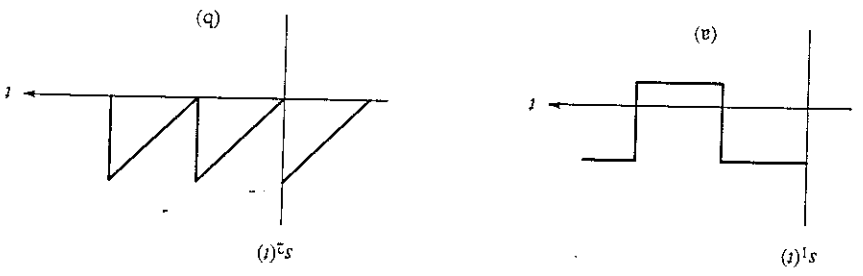
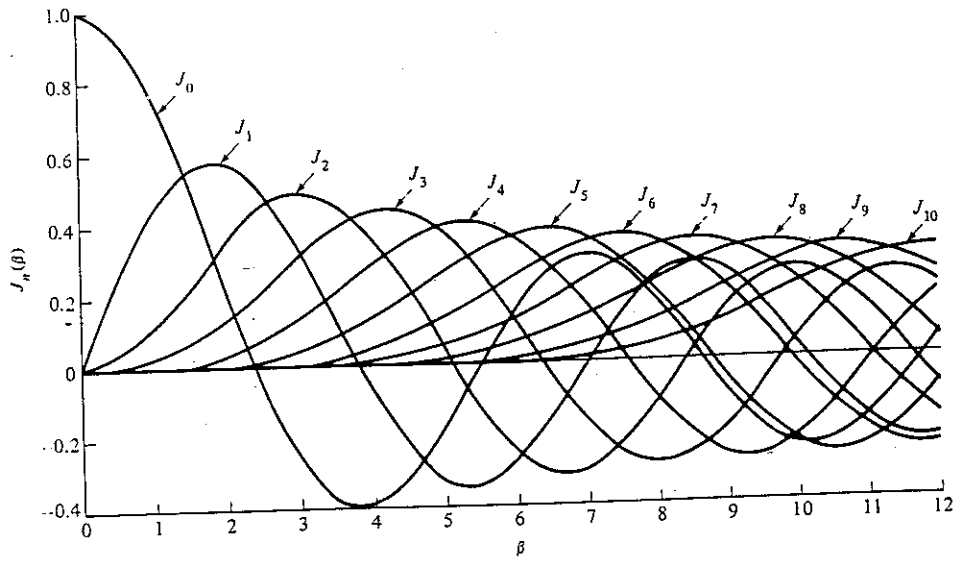


Figure 7.4 Waveforms for Example 7.3.





Plot of Bessel function of the first kind, $J_n(\beta)$.

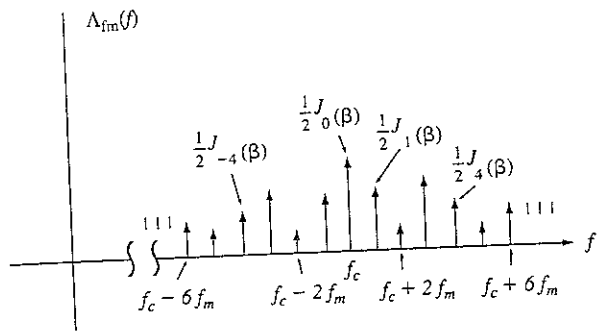
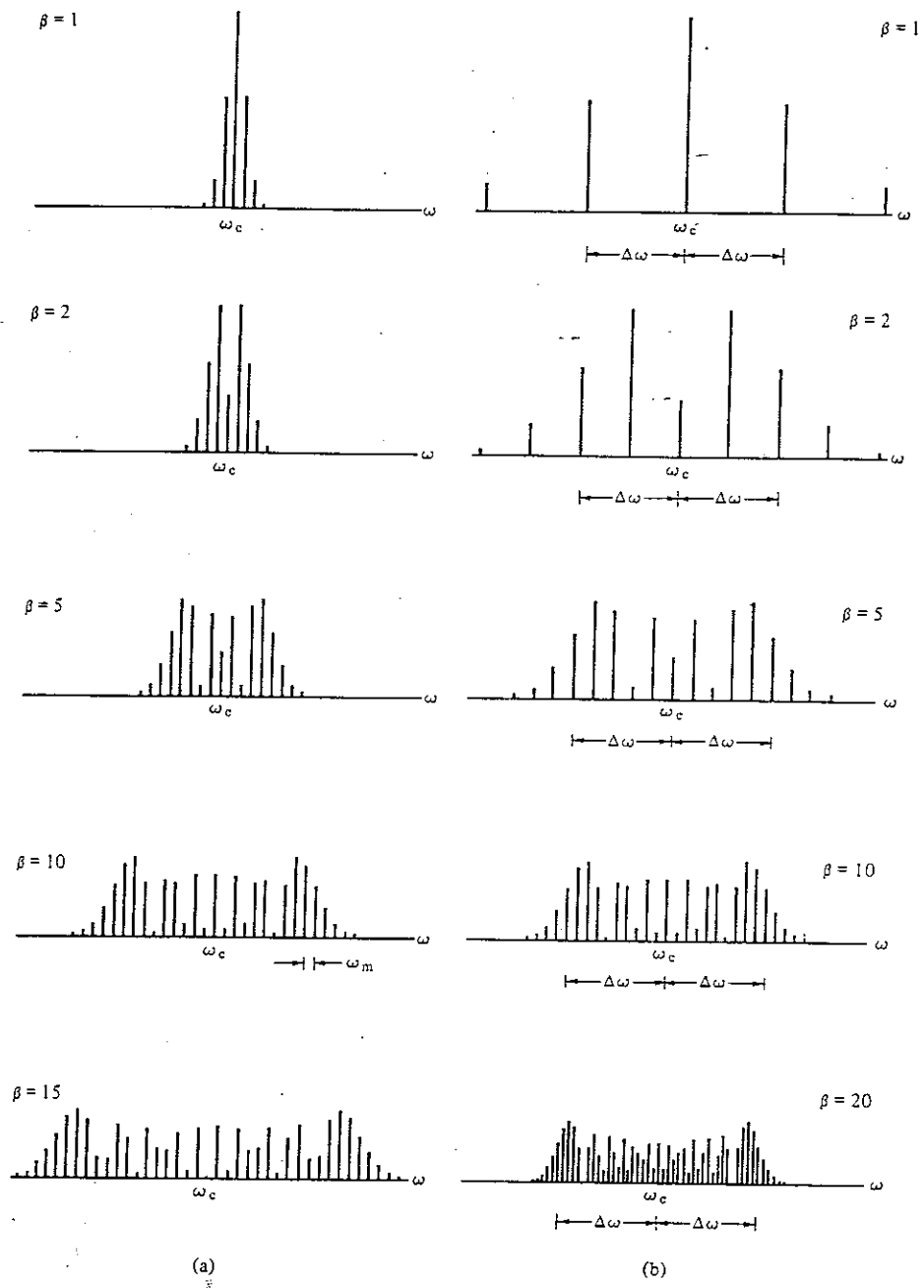


Figure 7.11 Fourier transform of FM with sinusoidal information signal.



Magnitude line spectra for FM waveforms with sinusoidal modulation: (a) for constant ω_m ; (b) for constant $\Delta\omega$.

So:

$$e^{j\beta \sin \omega_m t} = \sum_{n=-\infty}^{\infty} J_n(\beta) e^{jn\omega_m t}$$

So, we get

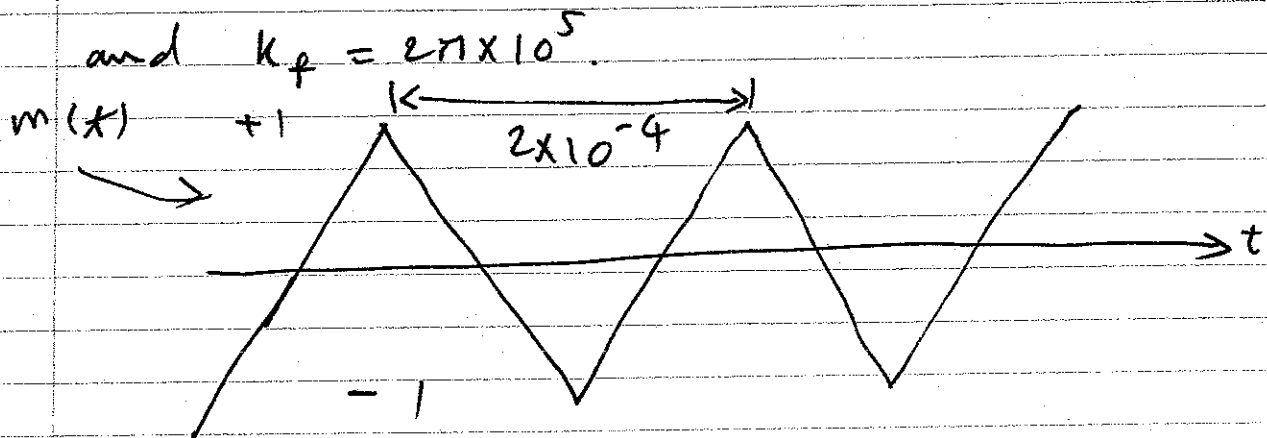
$$\varphi_{FM}(t) = A \sum_{n=-\infty}^{\infty} J_n(\beta) \cos(\omega_c + n\omega_m)t$$

From plot of $J_n(\beta)$ or from its tabulated values, one can observe that $J_n(\beta)$ is negligible for $n > \beta + 1$

So:

$$\begin{aligned} B_{FM} &= 2n f_m = 2(\beta + 1) f_m = 2(\beta + 1) B \\ &= 2(\Delta f + B) \end{aligned} \quad \begin{array}{l} \uparrow \\ \text{in general} \end{array}$$

Example: a) Find the approximate bandwidth of an FM signal with $m(t)$ given below



b) repeat if $m(t)$ is amplified by 2.

a)

$$m(t) = \sum_n C_n \cos n\omega_0 t \quad \omega_0 = \frac{2\pi}{2 \times 10^{-4}} = 10^4 \pi$$

$$C_n = \begin{cases} \frac{8}{\pi^2 n^2} & n \text{ odd} \\ 0 & n \text{ even} \end{cases} \quad \begin{aligned} C_1 &= 0.81 \\ C_3 &= \frac{0.81}{9} \\ C_5 &= \frac{0.81}{25} \end{aligned}$$

Power in Fundamental frequency is $\frac{64}{2\pi^4} = P_1$

in the third harmonic is $\frac{64}{2\pi^4 \times 9^2} \approx \frac{1}{1.21} P_1$

and in fifth harmonic $\frac{64}{2\pi^4 \times 25^2} = \frac{64}{2\pi^4 \times 625} \approx 0.16\%$ of P_1

So, we can ignore anything beyond third harmonic and assume $B = 15 \text{ kHz}$.

$$\Delta f = \frac{k_f m_p}{2\pi} = \frac{1}{2\pi} (2\pi \times 10^5)(1) = 100 \text{ kHz}$$

So

$$B_{FM} = 2(Af + B) = 230$$

we could also do

$$\beta = \frac{\Delta f}{B} = \frac{100}{15}$$

$$B_{FM} = 2B(\beta + 1) = 15 \times 2 \left(\frac{100}{15} + 1 \right) = 230 \text{ kHz}.$$

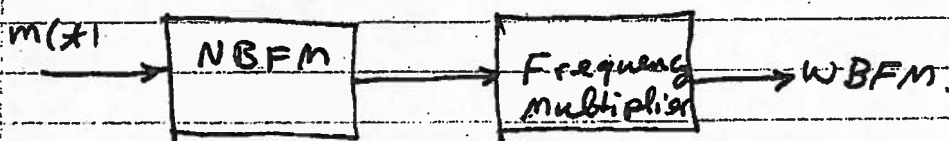
b) $m_p = 2$, $B = 15 \text{ kHz}$

$$\Delta f = \frac{1}{2\pi} \cdot k_f m_p = \frac{1}{2\pi} (2\pi \times 10^5) \times 2 = 200 \text{ kHz}$$

$$B_{FM} = 2(Af + B) = 430 \text{ kHz}.$$

Generation of FM waves:

Indirect FM modulation (Armstrong's method)



Example: Assume that we wish to modulate a signal to a frequency of 91.2 MHz with $\Delta f = 75 \text{ kHz}$. We begin by modulating to a frequency of 200 kHz with a $\Delta f = 25 \text{ Hz}$. (to make $\beta \ll 1$). Assume that $B = 15 \text{ kHz}$

(and range is 50 Hz to 15 kHz)

$$\text{Then } \beta = \frac{\Delta f}{f_m} = \frac{25}{50} = 0.5 \text{ even for the worst case of } f_m = 50 \text{ Hz.}$$

Then, in order to get $\Delta f = 75 \text{ kHz}$, we need to multiply the frequency by $\frac{75000}{25} = 3000$.

This can be done using two frequency multipliers: one by 64 and the second by 48.

This results in a multiplication by

$$64 \times 48 = 3072$$

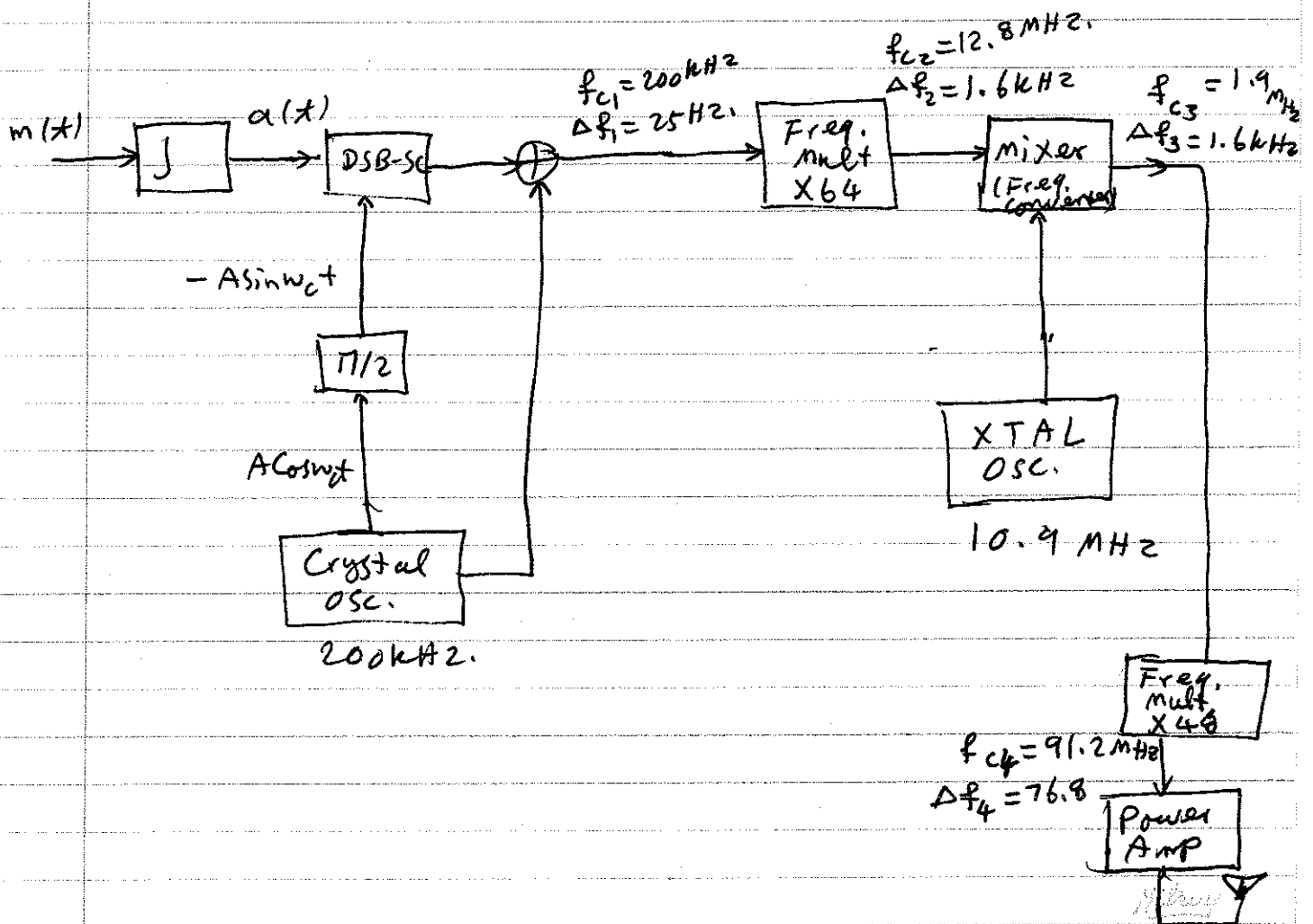
This gives $\Delta f = 76.8 \text{ kHz}$ which is close to 75 kHz .

But the ^{Carrier} frequency will be

$$200 \times 64 \times 48 \approx 600 \text{ MHz.}$$

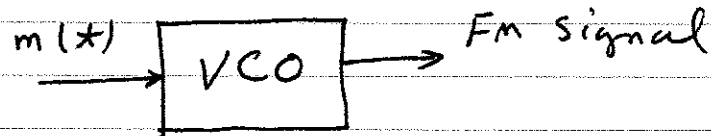
But, we needed $f_c = 91.2 \text{ MHz}$.

So, in order to get this: after the first multiplier, we translate the carrier by 10.9 MHz to 1.9 MHz .



Direct Generation of FM signal

$$\omega_i(t) = \omega_c + k_f m(t)$$



VCO is an LC circuit with

$$C = C_0 - k_m m(t)$$

$$\omega_o = \frac{1}{\sqrt{LC}} = \frac{1}{\sqrt{LC_0 \left[1 - \frac{k_m m(t)}{C_0}\right]}}$$

$$\omega_o = \frac{1}{\sqrt{LC_0}} \cdot \frac{1}{\left[1 - \frac{k_m m(t)}{C_0}\right]^{1/2}}$$

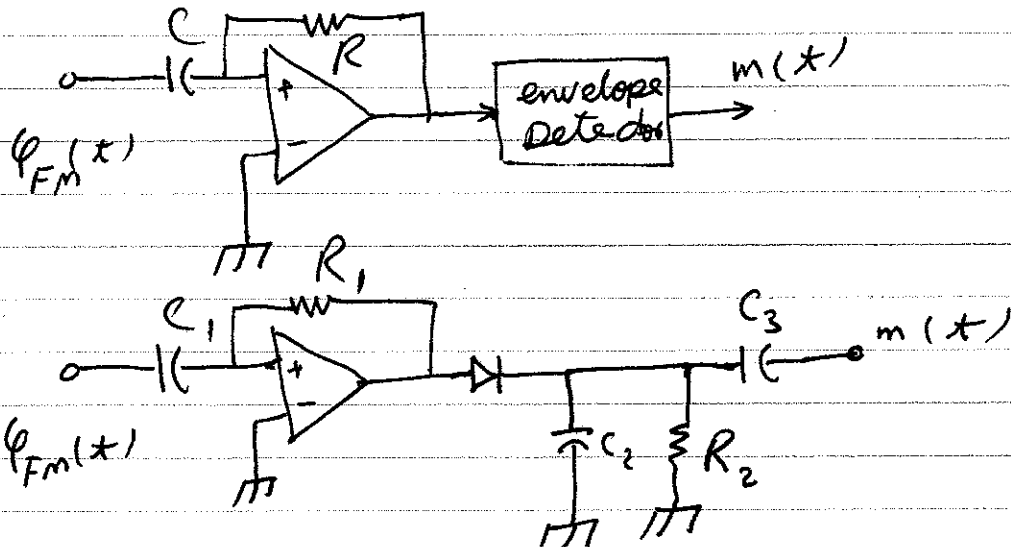
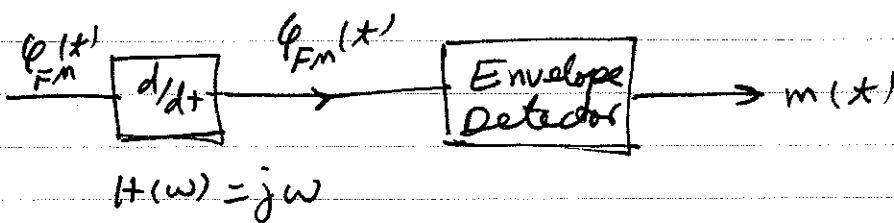
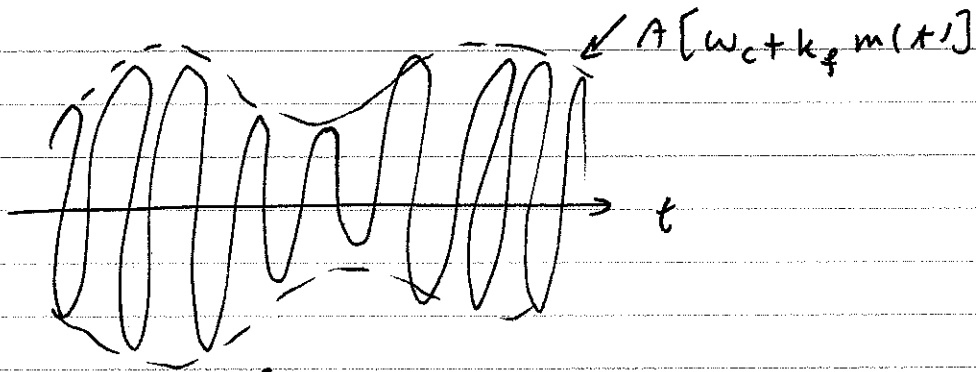
$$\omega_o \approx \frac{1}{\sqrt{LC_0}} \left[1 + \frac{k_m m(t)}{2C_0}\right] \quad \text{if } \frac{k_m m(t)}{C_0} \ll 1.$$

The reason for use of indirect FM is that direct FM lacks frequency stability.

Demodulation of FM

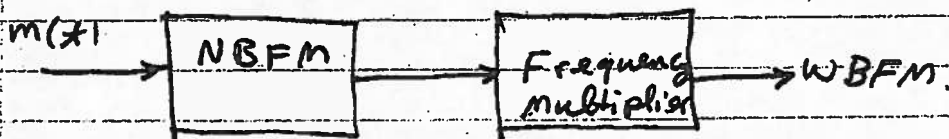
$$\varphi_{FM}(t) = A \cos\left[\omega_c t + k_f \int_{-\infty}^t m(\alpha) d\alpha\right]$$

$$\frac{d}{dt} \varphi_{FM}(t) = -A[\omega_c + k_f m(t)] \sin\left[\omega_c t + k_f \int_{-\infty}^t m(\alpha) d\alpha\right]$$



Generation of FM waves:

Indirect FM modulation (Armstrong's method)



Example: Assume that we wish to modulate a signal to a frequency of 91.2 MHz with $\Delta f = 75 \text{ kHz}$. We begin by modulating to a frequency of 200 kHz with a $\Delta f = 25 \text{ Hz}$. (to make $\beta \ll 1$). Assume that $B = 15 \text{ kHz}$

(and range is 50 Hz to 15 kHz)

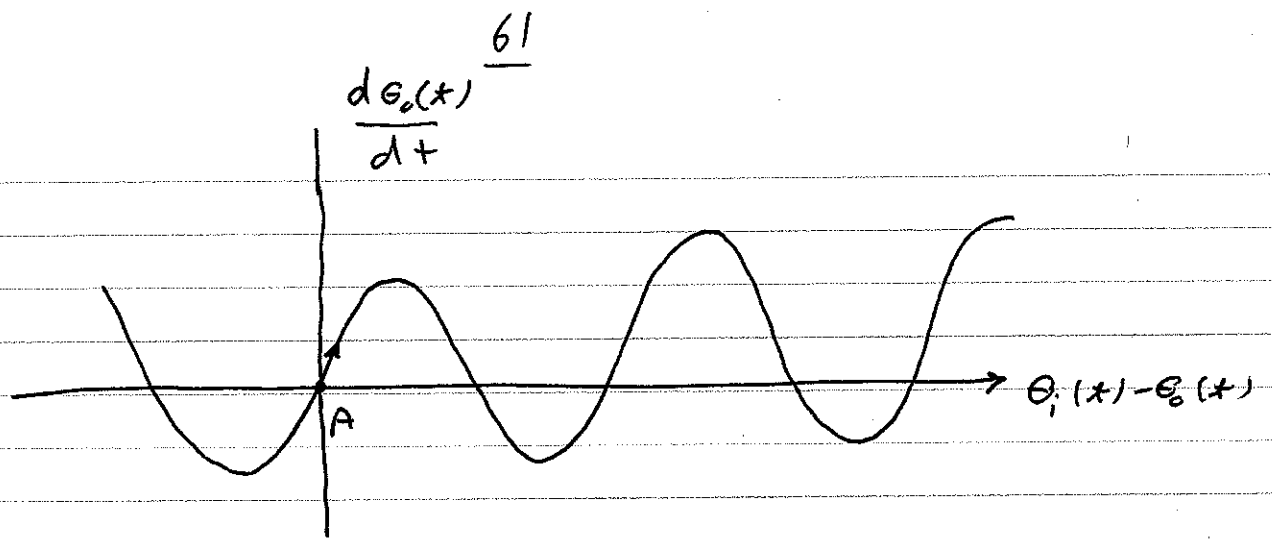
$$\text{Then } \beta = \frac{\Delta f}{f_m} = \frac{25}{50} = 0.5 \text{ even for the worst case of } f_m = 50 \text{ Hz.}$$

Then, in order to get $\Delta f = 75 \text{ kHz}$, we need to multiply the frequency by $\frac{75000}{25} = 3000$.

This can be done using two frequency multipliers: one by 64 and the second by 48.

This results in a multiplication by

$$64 \times 48 = 3072$$



at point A if $\theta_i(t) - \theta_o(t)$ increases (either as a result of an increase in $\theta_i(t)$ or ^a decrease in $\theta_o(t)$, former being usually the case) then

$\frac{d\theta_o(t)}{dt}$ ~~increases~~ becomes (more) positive, i.e., increasing $\theta_o(t)$, and as a result decreasing $\theta_i(t) - \theta_o(t)$. The net result would be to push the operating point back to A.

If $\theta_i(t) - \theta_o(t)$ becomes negative (as a result of a decrease in $\theta_i(t)$) then $\frac{d\theta_o(t)}{dt}$ becomes negative and $\theta_o(t)$ decreases and the operating point is pushed up to A.

So, in the steady-state, we have

$$\theta_i(t) - \theta_o(t) \approx 0$$

or

$$\theta_o(t) \approx \theta_i(t)$$

But

$$\theta_i(t) = k_f \int_{-\infty}^t m(\alpha) d\alpha$$

since

$$\varphi_{FM}(t) = A \cos \left[\omega_c t + k_f \int_{-\infty}^t m(\alpha) d\alpha \right]$$

therefore:

$$\theta_o(t) = k_f \int_{-\infty}^t m(\alpha) d\alpha$$

and

$$\frac{d}{dt} \theta_o(t) = c e_o(t) = k_f m(t)$$

or

$$e_o(t) = \frac{k_f}{c} m(t)$$

Pre-emphasis - De-emphasis (PDE)

The interference in FM increases as frequency.

In order to protect higher frequencies,

we may want to amplify higher frequency components (more than lower frequency ones)

before transmission (Emphasize them)

and at the receiver attenuate them

(less amplify them). This way, we

suppress ^{the} noise (since noise has not been emphasized while signal has been).

The preemphasis transfer function is

$$H_p(\omega) = K \frac{j\omega + \omega_1}{j\omega + \omega_2} \tag{5.39a}$$

where K , the gain, is set at a value of ω_2/ω_1 . Thus,

$$H_p(\omega) = \left(\frac{\omega_2}{\omega_1}\right) \frac{j\omega + \omega_1}{j\omega + \omega_2} \tag{5.39b}$$

For $\omega \ll \omega_1$,

$$H_p(\omega) \simeq 1 \tag{5.39c}$$

For frequencies $\omega_1 \ll \omega \ll \omega_2$,

$$H_p(\omega) \simeq \frac{j\omega}{\omega_1} \tag{5.39d}$$

Thus, the preemphasizer acts as a differentiator at intermediate frequencies (2.1 to 15 kHz), which effectively makes the scheme PM over these frequencies. This means that FM with PDE

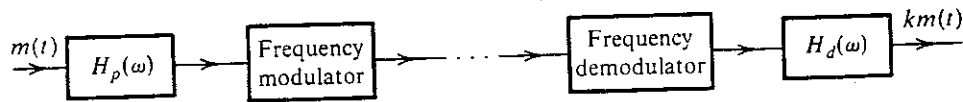


Figure 5.17 Preemphasis-deemphasis in an FM system.

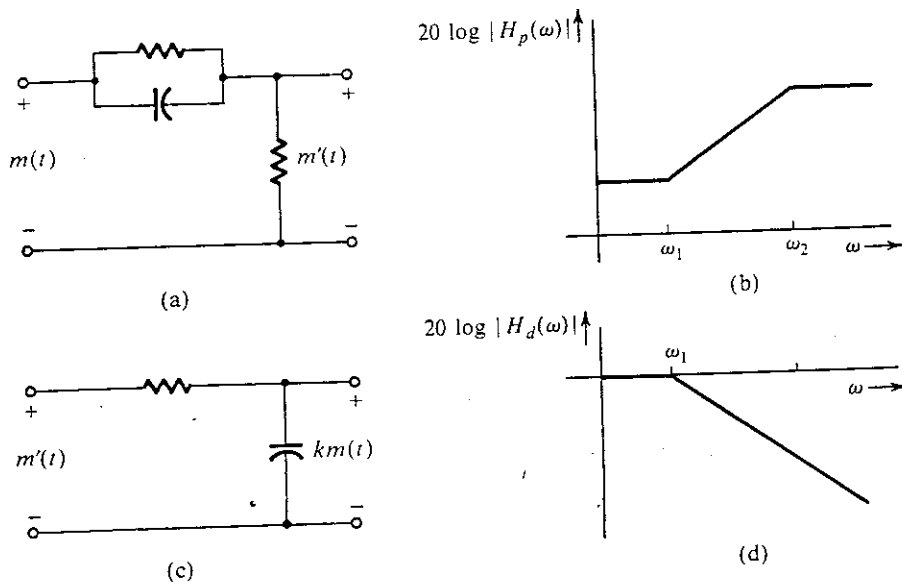


Figure 5.18 (a) Preemphasis filter. (b) Its frequency response. (c) Deemphasis filter. (d) Its frequency response

is FM over the modulating-signal frequency range of 0 to 2.1 kHz and is nearly PM over the range of 2.1 to 15 kHz as desired.

The deemphasis filter $H_d(\omega)$ is given by

$$H_d(\omega) = \frac{\omega_1}{j\omega + \omega_1}$$

Note that for $\omega \ll \omega_2$, $H_p(\omega) \simeq (j\omega + \omega_1)/\omega_1$. Hence, $H_p(\omega)H_d(\omega) \simeq 1$ over the baseband of 0 to 15 kHz.

FM broadcast

The frequency range assigned to FM broadcast is 88 to 108 MHz. with a separation of 200 kHz and a peak frequency deviation of $\Delta f = 75 \text{ kHz}$.

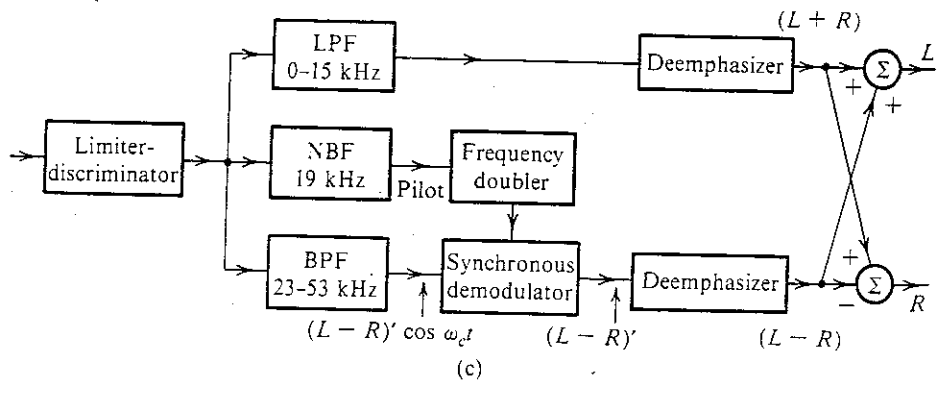
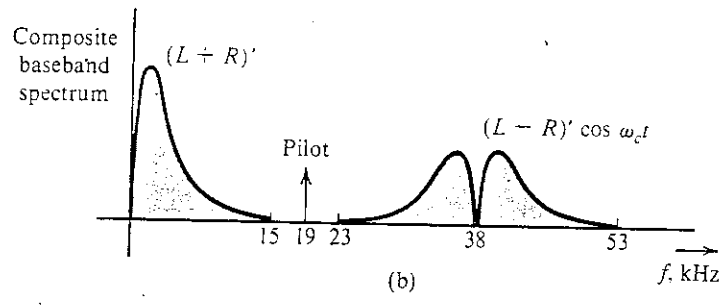
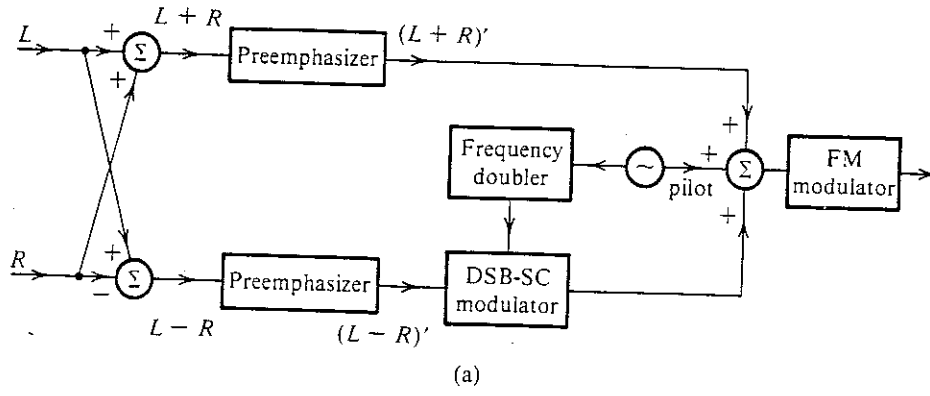


Figure 5.19 (a) FM stereo transmitter. (b) Spectrum of a baseband stereo signal. (c) FM stereo receiver.