

## 7 Angle Modulation

Wednesday, August 22, 2012  
8:50 AM

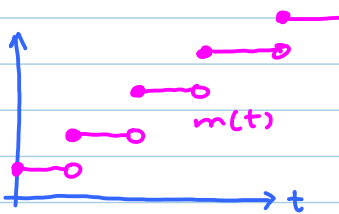
### 7.2 BW of $\alpha_{FM}(t)$

Recall that the instantaneous frequency of  $\alpha_{FM}(t)$  is given by

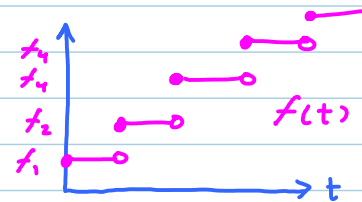
$$f(t) = f_c + k m(t) \quad \star$$

The BW analysis of FM is quite difficult and therefore we will consider a simplified version of  $\alpha_{FM}(t)$ .

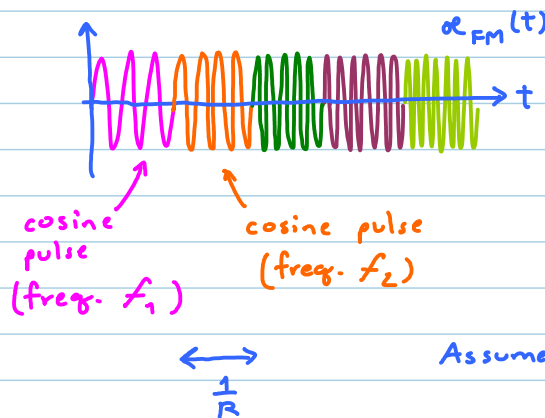
In this simplified version,  $m(t)$  is piecewise constant:



By  $\star$  above,  
this also means  
 $f(t)$  is  
piecewise constant



In class, we demonstrated this by considering  $\alpha_{FM}(t)$  constructed from 5 different tones.



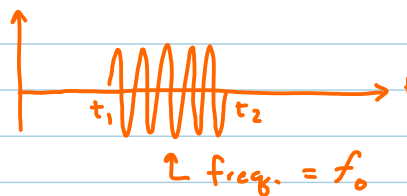
Assume that each tone lasts  $\frac{1}{R}$  sec.

Increasing the value of  $R$  means we take less time to complete the transmission.

Note that this  $\alpha_{FM}(t)$  is a sum of five cosine pulses.


We have seen in the HW that even though the Fourier transform of the cosine function gives two delta functions. The Fourier transform of a cosine pulse gives two sinc functions.

To see this, note that a cosine pulse is simply a product between cosine and rectangular pulse:

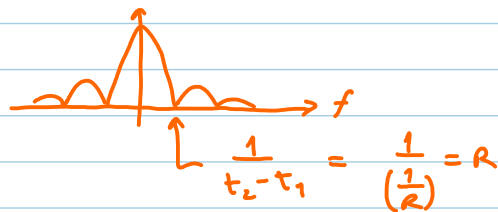


$$= \begin{cases} \cos(2\pi f_0 t), & t_1 \leq t \leq t_2 \\ 0, & \text{otherwise.} \end{cases}$$

$$= \cos(2\pi f_0 t) \times \begin{cases} 1, & t_1 \leq t \leq t_2 \\ 0, & \text{otherwise} \end{cases}$$

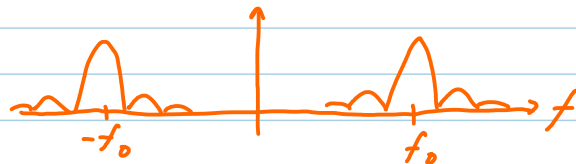
$$= \cos(2\pi f_0 t) \times \int_{t_1}^{t_2} 1 dt$$


The Fourier transform of the rectangular pulse centered around  $t=0$  is a sinc function. Here, the rectangular pulse is shifted to  $\frac{t_2+t_1}{2}$  and therefore there will be an extra factor of  $e^{-j2\pi f \frac{t_1+t_2}{2}}$  by the time-shift property. This does not change the magnitude of the sinc function. It is still



Recall that multiplication by  $\cos(2\pi f_0 t)$  will shift the spectrum of the rectangular pulse to  $\pm f_0$ .

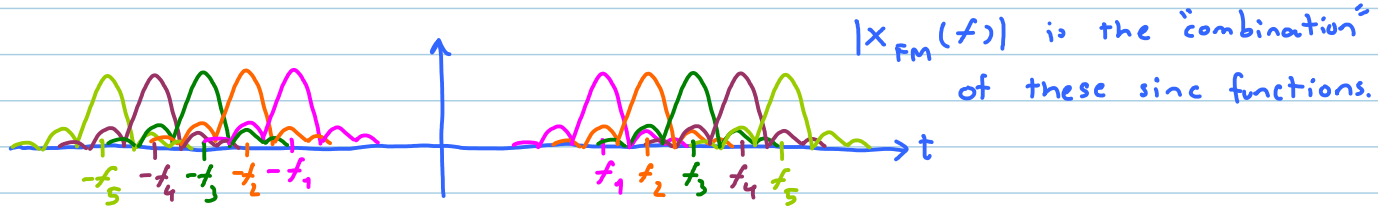
For the cosine pulse, this means its Fourier transform will be two sinc functions at  $\pm f_0$ .



Because  $x_{FM}(t)$  is a sum of cosine pulses, we can then see that its spectrum will be a sum of sinc functions centered at  $\pm$  frequencies of the pulses

The width of the sinc function is controlled by the time interval each tone takes in the time domain.

Large  $R \Rightarrow$  smaller  $\frac{1}{R} \Rightarrow$  large mainlobe of the sinc function.



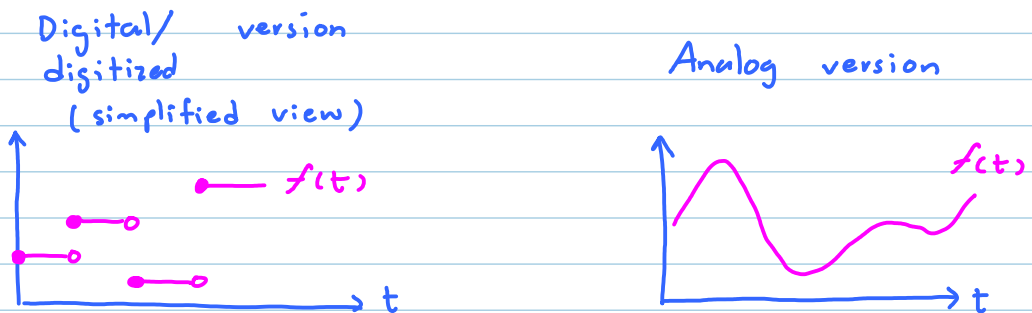
From the sketch above, we can see that the freq. content of  $X_{FM}(f)$  extends to  $\pm\infty$ . In other words, it is not band-limited.

However, one may say that the power of the sinc is mostly contained in its mainlobe. In which case, the "BW" becomes approximately

$$R + (f_5 - f_1) + R = (f_5 - f_1) + 2R$$

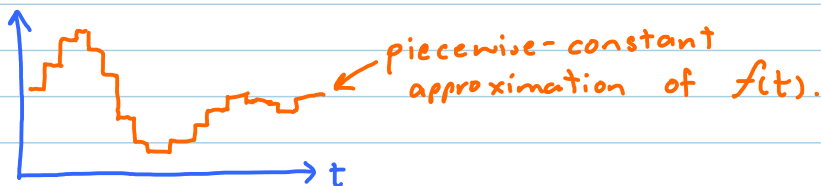
What we have been considering can be regarded as the digitized (or digital) version of FM.

The analog version is the one where  $m(t)$  and  $f(t)$  are not limited by the piecewise constant behavior.



This is where the analysis gets more difficult.

Our approach is to try to approximate the analog  $f(t)$  above by a piecewise constant function.



If the approximation is good, then we may be able to approximate the BW of  $x_{FM}(t)$  from the "modified  $x_{FM}(t)$ " which is constructed via the

approximation.

The value for each piece is determined by the value of  $f(t)$  around that time.

⇒ Sampling.