

Angle CW Modulation

Phase and Frequency Modulation

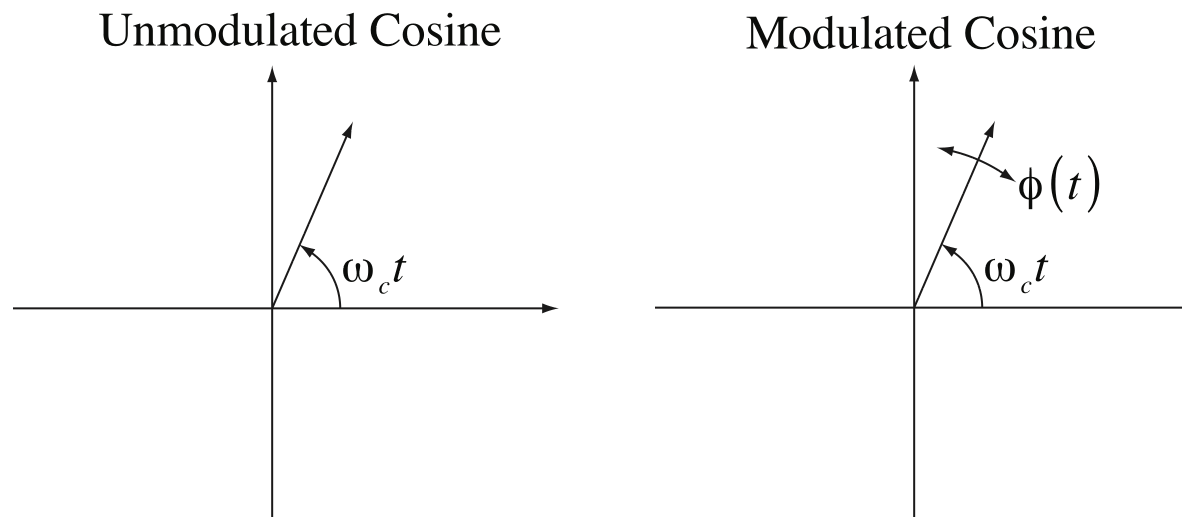
Consider a signal of the form $x_c(t) = A_c \cos(\omega_c t + \phi(t))$ where A_c and ω_c are constants. The envelope is a constant so the message cannot be in the envelope. It must instead lie in the variation of the cosine argument with time. Let $\theta_c(t) \triangleq \omega_c t + \phi(t)$ be the **instantaneous angle**. Then

$$x_c(t) = A_c \cos(\theta_c(t)) = A_c \operatorname{Re}(e^{j\theta_c(t)}).$$

$\theta_c(t)$ contains the message and this type of modulation is called **angle** or **exponential** modulation. If $\phi(t) = \phi_\Delta x(t)$ with $\phi_\Delta \leq 180^\circ$ so that $x_c(t) = A_c \cos(\omega_c t + \phi_\Delta x(t))$ the modulation is called **phase modulation (PM)** where ϕ_Δ is the **phase modulation index**.

Phase and Frequency Modulation

Think about what it means to modulate the phase of a cosine. The total argument of the cosine is $\omega_c t + \phi(t)$, an angle with units of radians (or degrees). When $\phi(t) = 0$, we simply have a cosine and the angle $\omega_c t$ is a linear function of time. Think of this angle as the angle of a phasor rotating at a constant angular velocity. Now add the effect of the phase modulation $\phi(t)$. The modulation adds a "wiggle" to the rotating phasor with respect to its position when it is unmodulated. The message is in the variation of the phasor's angle with respect to the constant angular velocity of the unmodulated cosine.



Phase and Frequency Modulation

The total argument of an unmodulated cosine is $\theta_c(t) = \omega_c t$ in which ω_c is a radian frequency. The time derivative of $\omega_c t$ is ω_c . We could also express the argument in cyclic frequency form as $\theta_c(t) = 2\pi f_c t$. Its time derivative is $2\pi f_c$. Therefore one way of defining the cyclic frequency of an unmodulated cosine is as $\frac{1}{2\pi} \frac{d}{dt}(\theta_c(t))$. Now let's apply this same idea to a modulated cosine whose argument is $\theta_c(t) = 2\pi f_c t + \phi(t)$.

Its time derivative is $2\pi f_c + \frac{d}{dt}(\phi(t))$. Now we define **instantaneous frequency** as

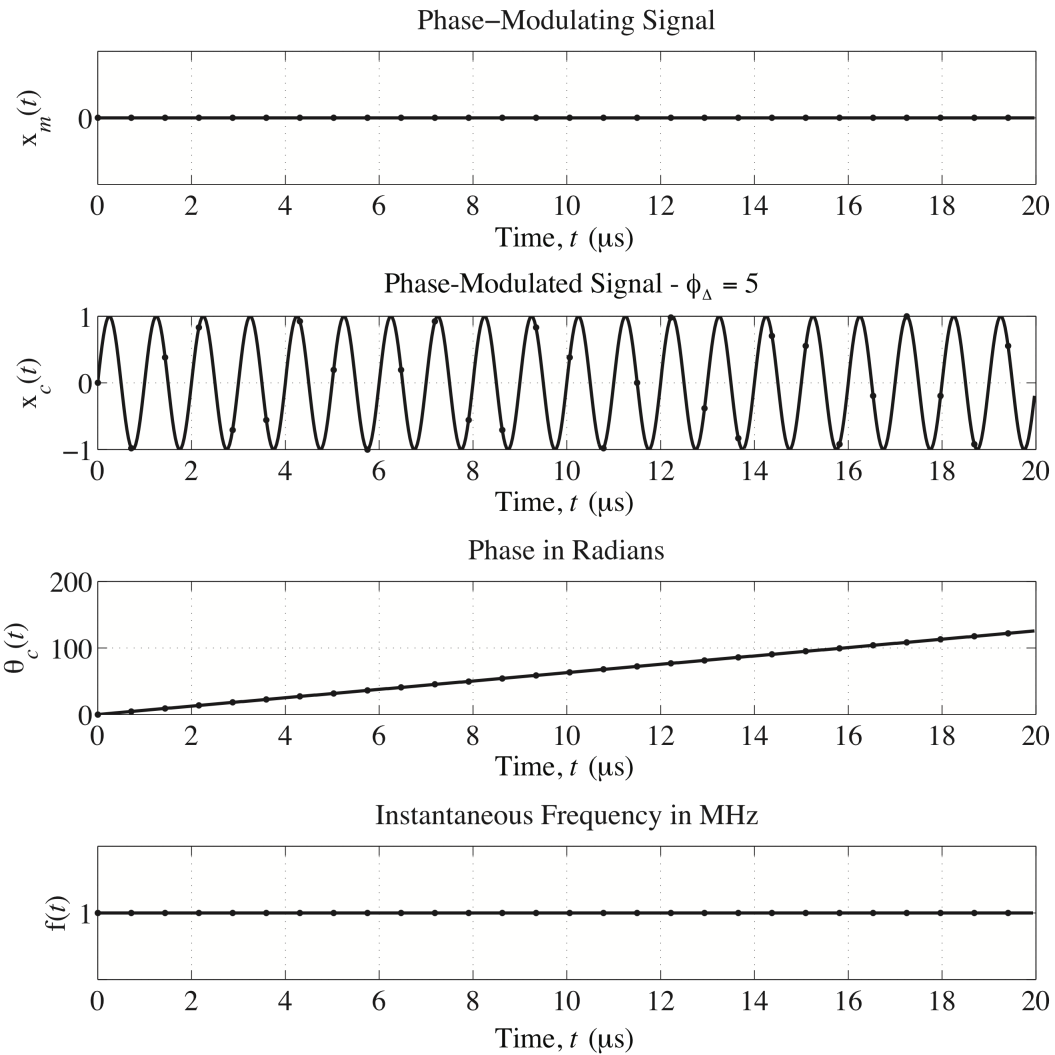
$$f(t) \triangleq \frac{1}{2\pi} \frac{d}{dt}(\theta_c(t)) = \frac{1}{2\pi} \left[2\pi f_c + \frac{d}{dt}(\phi(t)) \right] = f_c + \frac{1}{2\pi} \frac{d}{dt}(\phi(t)).$$

It is important to draw a distinction between instantaneous frequency $f(t)$ and spectral frequency f . They are definitely not the same. Let $x_c(t) = \cos(2\pi f_c t + \phi(t))$. It has a Fourier transform $X_c(f)$.

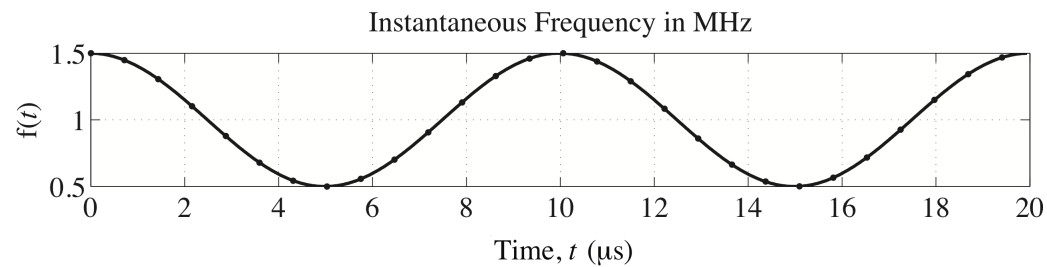
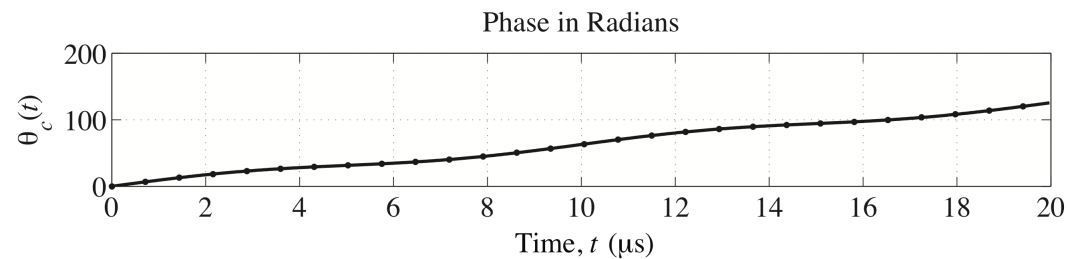
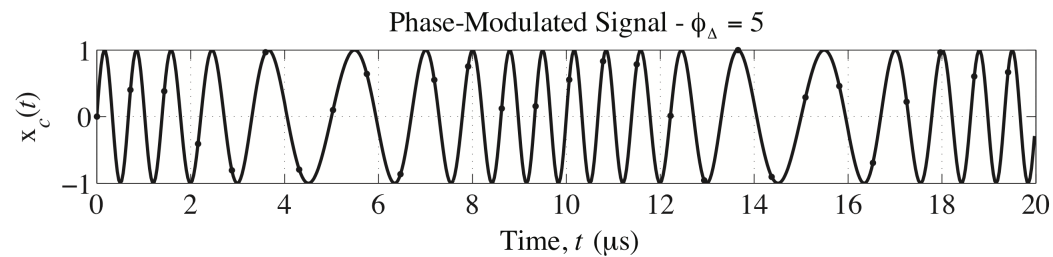
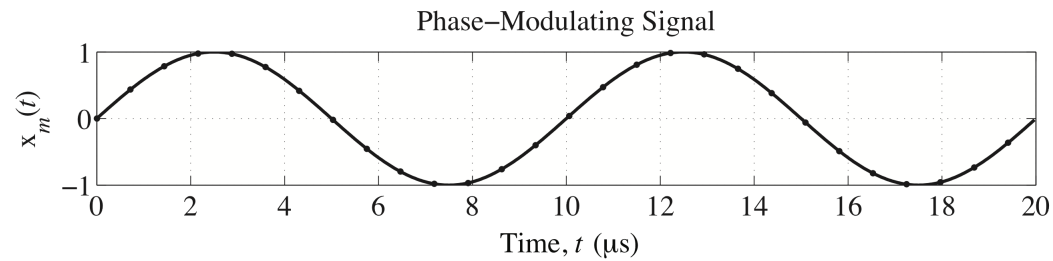
Spectral frequency f is the independent variable in $X_c(f)$ but $f(t) = f_c + \frac{1}{2\pi} \frac{d}{dt}(\phi(t))$.

Some Fourier transforms of phase and frequency modulated signals later will make this distinction clearer.

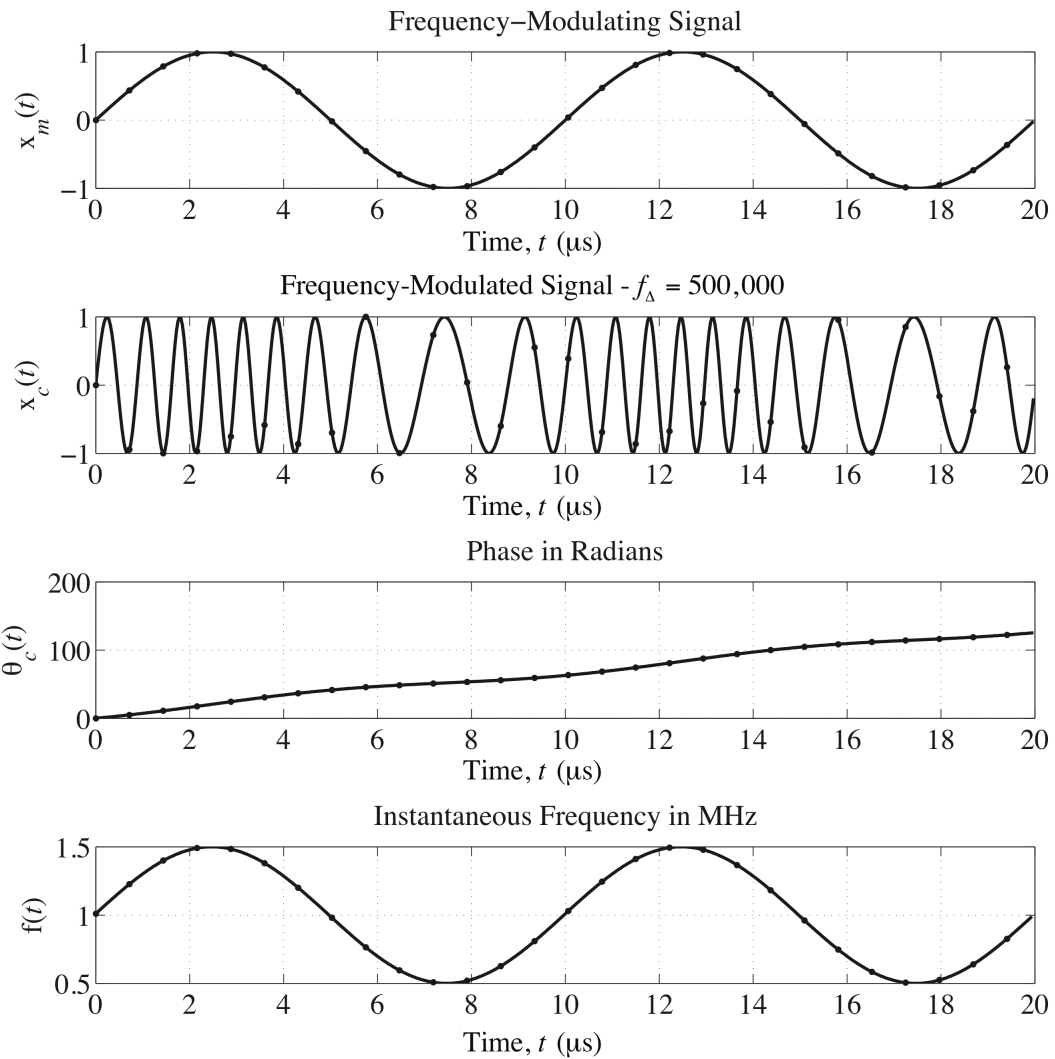
Phase and Frequency Modulation



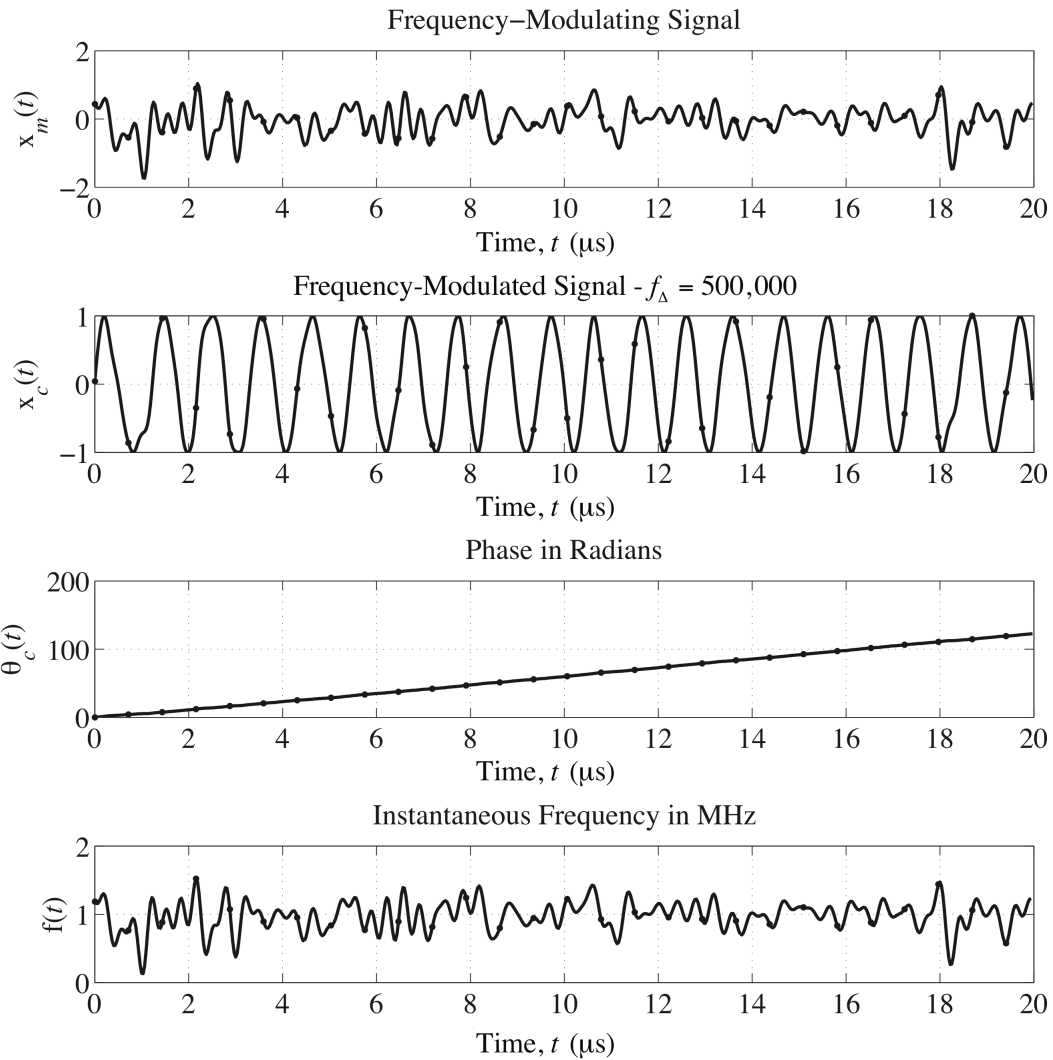
Phase and Frequency Modulation



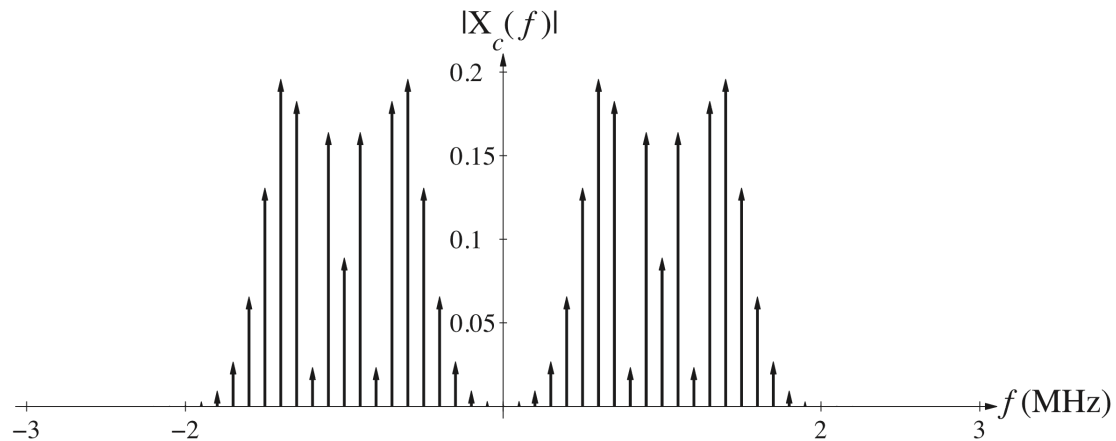
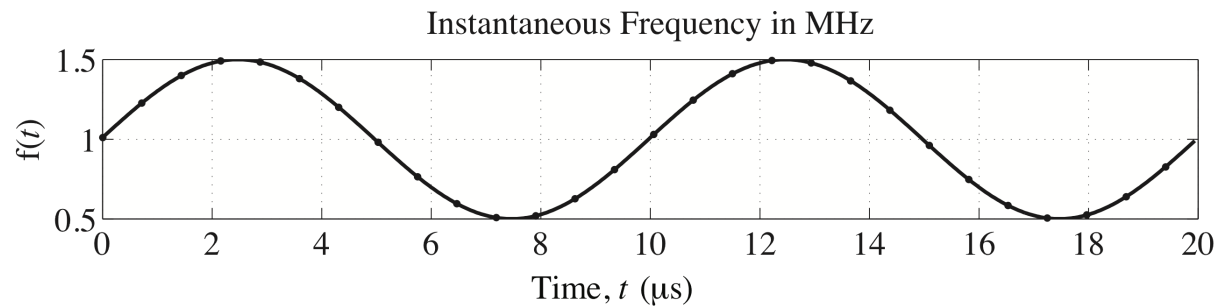
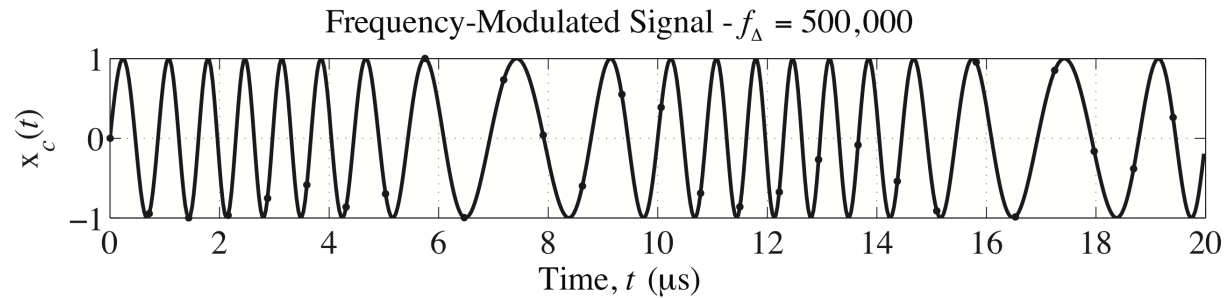
Phase and Frequency Modulation



Phase and Frequency Modulation



Phase and Frequency Modulation



Phase and Frequency Modulation

If we make the variation of the instantaneous frequency of a sinusoid be directly proportional to the message we are doing **frequency modulation (FM)**. In frequency modulation $f(t) = f_c + f_\Delta x(t)$, $f_\Delta < f_c$ where f_Δ is the modulation index for frequency modulation. Typically $f_\Delta \ll f_c$ because we desire to transmit a bandpass signal. In frequency modulation $\frac{d}{dt}(\phi(t)) = 2\pi f_\Delta x(t)$, therefore

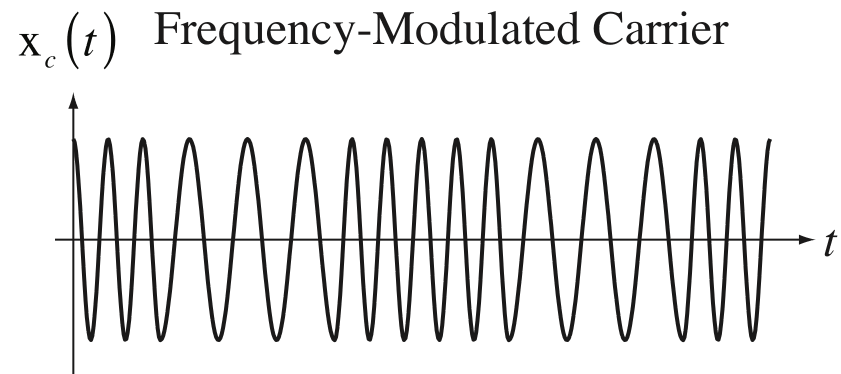
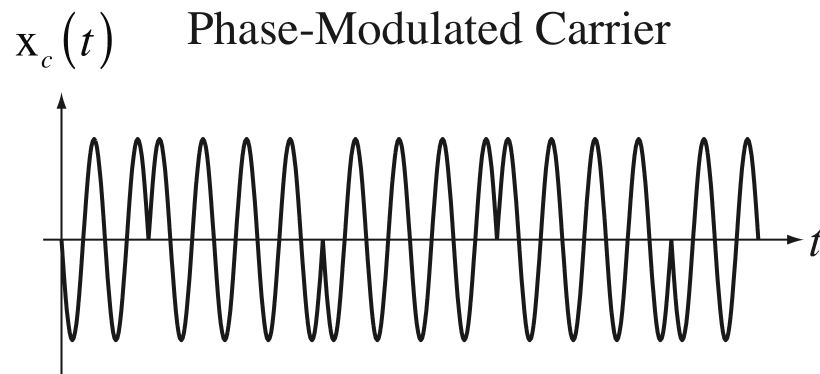
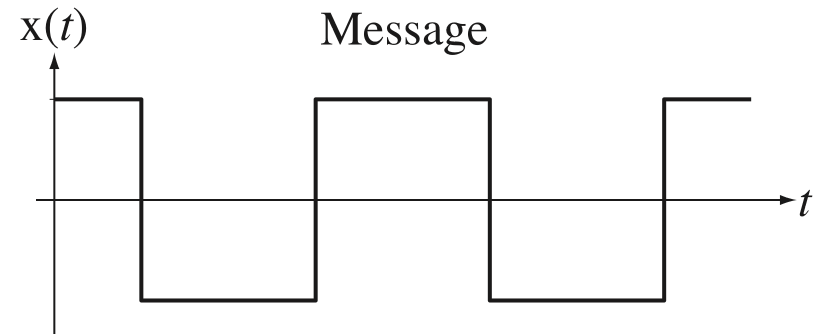
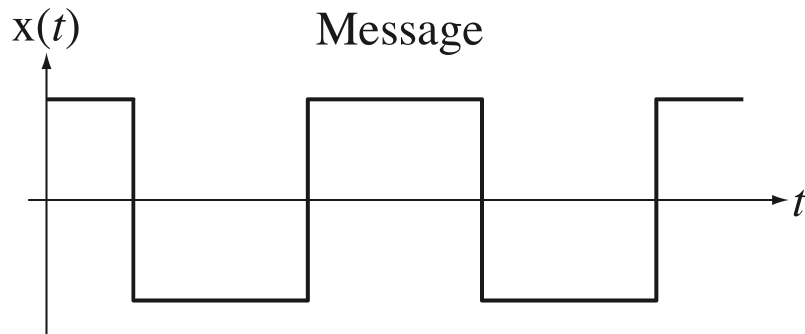
$$\phi(t) = \int_{t_0}^t 2\pi f_\Delta x(\lambda) d\lambda + \phi(t_0), \quad t \geq t_0$$

and

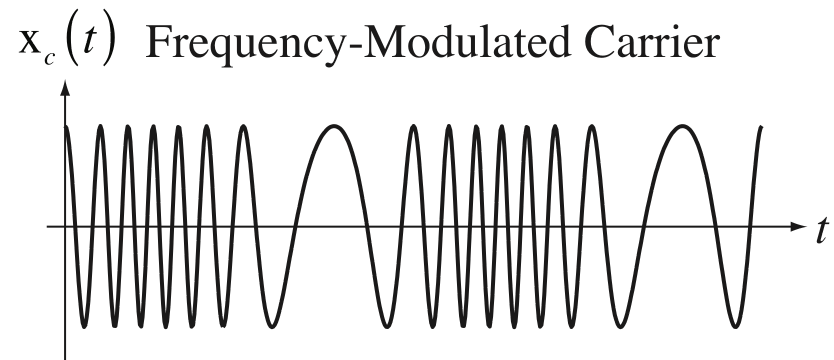
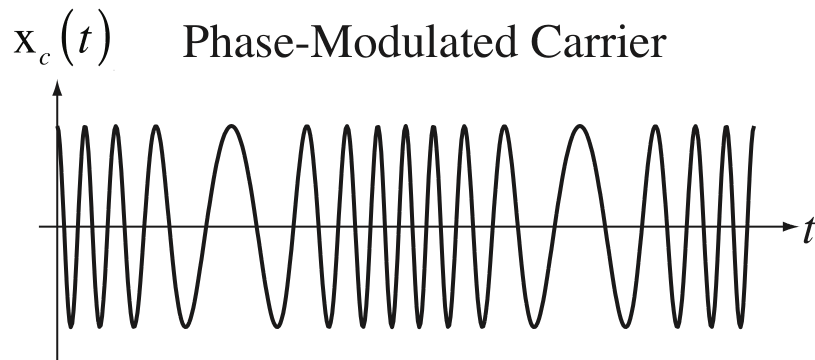
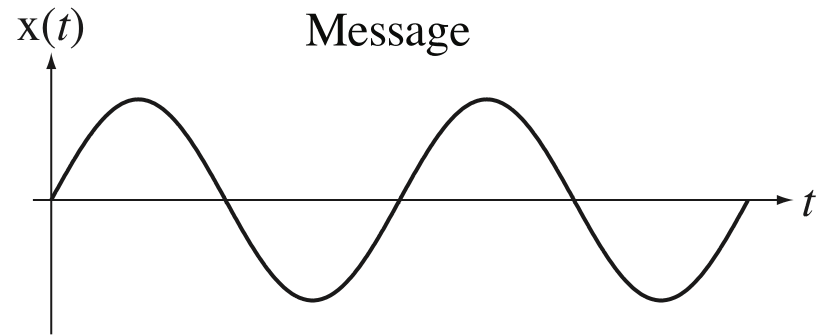
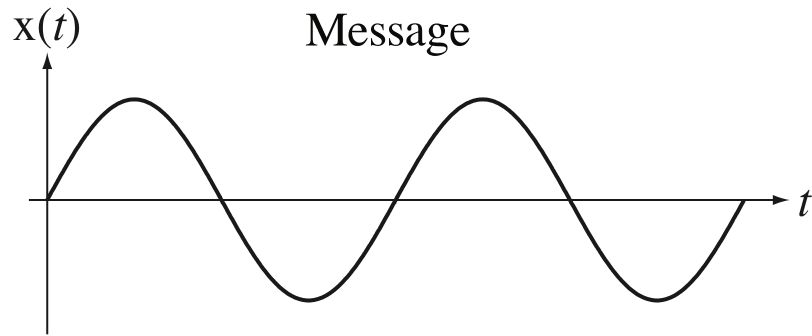
$$x_c(t) = A_c \cos \left(\omega_c t + 2\pi f_\Delta \int_{t_0}^t x(\lambda) d\lambda + \phi(t_0) \right).$$

So PM and FM are very similar. The difference is between integrating the message signal before phase modulating or not integrating it.

Phase and Frequency Modulation



Phase and Frequency Modulation



Phase and Frequency Modulation

For phase modulation $x_c(t) = A_c \cos(\omega_c t + \phi_\Delta x(t))$

For frequency modulation $x_c(t) = A_c \cos\left(\omega_c t + 2\pi f_\Delta \int_{t_0}^t x(\lambda) d\lambda\right)$

There is no simple expression for the Fourier transforms of these signals in the general case. Using $\cos(x + y) = \cos(x)\cos(y) - \sin(x)\sin(y)$

we can write for PM $x_c(t) = A_c \left[\cos(\omega_c t) \cos(\phi_\Delta x(t)) - \sin(\omega_c t) \sin(\phi_\Delta x(t)) \right]$

and for FM $x_c(t) = A_c \left[\cos(\omega_c t) \cos\left(2\pi f_\Delta \int_{t_0}^t x(\lambda) d\lambda\right) - \sin(\omega_c t) \sin\left(2\pi f_\Delta \int_{t_0}^t x(\lambda) d\lambda\right) \right]$

(under the assumption that $\phi(t_0) = 0$).

Phase and Frequency Modulation

If ϕ_Δ and f_Δ are small enough, $\cos(\phi_\Delta x(t)) \cong 1$ and $\sin(\phi_\Delta x(t)) \cong \phi_\Delta x(t)$

and $\cos\left(2\pi f_\Delta \int_{t_0}^t x(\lambda) d\lambda\right) \cong 1$ and $\sin\left(2\pi f_\Delta \int_{t_0}^t x(\lambda) d\lambda\right) \cong 2\pi f_\Delta \int_{t_0}^t x(\lambda) d\lambda$.

Then for PM $x_c(t) \cong A_c [\cos(\omega_c t) - \phi_\Delta x(t) \sin(\omega_c t)]$

and for FM $x_c(t) \cong A_c \left[\cos(\omega_c t) - 2\pi \sin(\omega_c t) f_\Delta \int_{t_0}^t x(\lambda) d\lambda \right]$

These approximations are called **narrowband PM** and **narrowband FM** and we can find their Fourier transforms.

For PM, $X_c(f) \cong (A_c/2) \left\{ [\delta(f-f_c) + \delta(f+f_c)] - j\phi_\Delta [X(f+f_c) - X(f-f_c)] \right\}$

For FM, $X_c(f) \cong (A_c/2) \left\{ [\delta(f-f_c) + \delta(f+f_c)] - f_\Delta \left[\frac{X(f+f_c)}{f+f_c} - \frac{X(f-f_c)}{f-f_c} \right] \right\}$

(under the assumption that the average value of $x(t)$ is zero)

Phase and Frequency Modulation

If the information signal is a sinusoid $x(t) = A_m \cos(\omega_m t) = A_m \cos(2\pi f_m t)$

then $X(f) = (A_m / 2) [\delta(f - f_m) + \delta(f + f_m)]$ and, in the narrowband approximation,

For PM,

$$x_c(t) \cong A_c \left[\cos(\omega_c t) - \phi_\Delta A_m \cos(\omega_m t) \sin(\omega_c t) \right]$$

$$X_c(f) \cong (A_c / 2) \left\{ \left[\delta(f - f_c) + \delta(f + f_c) \right] - \frac{jA_m \phi_\Delta}{2} \left[\begin{array}{l} \delta(f + f_c - f_m) + \delta(f + f_c + f_m) \\ -\delta(f - f_c - f_m) - \delta(f - f_c + f_m) \end{array} \right] \right\}$$

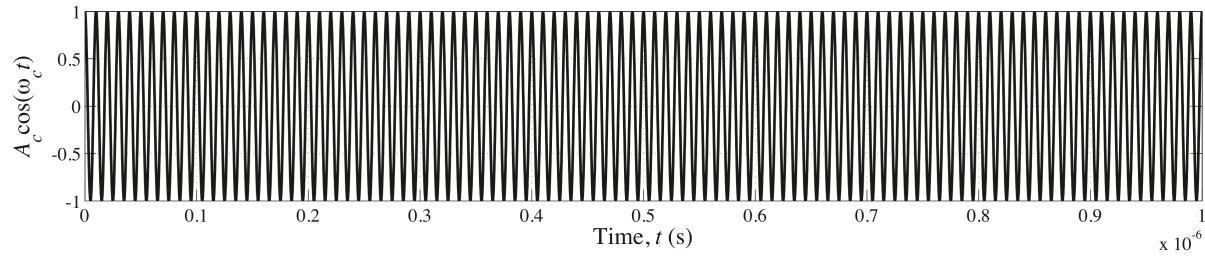
For FM,

$$x_c(t) \cong A_c \left[\cos(\omega_c t) - \frac{f_\Delta A_m}{f_m} \sin(\omega_c t) \sin(\omega_m t) \right]$$

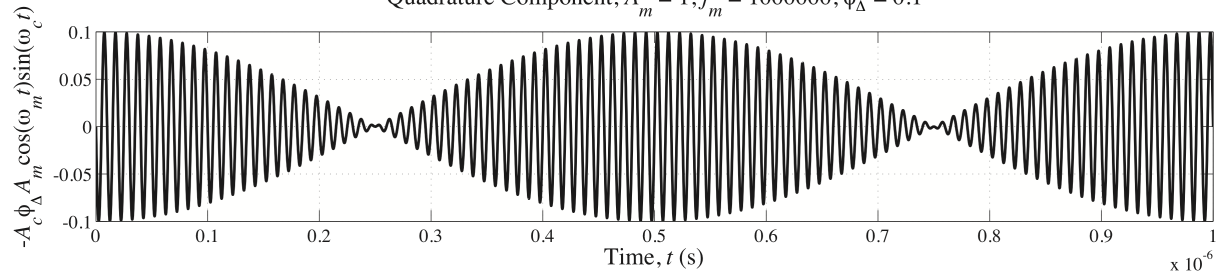
$$X_c(f) \cong (A_c / 2) \left\{ \left[\delta(f - f_c) + \delta(f + f_c) \right] - \frac{A_m f_\Delta}{2 f_m} \left[\begin{array}{l} \delta(f + f_c - f_m) - \delta(f + f_c + f_m) \\ -\delta(f - f_c - f_m) + \delta(f - f_c + f_m) \end{array} \right] \right\}$$

Phase and Frequency Modulation

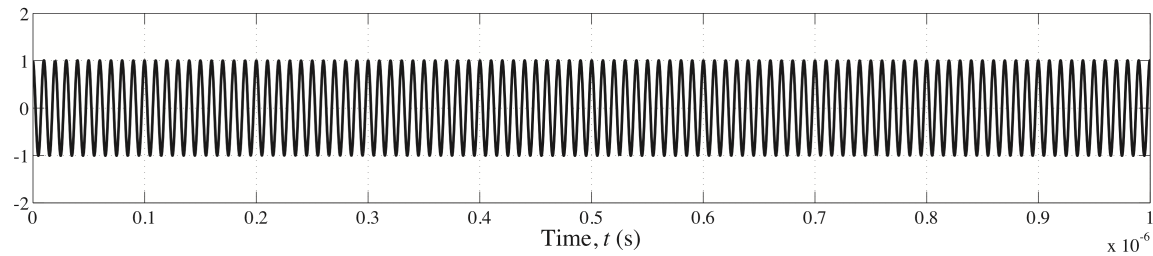
Unmodulated Carrier, $A_c = 1, f_c = 100000000$



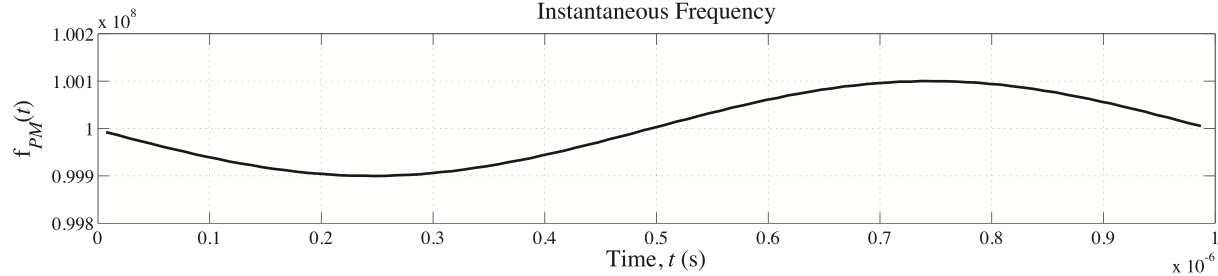
Quadrature Component, $A_m = 1, f_m = 1000000, \phi_\Delta = 0.1$



Narrowband PM

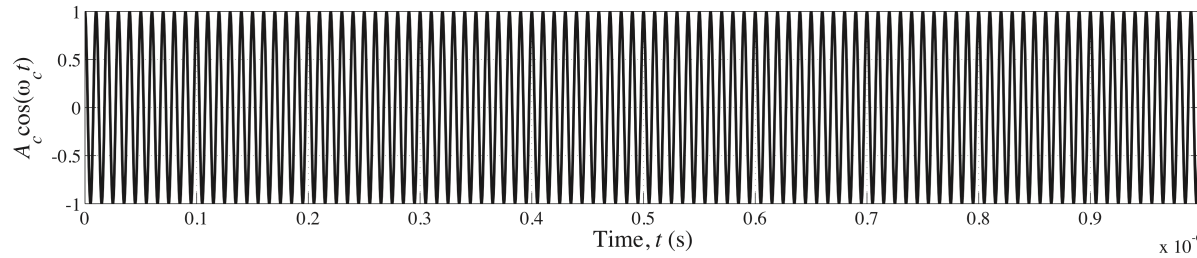


Instantaneous Frequency

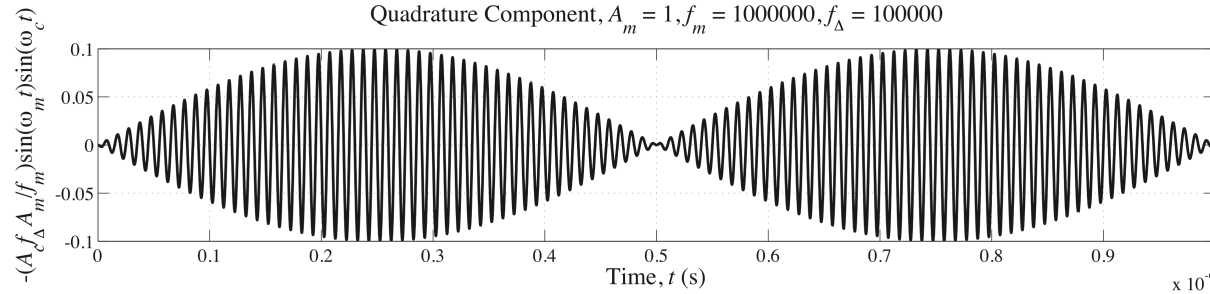


Phase and Frequency Modulation

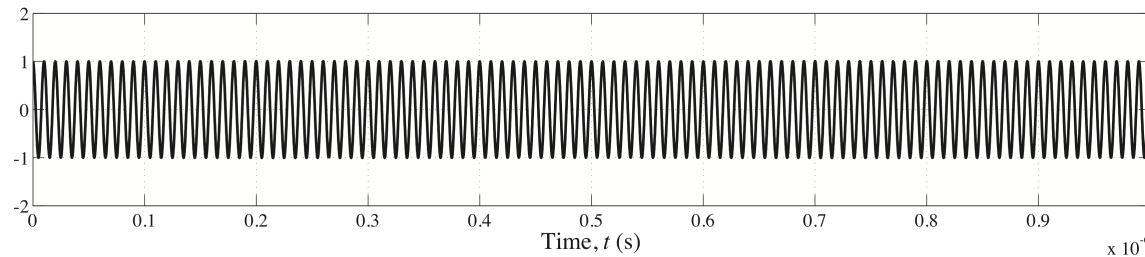
Unmodulated Carrier, $A_c = 1, f_c = 100000000$



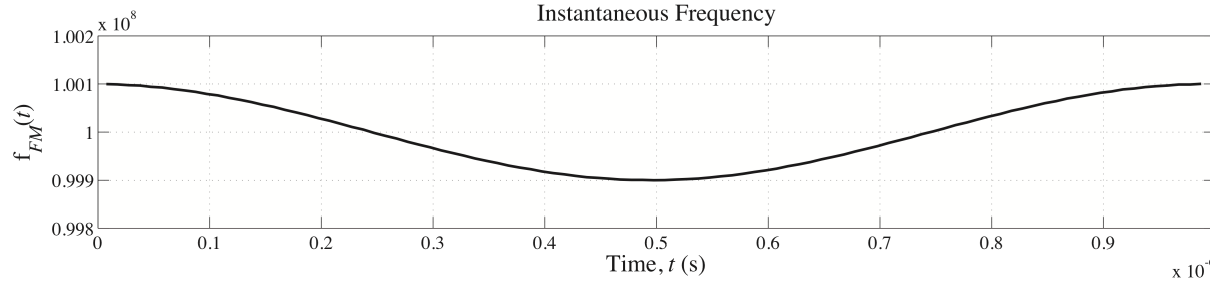
Quadrature Component, $A_m = 1, f_m = 1000000, f_\Delta = 100000$



Narrowband FM



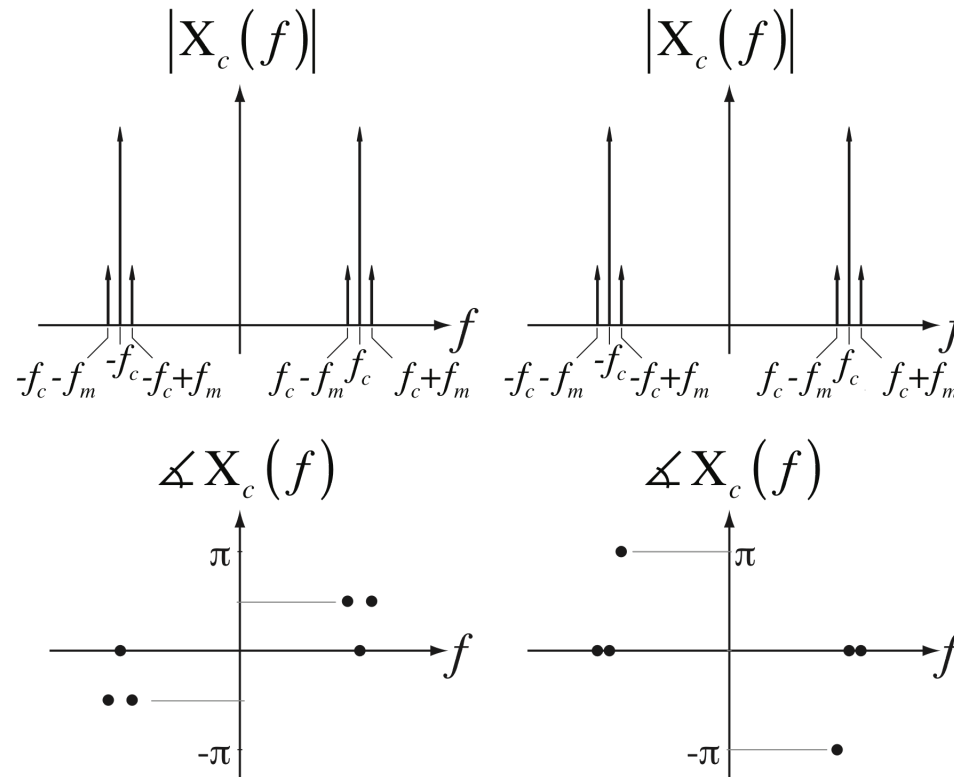
Instantaneous Frequency



Phase and Frequency Modulation

Narrowband PM and FM Spectra
for Tone Modulation

Tone-Modulated PM Tone-Modulated FM



Phase and Frequency Modulation

If the information signal is a sinc, $x(t) = \text{sinc}(2Wt)$ then $X(f) = (1/2W)\text{rect}(f/2W)$ and, in the narrowband approximation,

For PM,

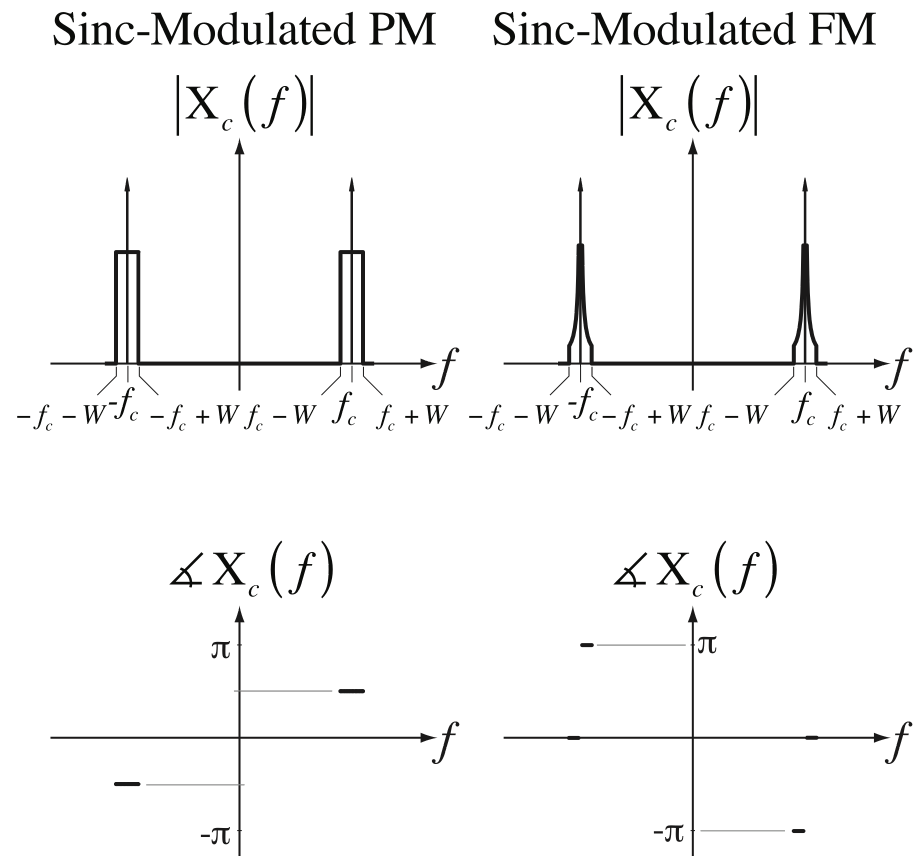
$$X_c(f) \cong (A_c/2) \left\{ \left[\delta(f - f_c) + \delta(f + f_c) \right] - j \frac{\phi_\Delta}{2W} \left[\text{rect}((f + f_c)/2W) - \text{rect}((f - f_c)/2W) \right] \right\}$$

For FM,

$$X_c(f) \cong (A_c/2) \left\{ \left[\delta(f - f_c) + \delta(f + f_c) \right] - \frac{f_m f_\Delta}{2W} \left[\frac{\text{rect}((f + f_c)/2W)}{f + f_c} - \frac{\text{rect}((f - f_c)/2W)}{f - f_c} \right] \right\}$$

Phase and Frequency Modulation

Narrowband PM and FM Spectra
for a Sinc Message



Phase and Frequency Modulation

If the narrowband approximation is not adequate we must deal with the more complicated wideband case. In the case of tone modulation we can handle PM and FM with basically the same analysis technique if we use the following conventions:

$$x(t) = \begin{cases} A_m \sin(\omega_m t) & , \text{ PM} \\ A_m \cos(\omega_m t) & , \text{ FM} \end{cases}$$

$$\text{For FM, } \phi(t) = 2\pi f_\Delta \int_{t_0}^t x(\lambda) d\lambda = 2\pi f_\Delta \int_{t_0}^t A_m \cos(\omega_m \lambda) d\lambda$$

$$\phi(t) = 2\pi \frac{A_m}{\omega_m} f_\Delta \sin(\omega_m t) = \frac{A_m}{f_m} f_\Delta \sin(\omega_m t)$$

Then, for PM and FM, $\phi(t) = \beta \sin(\omega_m t)$, where $\beta \triangleq \begin{cases} \phi_\Delta A_m & , \text{ PM} \\ (A_m / f_m) f_\Delta & , \text{ FM} \end{cases}$

$$\text{Then } x_c(t) = A_c \left[\cos(\beta \sin(\omega_m t)) \cos(\omega_c t) - \sin(\beta \sin(\omega_m t)) \sin(\omega_c t) \right]$$

Phase and Frequency Modulation

In $x_c(t) = A_c \left[\cos(\beta \sin(\omega_m t)) \cos(\omega_c t) - \sin(\beta \sin(\omega_m t)) \sin(\omega_c t) \right]$
 $\cos(\beta \sin(\omega_m t))$ and $\sin(\beta \sin(\omega_m t))$ are periodic with fundamental period $2\pi / \omega_m$. We can now use two useful results from applied mathematics (Abramowitz and Stegun, page 361)

$$\cos(z \sin(\theta)) = J_0(z) + 2 \sum_{k=1}^{\infty} J_{2k}(z) \cos(2k\theta) = J_0(z) + 2 \sum_{\substack{k=1 \\ k \text{ even}}}^{\infty} J_k(z) \cos(k\theta)$$

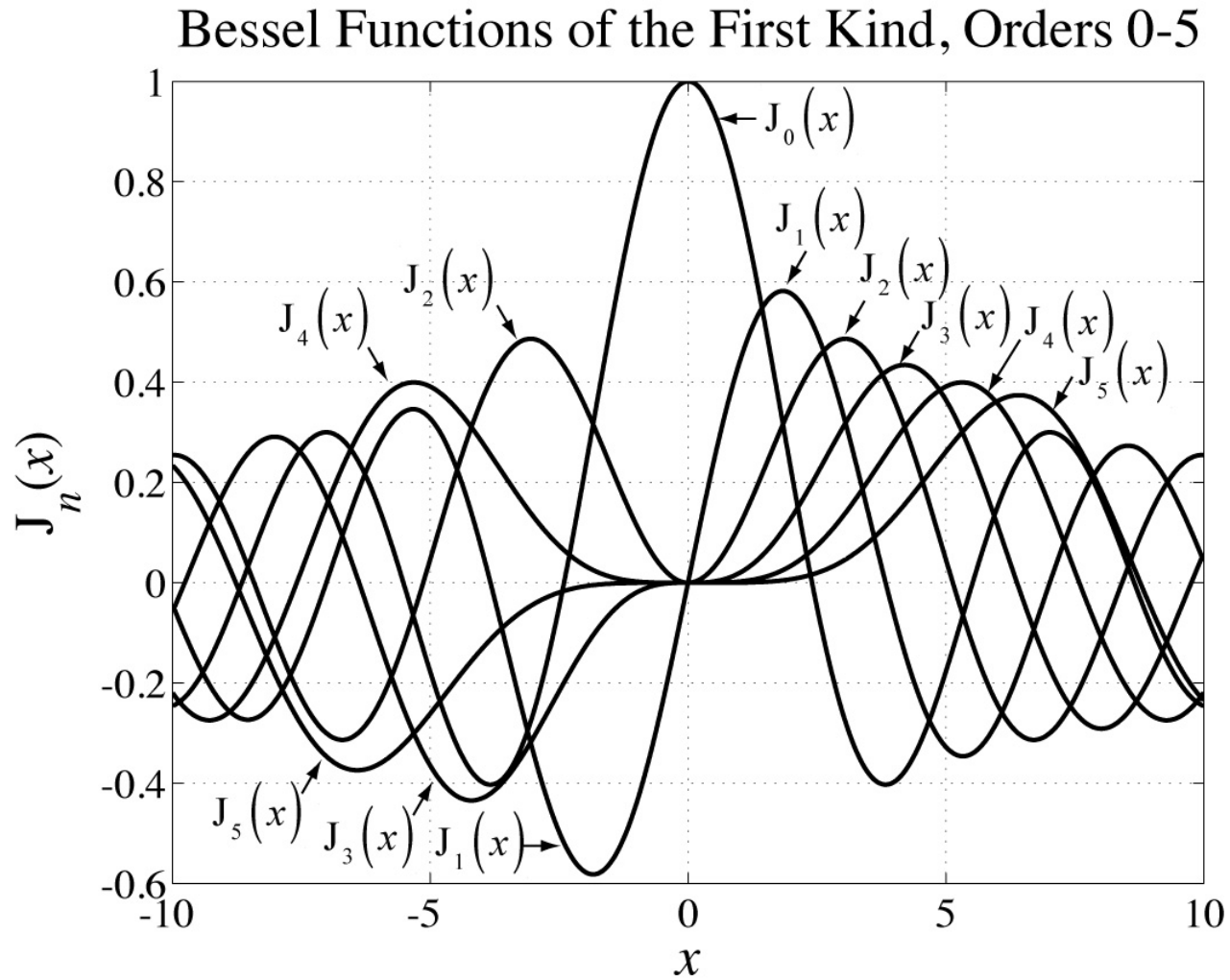
$$\sin(z \sin(\theta)) = 2 \sum_{k=0}^{\infty} J_{2k+1}(z) \sin((2k+1)\theta) = 2 \sum_{\substack{k=1 \\ k \text{ odd}}}^{\infty} J_k(z) \sin(k\theta)$$

Adapting them to our case

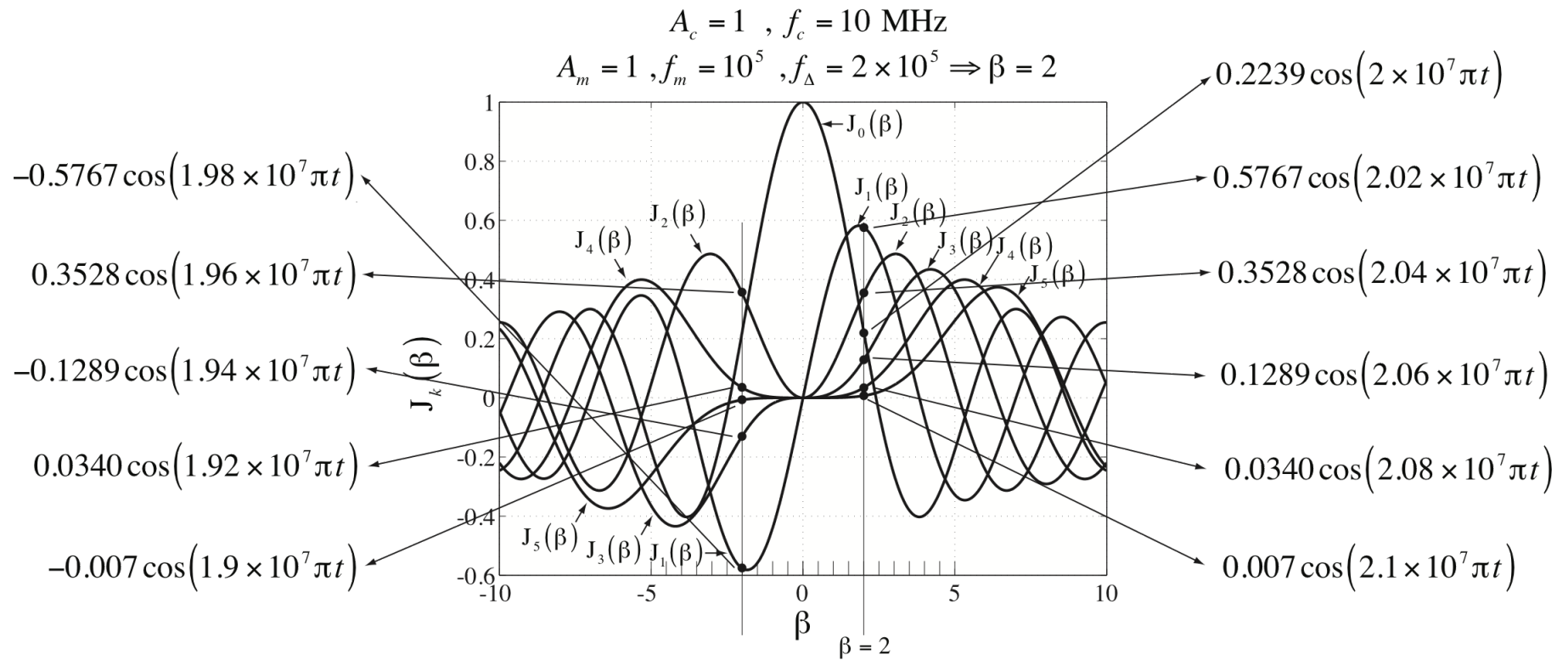
$$\cos(\beta \sin(\omega_m t)) = J_0(\beta) + 2 \sum_{\substack{k=1 \\ k \text{ even}}}^{\infty} J_k(\beta) \cos(k\omega_m t)$$

$$\sin(\beta \sin(\omega_m t)) = 2 \sum_{\substack{k=1 \\ k \text{ odd}}}^{\infty} J_k(\beta) \sin(k\omega_m t)$$

Phase and Frequency Modulation



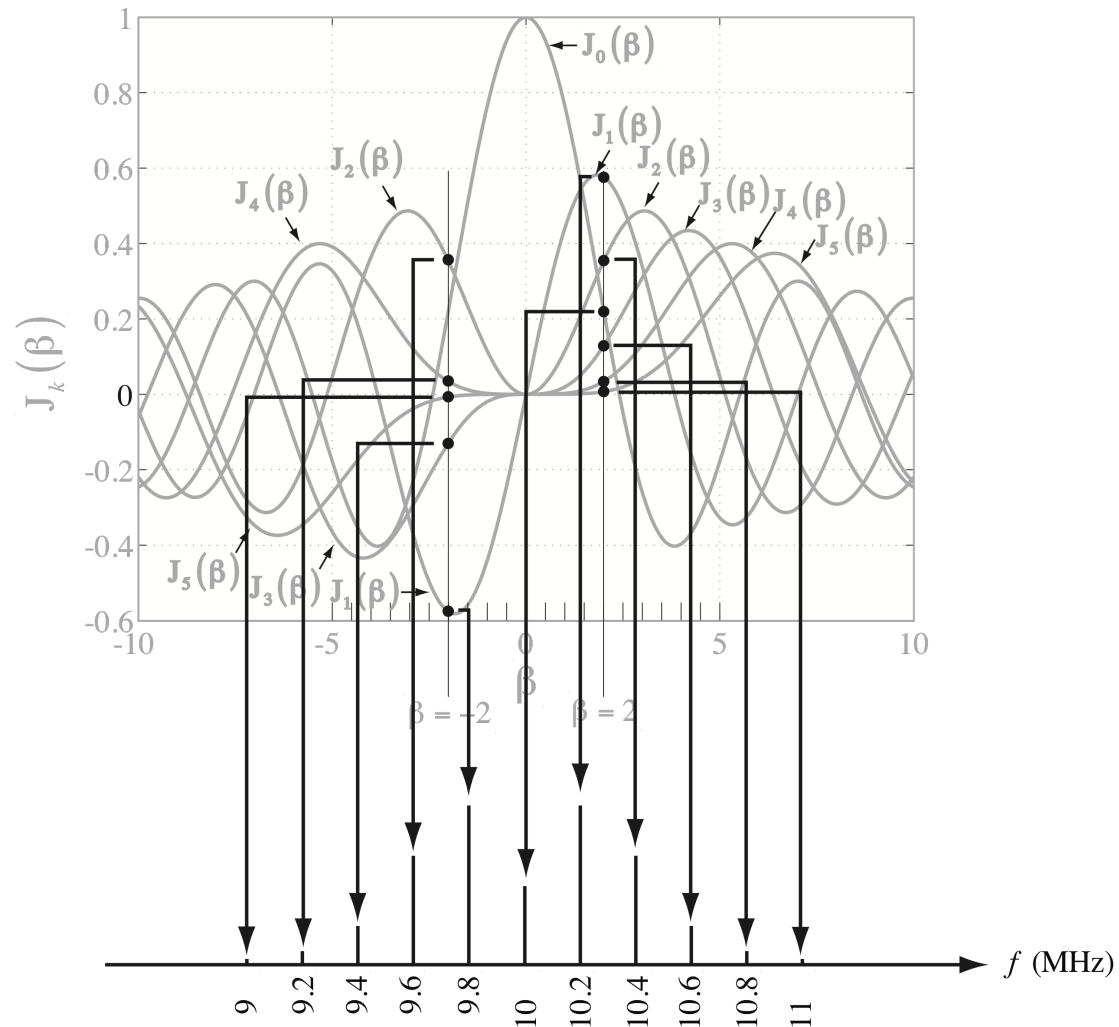
Phase and Frequency Modulation



Phase and Frequency Modulation

$$A_c = 1, f_c = 10 \text{ MHz}$$

$$A_m = 1, f_m = 10^5, f_\Delta = 2 \times 10^5 \Rightarrow \beta = 2$$



Phase and Frequency Modulation

$$x_c(t) = A_c \left\{ \left[J_0(\beta) + 2 \sum_{\substack{k=1 \\ k \text{ even}}}^{\infty} J_k(\beta) \cos(k\omega_m t) \right] \cos(\omega_c t) - \left[2 \sum_{\substack{k=1 \\ k \text{ odd}}}^{\infty} J_k(\beta) \sin(k\omega_m t) \right] \sin(\omega_c t) \right\}$$

$$x_c(t) = A_c \left\{ \begin{aligned} & J_0(\beta) \cos(\omega_c t) + 2 \sum_{\substack{k=1 \\ k \text{ even}}}^{\infty} J_k(\beta) \cos(\omega_c t) \cos(k\omega_m t) \\ & - 2 \sum_{\substack{k=1 \\ k \text{ odd}}}^{\infty} J_k(\beta) \sin(\omega_c t) \sin(k\omega_m t) \end{aligned} \right\}$$

$$x_c(t) = A_c \left\{ \begin{aligned} & J_0(\beta) \cos(\omega_c t) + \sum_{\substack{k=1 \\ k \text{ even}}}^{\infty} J_k(\beta) \left[\cos((\omega_c - k\omega_m)t) + \cos((\omega_c + k\omega_m)t) \right] \\ & - \sum_{\substack{k=1 \\ k \text{ odd}}}^{\infty} J_k(\beta) \left[\cos((\omega_c - k\omega_m)t) - \cos((\omega_c + k\omega_m)t) \right] \end{aligned} \right\}$$

This can also be written in the more compact form, $x_c(t) = A_c \sum_{k=-\infty}^{\infty} J_k(\beta) \cos((\omega_c + k\omega_m)t)$

Phase and Frequency Modulation

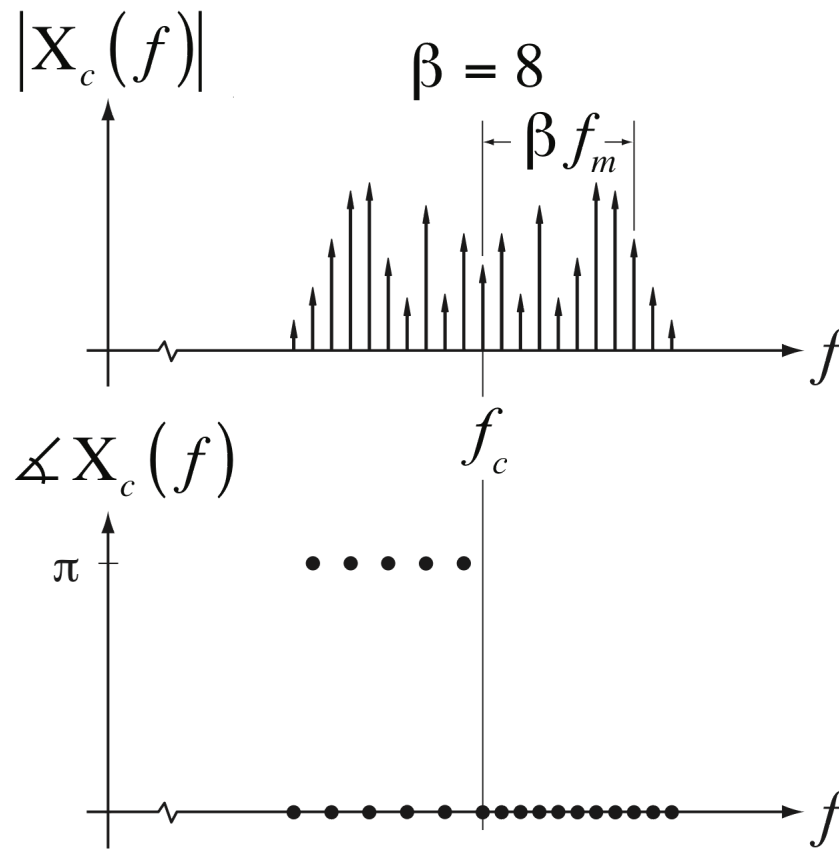
Now, to find the spectrum of $x_c(t)$ take the Fourier transform of $x_c(t)$.

$$X_c(f) = (A_c / 2) \sum_{k=-\infty}^{\infty} J_k(\beta) \left[\delta(f - (f_c + kf_m)) + \delta(f + (f_c + kf_m)) \right]$$

The impulses in the spectrum extend in frequency all the way to infinity. But beyond βf_m the impulse strengths die rapidly. For practical purposes the bandwidth is approximately $2\beta f_m$.

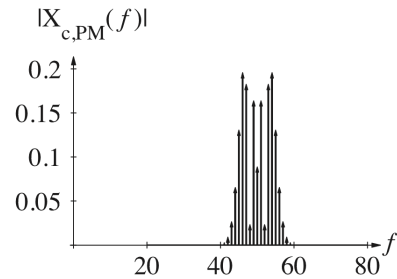
Phase and Frequency Modulation

Wideband FM Spectrum
for Cosine-Wave Modulation

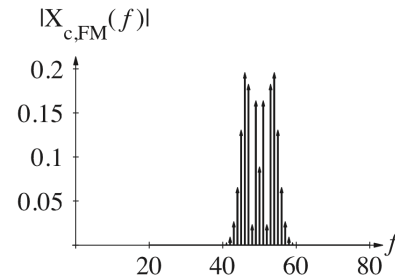


Phase and Frequency Modulation

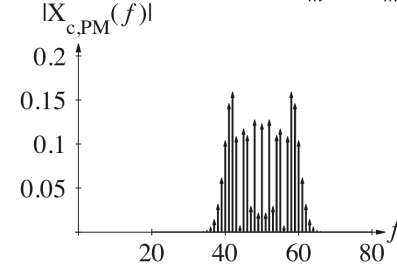
Tone PM, $\beta = 5, \phi_\Delta = 5, A_m = 1, f_m = 1$



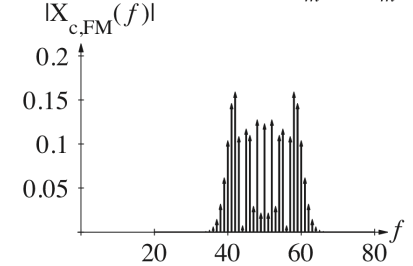
Tone FM, $\beta = 5, f_\Delta = 5, A_m = 1, f_m = 1$



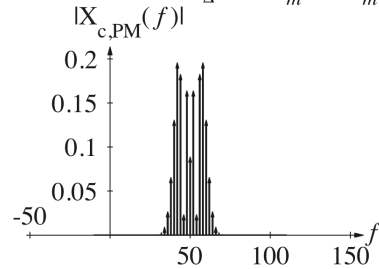
Tone PM, $\beta = 10, \phi_\Delta = 5, A_m = 2, f_m = 1$



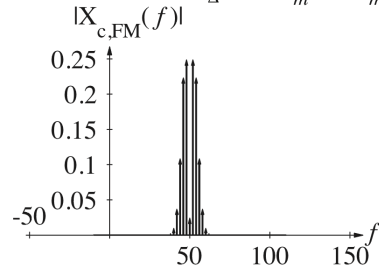
Tone FM, $\beta = 10, f_\Delta = 5, A_m = 2, f_m = 1$



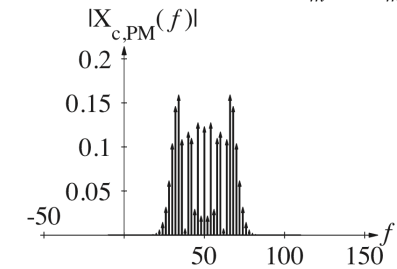
Tone PM, $\beta = 5, \phi_\Delta = 5, A_m = 1, f_m = 2$



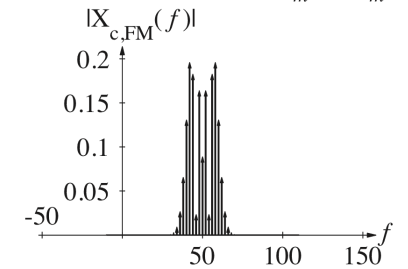
Tone FM, $\beta = 2.5, f_\Delta = 5, A_m = 1, f_m = 2$



Tone PM, $\beta = 10, \phi_\Delta = 5, A_m = 2, f_m = 2$



Tone FM, $\beta = 5, f_\Delta = 5, A_m = 2, f_m = 2$



Transmission Bandwidth and Distortion

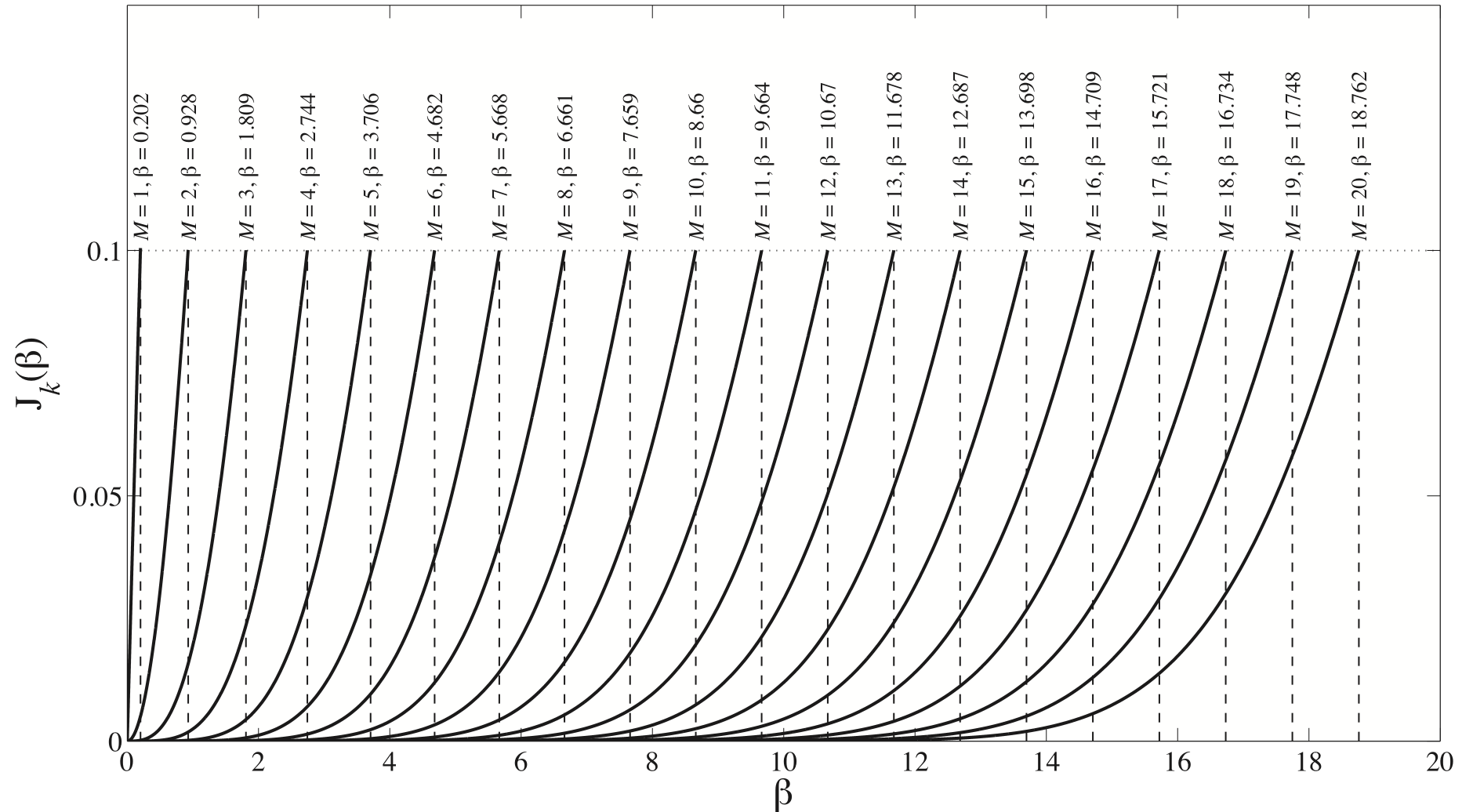
The bandwidth required for transmitting an FM signal is theoretically infinite. That is, an infinite bandwidth would be required to transmit an FM signal *perfectly*, even if the modulating signal is bandlimited. Fortunately, in practical systems, perfection is not required and we can get by with a finite bandwidth. With tone modulation, the bandwidth required depends on the modulation index β . The spectral line magnitudes fall off rapidly at positive frequencies for which $|f - f_c| > \beta f_m$. So for tone modulation the bandwidth required for transmission would be approximately $2\beta f_m$. In the narrowband case when β is very small we cannot exactly follow this rule because we would have no modulation at all. So there is a "floor" of at least $2f_m$.

Transmission Bandwidth and Distortion

In determining bandwidth what really matters is the worst case and how much distortion we can tolerate. Suppose we agree that any spectral lines of magnitude less than ε can be omitted. Of course the value of ε depends on the application. Typical values lie in the range $0.01 < \varepsilon < 0.1$. If $|J_M(\beta)| > \varepsilon$ and $|J_{M+1}(\beta)| < \varepsilon$ then the bandwidth for transmitting that tone modulation would be $B = 2M(\beta)f_m$ and we put a lower limit $M(\beta) \geq 1$ to account for the bandwidth floor in the narrowband case.

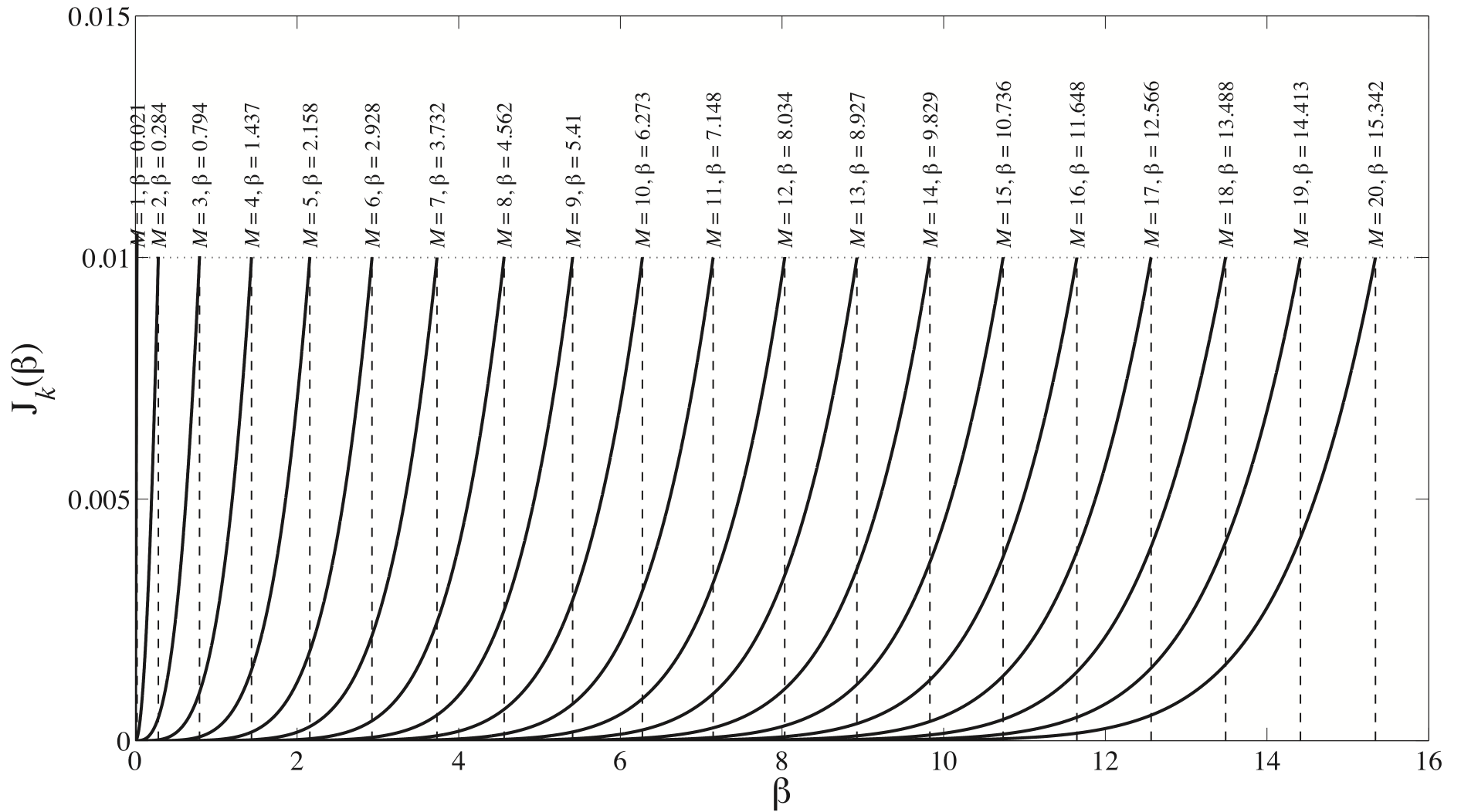
Transmission Bandwidth and Distortion

$\varepsilon = 0.1$ Case



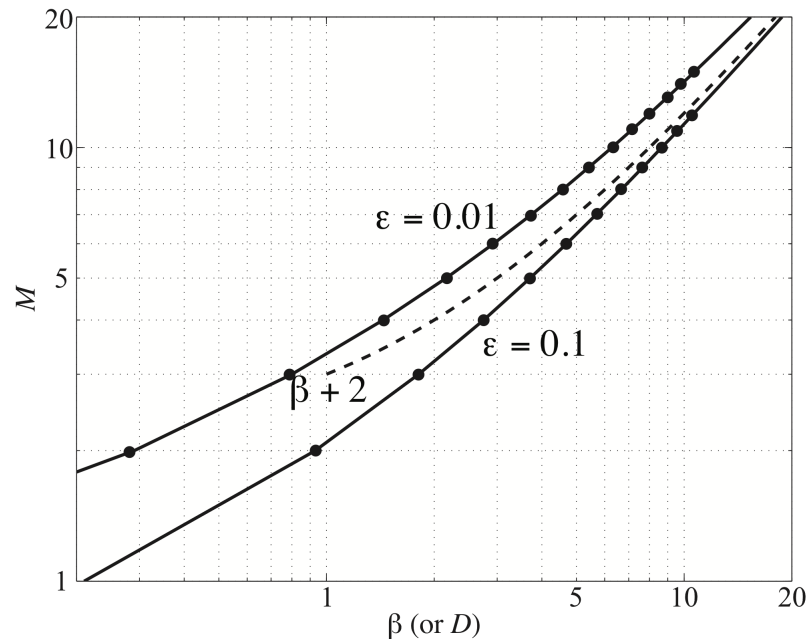
Transmission Bandwidth and Distortion

$\varepsilon = 0.01$ Case



Transmission Bandwidth and Distortion

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Transmission Bandwidth and Distortion

In the equation $B = 2M(\beta)f_m$, if we substitute $\beta + 2$ for $M(\beta)$ we get

$$B \cong 2(\beta + 2)f_m = 2\left(\frac{A_m f_\Delta}{f_m} + 2\right)f_m = 2(A_m f_\Delta + 2f_m). \text{ To estimate the actual}$$

required bandwidth for transmission of FM we take the worst case and set

$A_m = 1$ and $f_m = W$. Then $B \cong 2(f_\Delta + 2W)$ for $\beta > 2$. This estimate of required transmission bandwidth is based on tone modulation but it can be shown that it is a reasonable estimate for any general modulation of the same bandwidth.

The **deviation ratio** is defined by $D \triangleq f_\Delta / W$. It is the maximum phase deviation under worst case conditions. It serves the same purpose for general modulation that β does for tone modulation. Then $B_T = 2M(D)W$ and we can use the same relationship between M and β used earlier to find $M(D)$. **Carson's rule** is a handy approximation based on these principles that says $B_T \cong 2(D + 1)W$ for either $D \gg 1$ or $D \ll 1$. But for the more common case of $2 < D < 10$ a better approximation is $B_T \cong 2(D + 2)W$, $2 < D < 10$.

Generation and Detection of FM and PM

The most direct and straightforward way of generating FM is to use a device known as a **voltage-to-frequency converter (VCO)**. One way this can be done is by varying with time the capacitance in an LC parallel resonant oscillator. Let the capacitance be the capacitance of a varactor diode in parallel with another capacitor forming $C(t) = C_0 - C x(t)$. The time-varying LC resonant frequency is

$$f(t) = \frac{1}{2\pi} \frac{d}{dt}(\theta(t)) \Rightarrow \frac{d}{dt}(\theta(t)) = \frac{1}{\sqrt{LC(t)}} = \frac{1}{\sqrt{LC_0}} \frac{1}{\sqrt{1 - \frac{C}{C_0} x(t)}}$$

We can use the formula (Abramowitz and Stegun, page 15),

$$(1+x)^\alpha = 1 + \alpha x + \frac{\alpha(\alpha-1)}{2!} x^2 + \frac{\alpha(\alpha-1)(\alpha-2)}{3!} x^3 + \dots$$

to write

$$\frac{d}{dt}(\theta(t)) = \frac{1}{\sqrt{LC_0}} \left[1 - \frac{C}{C_0} x(t) \right]^{-1/2} = \frac{1}{\sqrt{LC_0}} \left[1 + \frac{1}{2} \frac{C}{C_0} x(t) + \frac{3}{8} \left(\frac{C}{C_0} x(t) \right)^2 + \dots \right]$$

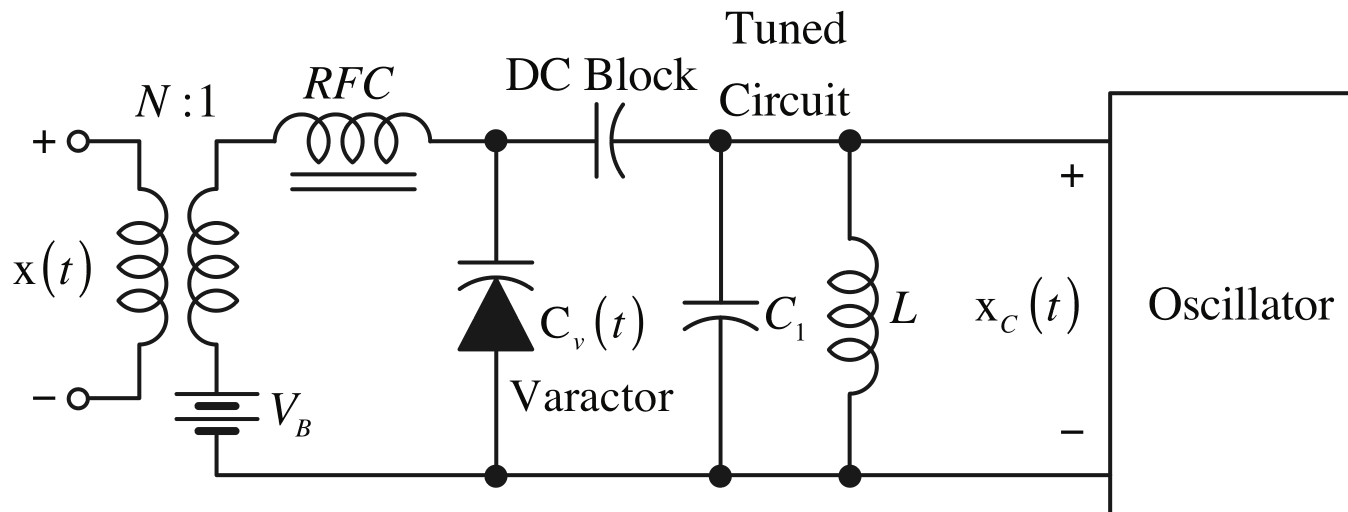
Generation and Detection of FM and PM

If $C x(t)$ is "small enough", then $\frac{d}{dt}(\theta(t)) \cong \frac{1}{\sqrt{LC_0}} \left[1 + \frac{1}{2} \frac{C}{C_0} x(t) \right]$ and

$\theta(t) = 2\pi f_c t + 2\pi \frac{C}{2C_0} f_c \int^t x(\lambda) d\lambda$. This is in the form of FM with $f_\Delta = \frac{C}{2C_0} f_c$.

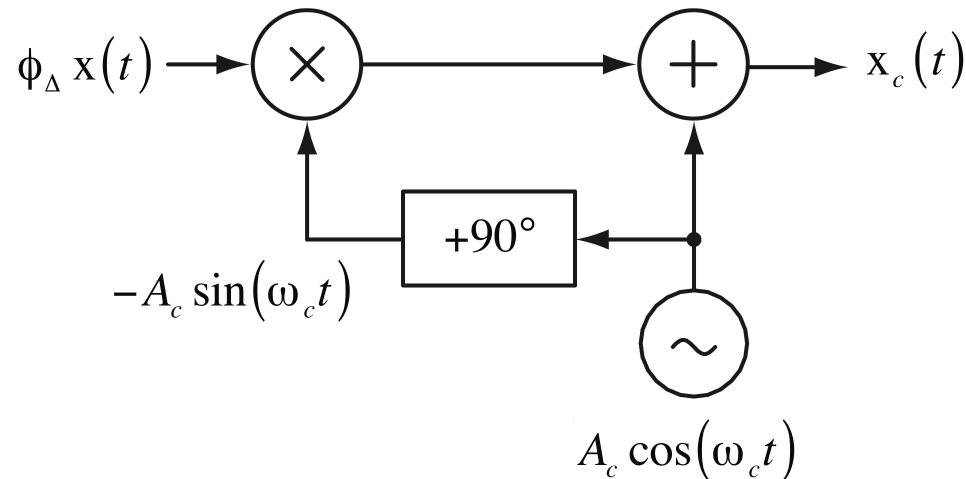
Since $|x(t)| \leq 1$, the approximation is good to within one percent if $C / C_0 < 0.013$.

So, taking that as an upper limit, $f_\Delta = \frac{C}{2C_0} f_c \leq 0.006 f_c$. This is a practical result that usually causes no design problems.



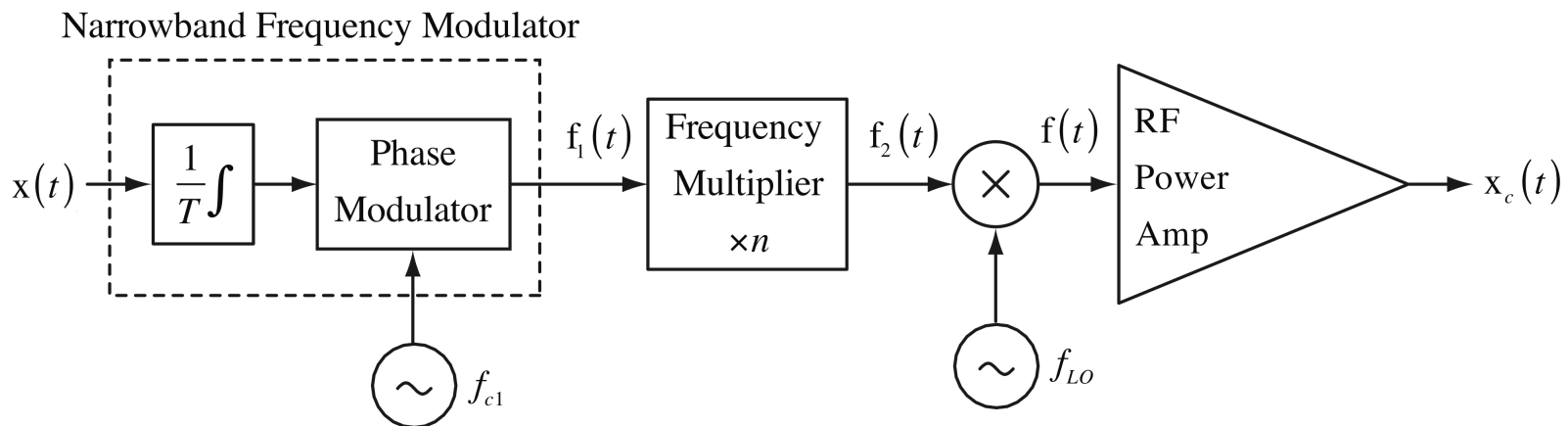
Generation and Detection of FM and PM

Another method for generating FM is to use a phase modulator, which produces PM, but integrate the message before applying it to the phase modulator. A narrowband phase modulator can be made by simulating the narrowband approximation $x_c(t) = A_c \cos(\omega_c t) - A_c \phi_\Delta x(t) \sin(\omega_c t)$.



Generation and Detection of FM and PM

A third method for generating FM is called **indirect FM**. First, integrate the message $x(t)$. Then use the integral of the message $\frac{1}{T} \int^t x(\lambda) d\lambda$ as the input signal to a narrowband phase modulator with a carrier frequency f_{c1} . This produces a signal with instantaneous frequency $f_1(t) = f_{c1} + \frac{\phi_{\Delta}}{2\pi T} x(t)$.



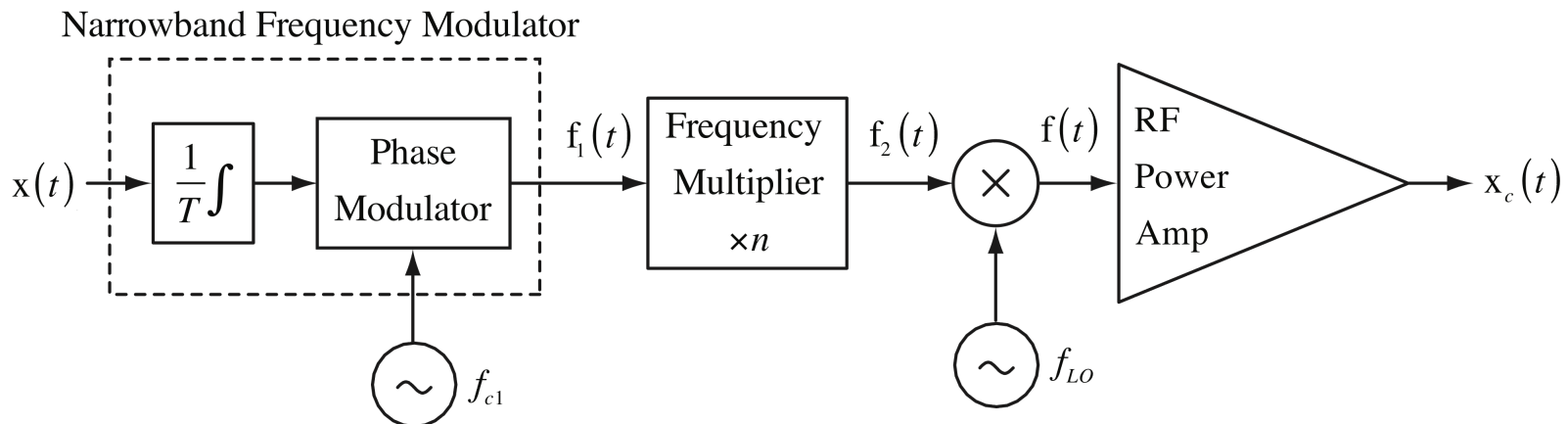
Generation and Detection of FM and PM

Next frequency-multiply the narrowband FM signal by a factor of n . This moves the carrier frequency to nf_{c1} , creating a signal with instantaneous frequency

$$f_2(t) = nf_{c1} + n \frac{\phi_{\Delta}}{2\pi T} x(t). \text{ The effective value of the frequency deviation is now}$$

$$f_{\Delta} = n \frac{\phi_{\Delta}}{2\pi T}. \text{ This changes the range of frequency variation but not the rate of$$

frequency variation. Then, if needed, shift the entire FM spectrum to whatever carrier frequency is required and amplify for transmission.



Generation and Detection of FM and PM

There are four common methods of detecting FM:

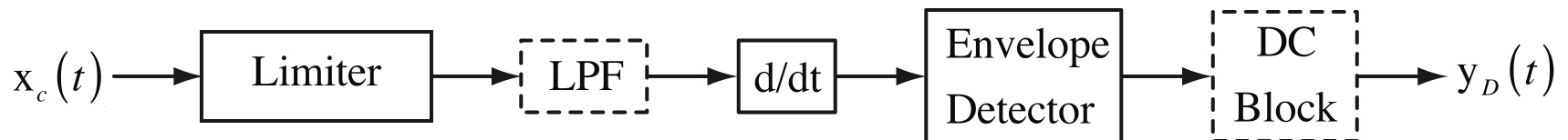
1. FM-to-AM Conversion Followed by Envelope Detection
2. Phase-Shift Discrimination
3. Zero-Crossing Detection
4. Frequency Feedback

FM-to-AM conversion can be done by time-differentiating the modulated signal.

Let $x_c(t) = A_c \cos(\theta_c(t))$ with $\dot{\theta}(t) = 2\pi[f_c + f_\Delta x(t)]$. Then

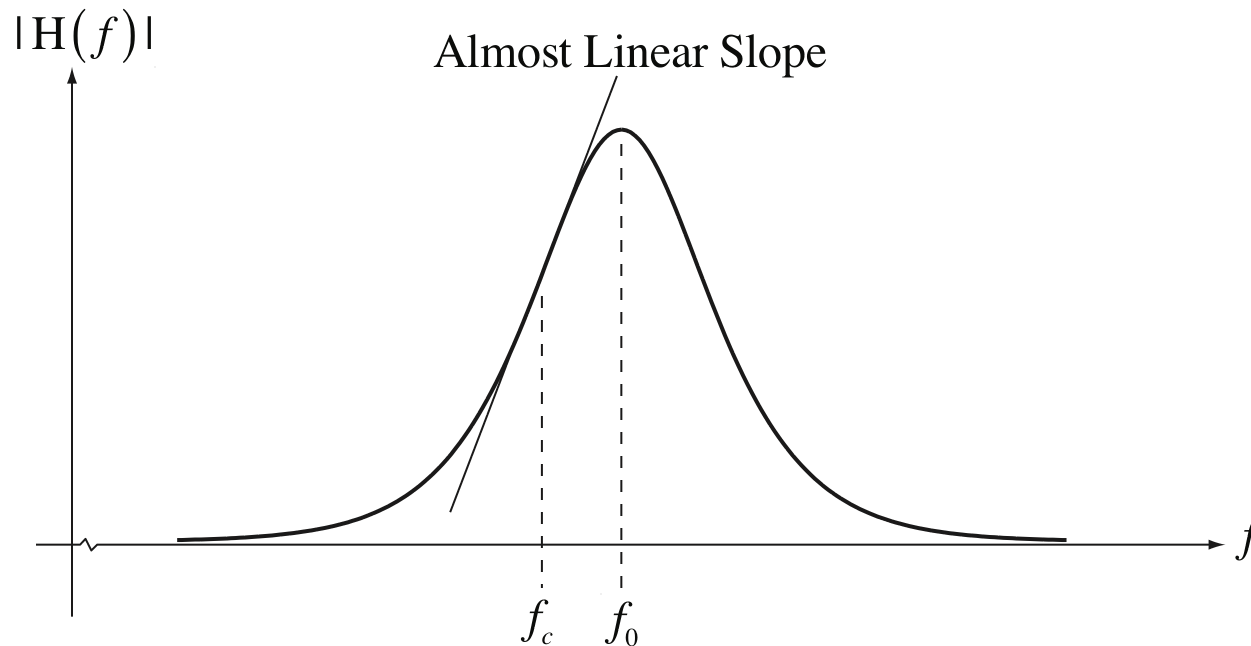
$$\dot{x}_c(t) = -A_c \dot{\theta}(t) \sin(\theta_c(t)) = 2\pi A_c [f_c + f_\Delta x(t)] \sin(\theta_c(t) \pm 180^\circ).$$

The message can then be recovered by an envelope detector.



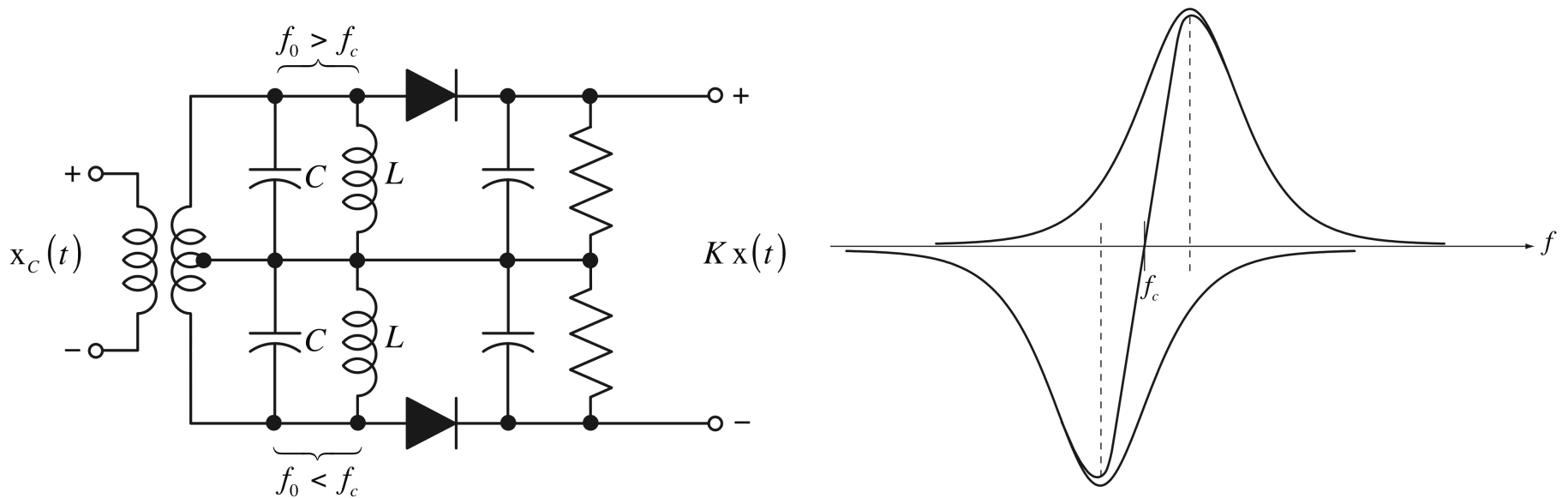
Generation and Detection of FM and PM

The "differentiator" in FM-to-AM detection need not be a true differentiator. All that is really needed is a frequency response magnitude that has a linear (or almost linear) slope over the bandwidth of the FM signal. Just below and just above resonance a tuned circuit resonator has an almost linear magnitude dependence on frequency. This type of detection is commonly called **slope detection**.



Generation and Detection of FM and PM

The linearity of slope detection can be improved by using two resonant circuits instead of only one. This type of circuit is called a **balanced discriminator**.



Interference

Let the total received signal at a receiver be

$$v(t) = A_c \cos(\omega_c t) + A_i \cos((\omega_c + \omega_i)t + \phi_i)$$

where the first term represents the desired signal and the second term represents interference. Also define $\rho \triangleq A_i / A_c$ and $\theta_i(t) \triangleq \omega_i t + \phi_i$. Then

$$v(t) = A_c \left[\cos(\omega_c t) + \rho \cos(\omega_c t + \theta_i) \right] = A_c \left\{ \cos(\omega_c t) + \rho \begin{bmatrix} \cos(\omega_c t) \cos(\theta_i(t)) \\ -\sin(\omega_c t) \sin(\theta_i(t)) \end{bmatrix} \right\}$$
$$v(t) = A_c \left\{ \left[1 + \rho \cos(\theta_i(t)) \right] \cos(\omega_c t) - \rho \sin(\theta_i(t)) \sin(\omega_c t) \right\}$$

The in-phase component is $A_c \left[1 + \rho \cos(\theta_i(t)) \right] \cos(\omega_c t)$ and the quadrature component is $-A_c \rho \sin(\theta_i(t)) \sin(\omega_c t)$. The envelope is

$$A_v(t) = A_c \sqrt{\left[1 + \rho \cos(\theta_i) \right]^2 + \rho^2 \sin^2(\theta_i(t))} = A_c \sqrt{1 + \rho^2 + 2\rho \cos(\theta_i(t))}. \text{ The phase relative}$$

to the desired signal is $\phi_v(t) = \tan^{-1} \left(\frac{\rho \sin(\theta_i(t))}{1 + \rho \cos(\theta_i(t))} \right)$.

Interference

The envelope and phase of the total received signal

$$A_v(t) = A_c \sqrt{1 + \rho^2 + 2\rho \cos(\theta_i(t))} \text{ and } \phi_v(t) = \tan^{-1} \left(\frac{\rho \sin(\theta_i(t))}{1 + \rho \cos(\theta_i(t))} \right)$$

show that the effect of the interference on the received signal is to create both amplitude and phase modulation. If $\rho \ll 1$, then

$$A_v(t) \cong A_c \sqrt{1 + 2\rho \cos(\theta_i(t))} \cong A_c [1 + \rho \cos(\theta_i(t))] \text{ and } \phi_v(t) \cong \tan^{-1}(\rho \sin(\theta_i(t))) \cong \rho \sin(\theta_i(t))$$

or

$$A_v(t) \cong A_c [1 + \rho \cos(\omega_i t + \phi_i)] \text{ and } \phi_v(t) \cong \rho \sin(\omega_i t + \phi_i)$$

This result has the form of AM tone modulation with $\mu = \rho$ and simultaneous PM or FM tone modulation with $\beta = \rho$. If $\rho \gg 1$, then

$$A_v(t) = \rho A_c \sqrt{1 + 2\rho^{-1} \cos(\omega_i t + \phi_i)} \cong \rho A_c [1 + \rho^{-1} \cos(\omega_i t + \phi_i)] \text{ and } \phi_v(t) = \omega_i t + \phi_i$$

Interference

In the weak interference case

$$A_v(t) \cong A_c [1 + \rho \cos(\omega_i t + \phi_i)] \text{ and } \phi_v(t) \cong \rho \sin(\omega_i t + \phi_i)$$

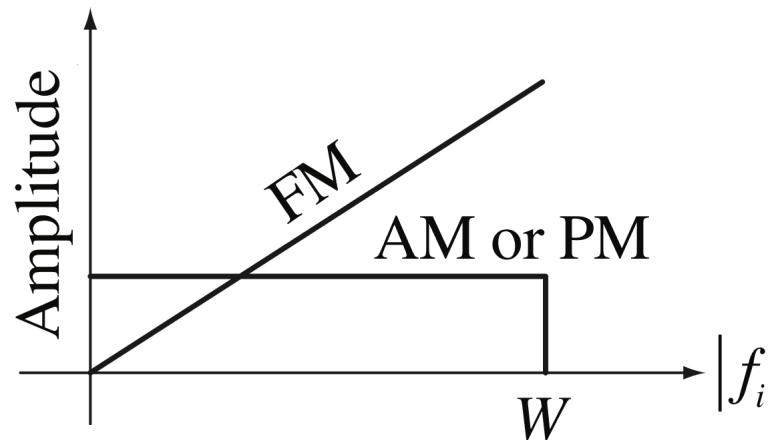
if we demodulate with an envelope, phase or frequency demodulator we get
(with $\phi_i = 0$)

Envelope Detector: $K_D [1 + \rho \cos(\omega_i t)]$

Phase Detector: $K_D \rho \sin(\omega_i t)$

Frequency Detector: $K_D \rho f_i \cos(\omega_i t)$

For AM or PM demodulation the demodulated signal strength is proportional to ρ . For FM demodulation the demodulated signal strength is proportional to the product of ρ and f_i .



Interference

The effects of interference on FM signals increases with frequency. So one way to reduce the effect is to lowpass filter the demodulated output. Of course this also lowpass filters the message, an undesirable outcome. To avoid the lowpass filtering effect on the message a technique called **preemphasis** is often used. The higher frequency parts of the message are preemphasized before transmission by passing them through a preemphasis filter with frequency response $H_{pe}(f)$ that amplifies the higher frequencies more than the lower frequencies. Then, after transmission and frequency demodulation, the demodulated signal is passed through a **deemphasis** filter whose frequency response is $H_{de}(f) = \frac{1}{H_{pe}(f)}$.

Interference

A typical deemphasis filter has a frequency response $H_{de}(f) = \frac{1}{1 + jf / B_{de}}$ in

which B_{de} is less than the cutoff frequency of the normal sharp-cutoff lowpass filter that determines the bandwidth. That makes the corresponding preemphasis filter have a frequency response $H_{pe}(f) = 1 + jf / B_{de}$.

Interference

A phenomenon that most people have experienced in receiving FM signals is the so-called **capture effect**. Suppose there are two FM stations, both transmitting in the same bandwidth and of approximately equal signal strength at the receiver. Their signal strengths will fluctuate some causing one to be stronger for a time and then the other. The stronger signal will "capture" the receiver for a short time and will dominate the demodulated signal. But then later the other signal will dominate and capture the receiver. The two stations switch back and forth and the listener hears a time-multiplexed version of both signals. To keep the math simple, assume we have one unmodulated carrier and one modulated carrier. This is exactly the "interfering sinusoid" case we analyzed earlier with the results

$$A_v(t) = A_c \sqrt{1 + \rho^2 + 2\rho \cos(\theta_i(t))} \text{ and } \phi_v(t) = \tan^{-1} \left(\frac{\rho \sin(\theta_i(t))}{1 + \rho \cos(\theta_i(t))} \right)$$

with $\theta_i(t) = \phi_i(t)$, the phase modulation of the interfering signal.

Interference

$$A_v(t) = A_c \sqrt{1 + \rho^2 + 2\rho \cos(\theta_i(t))} \text{ and } \phi_v(t) = \tan^{-1} \left(\frac{\rho \sin(\theta_i(t))}{1 + \rho \cos(\theta_i(t))} \right)$$

The demodulated signal is then

$$y_D(t) = \dot{\phi}_v(t) = \frac{d}{dt} \left(\tan^{-1} \left(\frac{\rho \sin(\phi_i(t))}{1 + \rho \cos(\phi_i(t))} \right) \right).$$

Using $\frac{d}{dz}(\tan^{-1}(z)) = \frac{1}{1+z^2}$ and the chain rule of differentiation,

$$y_D(t) = \frac{1}{1 + \left(\frac{\rho \sin(\phi_i(t))}{1 + \rho \cos(\phi_i(t))} \right)^2} \times \frac{[1 + \rho \cos(\phi_i(t))] \rho \cos(\phi_i(t)) \dot{\phi}(t) + \rho \sin(\phi_i(t)) \rho \sin(\phi_i(t)) \dot{\phi}(t)}{[1 + \rho \cos(\phi_i(t))]^2}$$

$$y_D(t) = \frac{\rho \cos(\phi_i(t)) + \rho^2}{[1 + \rho \cos(\phi_i(t))]^2 + \rho^2 \sin^2(\phi_i(t))} \dot{\phi}(t)$$

$$y_D(t) = \frac{\rho [\rho + \cos(\phi_i(t))]}{1 + \rho^2 + 2\rho \cos(\phi_i(t))} \dot{\phi}(t) = \alpha(\rho, \phi_i) \dot{\phi}(t)$$

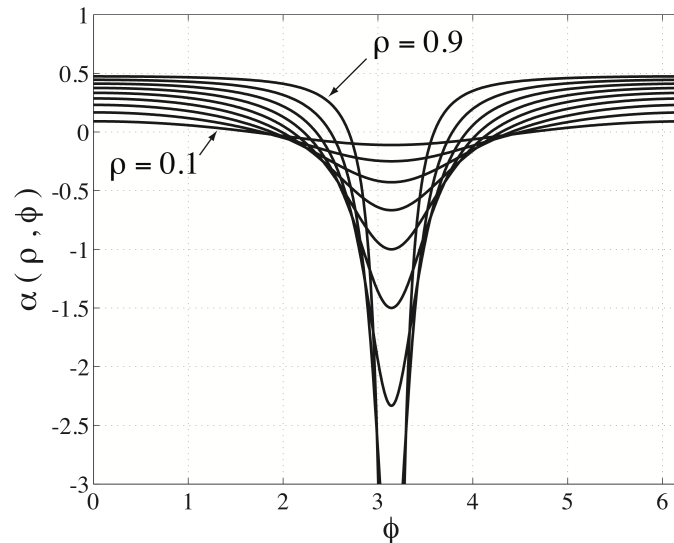
$$\text{where } \alpha(\rho, \phi_i) = \frac{\rho [\rho + \cos(\phi_i(t))]}{1 + \rho^2 + 2\rho \cos(\phi_i(t))}$$

Interference

$$y_D(t) = \alpha(\rho, \phi_i) \dot{\phi}(t)$$

The $\dot{\phi}(t)$ factor suggests that the interference may be intelligible if $\alpha(\rho, \phi_i)$ is relatively constant with time. If $\rho \gg 1$, then $\alpha(\rho, \phi_i) \cong 1$ and $y_D(t) \cong \dot{\phi}(t)$. But we wish to examine the case in which the two signals are approximately equal in strength, implying that $\rho \cong 1$.

$$\alpha(\rho, \phi_i) = \frac{\rho[\rho + \cos(\phi_i(t))]}{1 + \rho^2 + 2\rho \cos(\phi_i(t))} = \begin{cases} \rho / (1 + \rho) & , \phi_i = 0 + 2n\pi \\ \rho^2 / (1 + \rho^2) & , \phi_i = \pi / 2 + n\pi \\ -\rho / (1 - \rho) & , \phi_i = \pi + 2n\pi \end{cases} , n \text{ an integer}$$



Interference

$$y_D(t) = \alpha(\rho, \phi_i) \dot{\phi}(t) \quad , \quad \alpha(\rho, \phi_i) = \frac{\rho[\rho + \cos(\phi_i(t))]}{1 + \rho^2 + 2\rho \cos(\phi_i(t))}$$

As $\rho \rightarrow 1$, $\alpha \rightarrow 0.5$ and $y_D(t) \rightarrow 0.5\dot{\phi}(t)$.

For $\rho < 1$, the strength of the demodulated interference depends mostly on the peak-to-peak value of α

$$\alpha_{p-p} = \alpha(\rho, 0) - \alpha(\rho, \pi) = \frac{2\rho}{(1-\rho)^2}$$

The interference effect is small-to-negligible for $\rho < 0.7$ and the interference captures the demodulated output signal when $\rho > 0.7$.

