

HW -
 5.1 - 1, 2, 3) - instantaneous freq.
 5.2 - 1, 2, 3, - bandwidth

Also find full spectrum of $\phi_{em}(t)$
 Find BW by Carson's rule
 and by Bessel functions,
 with all components over 1%
 of the max component.

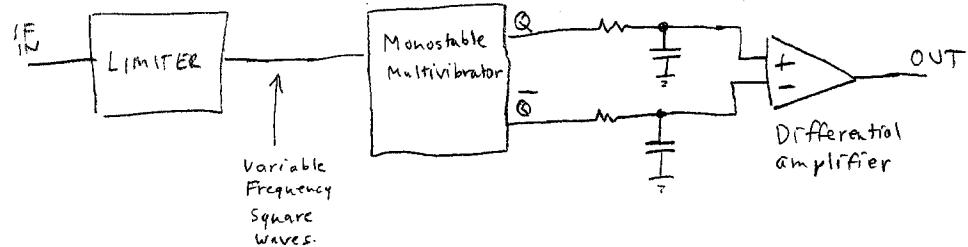
5.3 - 1, 2 - Also design (block diagram)
 the same specs using direct FM using
 a VCO that adjusts over the range
 of 10 - 14 MHz.

5.4 - 1, 2

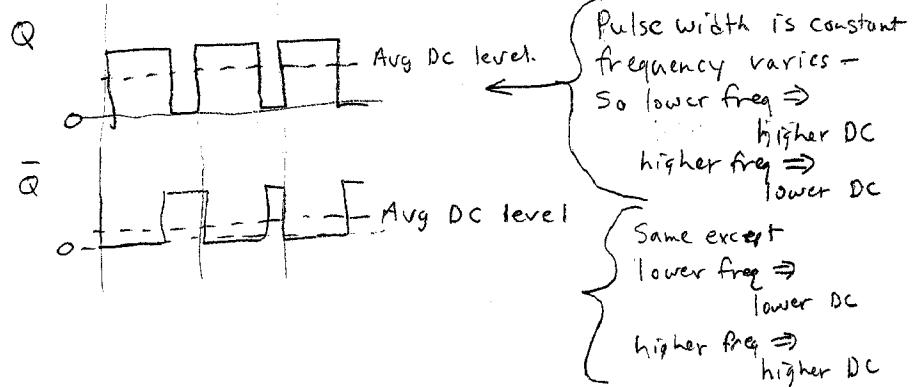
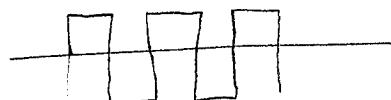
8A-1

"Pulse count" or "zero-crossing" detector

8A-2



LIMITER OUT



Advantages:

Good linearity, even when Af is high
 with respect to f_c . (high modulation)

Disadvantages:

Until recently, difficult to build at high frequencies.

Low output when Af is small with respect to f_c .

Phase-locked loop (most common method today)

Recall, how we synchronized the BFO (6A-3-6)
(see notes)

For FM detection, use the "error out" ($e_o(t)$)

It is the FM signal.

$$e_o(t) = \frac{AB}{2} \sin(\theta_i - \theta_o) \quad \leftarrow \text{No loop filter}$$

$\downarrow = \theta_e$

Instantaneous frequency of VCO - .

$$\omega_{VCO} = \omega_c + c e_o(t)$$

\uparrow
some constant.

if VCO output is $B \cos(\omega_c + \theta_o(t))$

then instantaneous freq is $\omega_c + \dot{\theta}_o(t)$

$$\text{so } \dot{\theta}_o(t) = c e_o(t)$$

With a general loop filter - .

$$e_o(t) = h(t) * \frac{1}{2} AB \sin(\theta_e)$$

\uparrow
convolution

The loop filter has a big effect
on FM demodulation

Applying some math (p237)

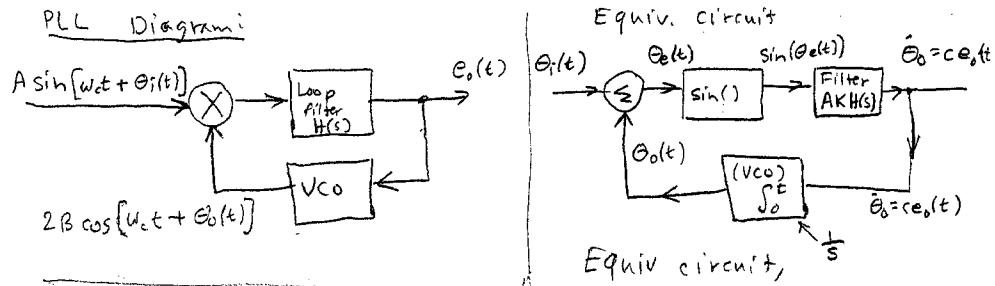
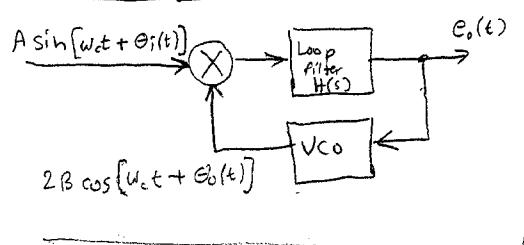
$$e_o(t) = \frac{1}{c} \dot{\theta}_o(t) \approx \frac{K_f}{c} m(t)$$

\uparrow

Shows that e_o tracks $m(t)$
so this demodulates FM.

PLL - Loop filter comments

PLL Diagram:



Equiv circuit,
thinking of it as
a general feedback circuit.

The VCO is like an integrator because - .

"Output" is frequency

"input" represents phase.

frequency = phase.

$\theta_i(t)$ is the "input" --- contained in the phase of the signal

$\theta_o(t)$ is the feedback --- contained in the phase of the VCO

$\dot{\theta}_o(t)$ is the frequency offset of the VCO.

$\theta_e(t)$ is the phase error

Book says "first order loop" -- means loop response = $\frac{1}{s}$

filter response = $\frac{1}{s}$
(no filter).

This can't track a signal.

"Second order loop" means filter response = $\frac{1}{s+a}$

$$\text{loop response} = \frac{\theta_o(t)}{\theta_i(t)} = \frac{FB}{1 + FB}$$

$FB = AKH(s) \cdot \frac{1}{s}$

like a non-inverting op-amp.

$$= \frac{AKH(s)}{1 + AKH(s)} = \frac{s}{s + AKH(s)} = \frac{s}{s + \frac{1}{s+A}} =$$

w subst.