Some basic definitions

1.
$$\frac{dBW}{}$$
 = 10 log₁₀ (*Power* (power expressed in watts))

So 1 watt of power $\equiv 0 \, dBW$

10 watt of power $\equiv 10 \, dBW$

2.
$$dBm = 10 \log_{10}(Power \text{ (power expressed in milliwatts))}$$

So 1 mW of power $\equiv 0 \, dBm$

 $10 \text{ mW of power} \equiv 10 \text{ dBm}$

 $1000 \text{ mW } (1 \text{ Watt}) \equiv 30 \text{ dBm}$

3.
$$dBV = 20 \log_{10}(voltage \text{ expressed in volts})$$

So 1 V \equiv 0 dBV

 $10 \text{ V} \equiv 20 \text{ dBV}$

 $1000V \equiv 60 \, dBV$

4. $dB = 20log_{10}$ (ratio of voltage or currents)

or $= 10 \log_{10}$ (ratio of power quantities)

5. Neper(Np) = \ln (ratio of voltage or currents)

$$Np = \ln\left(\frac{x_1}{x_2}\right)$$

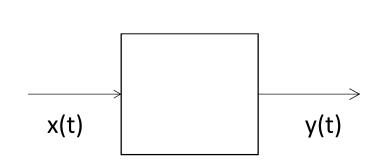
$$1dB = 10\log_{10}\left(\frac{x_1^2}{x_2^2}\right)$$

$$= 20\log_{10}\left(\frac{x_1}{x_2}\right) \Rightarrow \frac{x_1}{x_2} = 10^{(1/20)}$$

Again, $\frac{x_1}{x_2} = e^L$ where L is the Np value of $\frac{x_1}{x_2}$

Hence,
$$10^{(1/20)} = e^L \implies L = \ln(10^{(1/20)}) = 0.115 \text{ Np}$$

Non Linearity and Time Variance



This system is linear if for:

$$x_1(t) \rightarrow y_1(t)$$

$$x_2(t) \rightarrow y_2(t)$$

then we have:

$$a_1x_1(t) + a_2x_2(t) \rightarrow a_1y_1(t) + a_2y_2(t)$$

This system is time invariant if for

$$x(t) \rightarrow y(t)$$

we have

$$x(t-\tau) \to y(t-\tau)$$

Impulse Response of a Linear System

The impulse response for a linear system is obtained from:



The output y(t) for a general input x(t) can be expressed in terms of $h(t,\tau)$:

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

For a time - invarying and causal system:

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

In the frequency domain this gives:

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

with
$$H(j\omega) = \Im[h(t)]$$
, $X(j\omega) = \Im[x(t)]$, $Y(j\omega) = \Im[y(t)]$,

where $\Im[x]$ is the Fourier transform:

$$\Im[x(t)] = \int_{-\infty}^{\infty} x(t) \exp(-j\omega t) dt$$

Memoryless Non-Linear System

A system is called memoryless if its output does not depend on the past values of the input:

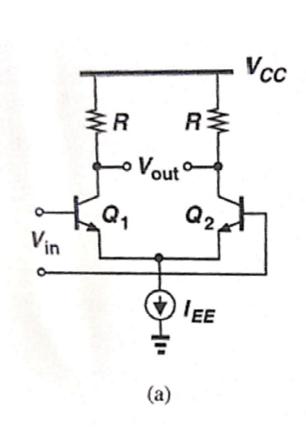
$$y(t) = f\big[x(t)\big]$$

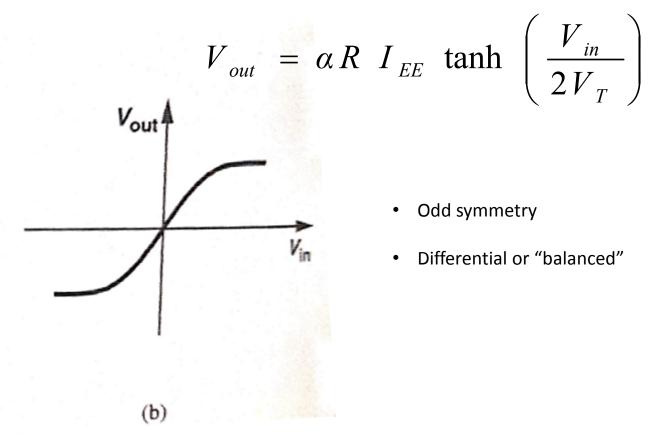
Example:

$$y(t) = \alpha_0 + \alpha_1 x(t) + \alpha_2 x(t)^2 + \alpha_3 x(t)^3$$

Example: A Differential Pair at low Frequency

This is a memoryless circuit if the transistor capacitances are neglected.





- Odd symmetry
- Differential or "balanced"

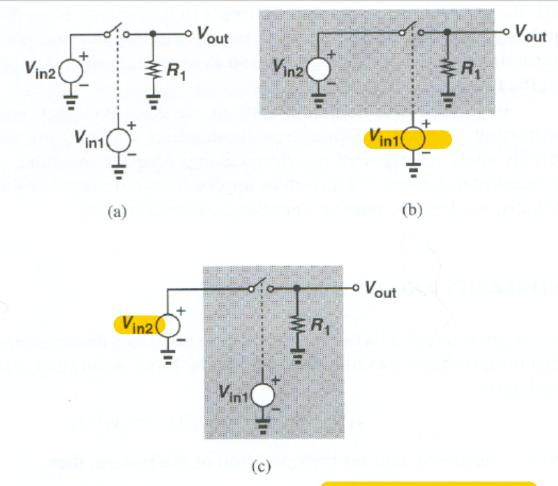
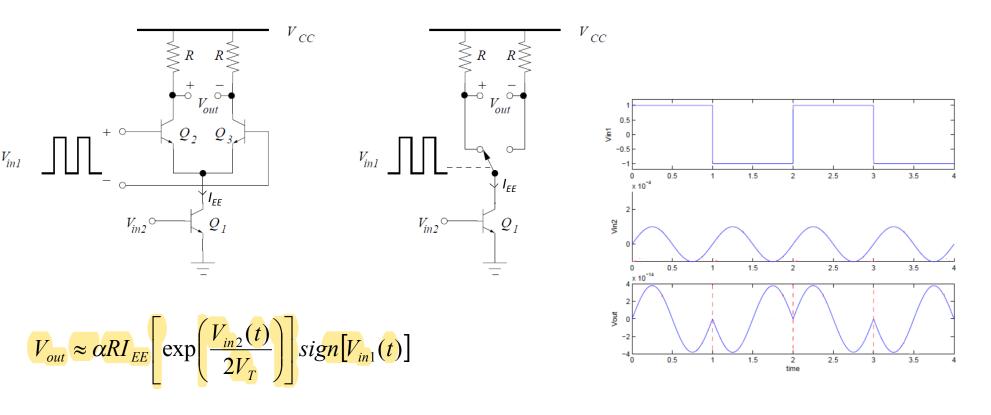


Figure 2.1 (a) Simple switching circuit, (b) nonlinear time-variant system.

Implementation of a Mixer with a Differential Pair



Effect of Non-Linearity: Harmonics

Consider a memoryless system given by:

$$y(t) = \alpha_0 + \alpha_1 x(t) + \alpha_2 x(t)^2 + \alpha_3 x(t)^3$$

For $x(t) = A_{in} \cos \omega t$ the output y(t) is:

$$y(t) = \alpha_0 + \frac{\alpha_2 A_{in}^2}{2} + \left(\alpha_1 A_{in} + \frac{3\alpha_3 A_{in}^3}{4}\right) \cos \omega t + \frac{\alpha_2 A_{in}^2}{2} \cos 2\omega t + \frac{\alpha_3 A_{in}^3}{4} \cos 3\omega t$$

Harmonics are generated.

At the fundamenta 1 frequency the output amplitude is:

$$A_{out} = \alpha_1 A_{in} + \frac{3\alpha_3 A_{in}^3}{4}$$

and the gain at the fundamenta 1 frequency varies (decreases since α_3 is usually negative) with the input signal amplitude:

$$G(A_{in}) = \alpha_1 + \frac{3\alpha_3 A_{in}^2}{4}$$

Gain Compression

The 1 dB compression point is defined as the point $(A_{in,1dB}, A_{out,1dB})$ where the gain $G(A_{in})$ drops by 1 dB:

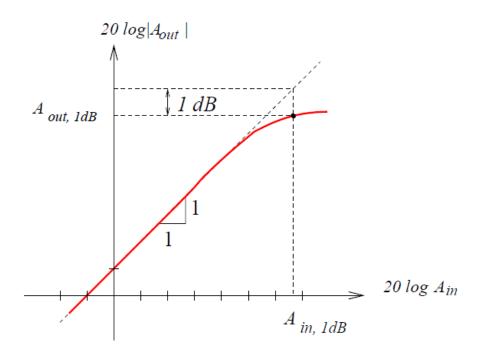
20
$$\log[G(A_{in,1dB})] = 20 \log[\alpha_1] - 1dB$$

$$\Rightarrow 20 \log \left(\alpha_1 + \frac{3}{4} \alpha_3 A_{in,1dB}^2 \right) = 20 \log \left[\alpha_1 \right] - 20 \log 10^{(1/20)} = 20 \log \left(\frac{\alpha_1}{10^{(1/20)}} \right)$$

$$\Rightarrow \alpha_1 + \frac{3}{4}\alpha_3 A_{in,1dB}^2 = \frac{\alpha_1}{10^{(1/20)}} \Rightarrow \frac{3}{4}\alpha_3 A_{in,1dB}^2 = -0.1087\alpha_1$$

$$\Rightarrow A_{in,1dB}^2 = -0.145 \frac{\alpha_1}{\alpha_3} [:: \alpha_1 \text{ and } \alpha_3 \text{ are of opposite signs}]$$

$$\Rightarrow A_{in,1dB} = \sqrt{0.145 \left| \frac{\alpha_1}{\alpha_3} \right|}$$

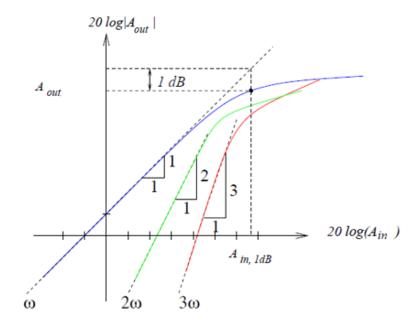


The input amplitude at the 1dB compression point is then: $A_{in,1dB} = \sqrt{0.145 \left| \frac{\alpha_1}{\alpha_3} \right|}$

Harmonics

$$A_{out}(2\omega)\big|_{dB} = 20\log\left[A_{out}(2\omega)\right] = 20\log\left[\frac{\alpha_2 A_{in}^2}{2}\right] = \frac{\alpha_2}{2}\Big|_{dB} + 2*A_{in}\Big|_{dB}$$

$$A_{out}(3\omega)\Big|_{dB} = 20\log[A_{out}(3\omega)] = 20\log\left[\frac{\alpha_3 A_{in}^3}{4}\right] = \frac{\alpha_3}{4}\Big|_{dB} + 3*A_{in}\Big|_{dB}$$



Desensitization and Blocking

If at the input we have a weak desired signal (A_1, ω_1) in the presence of a strong interferer (A_2, ω_2) with $\omega_1 \neq \omega_2$:

$$x(t) = A_1 \cos \omega_1 t + A_2 \cos \omega_2 t$$
the output y(t) is then: high
$$y(t) = \left(\alpha_1 A_1 + \frac{3}{4} \alpha_3 A_1^3 + \frac{3}{2} \alpha_3 A_1 A_2^2\right) \cos \omega_1 t + \dots$$

$$\approx \left(\alpha_1 + \frac{3}{2} \alpha_3 A_2^2\right) A_1 \cos \omega_1 t + \dots \text{ for } A_1 <<< A_2.$$

For α_3 negative, the strong interferer signal (A_2, ω_2) can completely cancel or block the desired weak signal (A_1, ω_1) :

$$\alpha_1 + \frac{3}{2}\alpha_3 A_2^2 = 0$$

A receiver may need to withstand blocking signal 60 to 70 dB greater.

Cross Modulation

If together with the weak desired signal (A_1, ω_1) we have a strong interferer which is modulated at the input (A_2, ω_2)

 $x(t) = A_1 \cos \omega_1 t + A_2 (1 + m \cos \omega_m t) \cos \omega_2 t$ the output y(t) is then:

$$y(t) = \alpha_1 + \frac{3}{2}\alpha_3 A_2^2 (1 + m\cos\omega_m t)^2 A_1\cos\omega_1 t + \dots$$

The desired signal (A_1, ω_1) contains at the output the modulation of the interferer signal (A_2, ω_2) .

Intermodulation

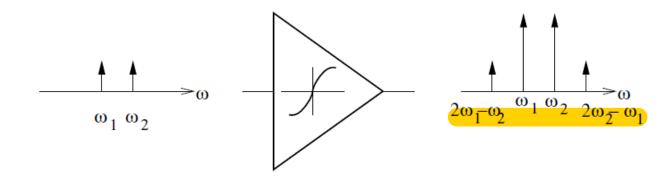
If at the input we have two signals (A_1, ω_1) and (A_2, ω_2)

$$x(t) = A_1 \cos \omega_1 t + A_2 \cos \omega_2 t$$

the output contains terms of frequencie s $n\omega_1 + m\omega_2$. Undesired terms appear at $2\omega_1 - \omega_2$ and $2\omega_2 - \omega_1$ frequencie s:

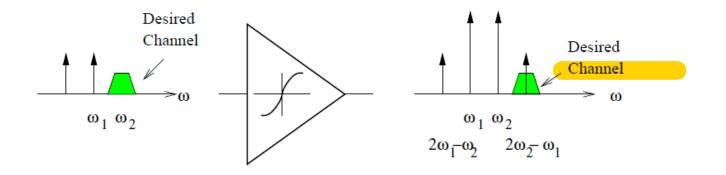
$$y = (\alpha_1 A_1 + ...) \cos \omega_1 t + (\alpha_1 A_2 + ...) \cos \omega_2 t$$

$$+ \frac{3}{4} \alpha_3 A_1^2 A_2 \cos(2\omega_1 - \omega_2) + \frac{3}{4} \alpha_3 A_1 A_2^2 \cos(2\omega_2 - \omega_1)$$

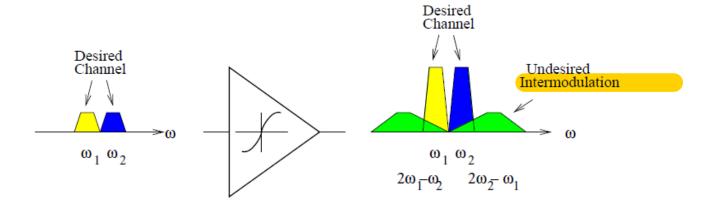


Undesirable Impacts of Intermodulation

Scenario 1: Two strong interferers perturbs a weak desired channel (LNA)



Scenario 2: Two channels blending (PA)



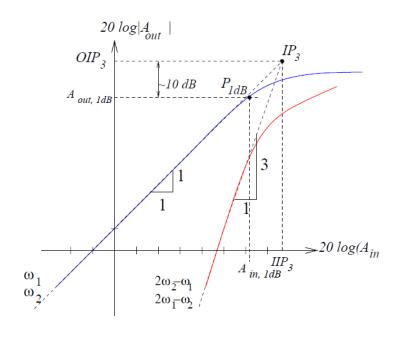
IP3: 3rd Order Intercept Point

That i/p power wher e the o/p power of fundamenta l = o/p power of 3rd order IM products

$$\alpha_1 A_{in,IP3} + \frac{3}{4} \alpha_3 A_{in,IP3}^3 = 0 \Rightarrow A_{in,IP3}^2 = -\frac{4}{3} \frac{\alpha_1}{\alpha_3} (:: \alpha_1 \text{ and } \alpha_3 \text{ are of opposite signs})$$

$$\Rightarrow A_{in,IP3} = \sqrt{\frac{4}{3} \left| \frac{\alpha_1}{\alpha_3} \right|}$$

$$\frac{A_{\text{out, IP3}}}{A_{\text{out,1dB}}} \approx \frac{A_{\text{in, IP3}}}{A_{\text{in,1dB}}} = \sqrt{\frac{4/3}{0.145}} \approx 10 \text{ } dB$$

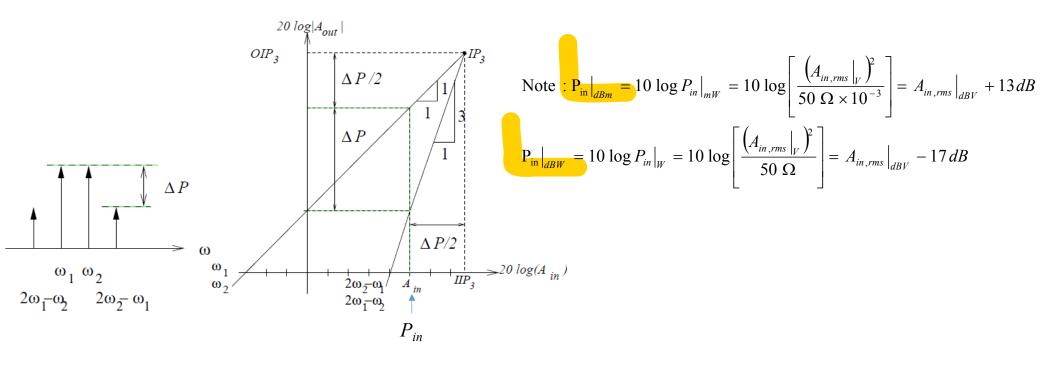


Quick IP3 Calculation

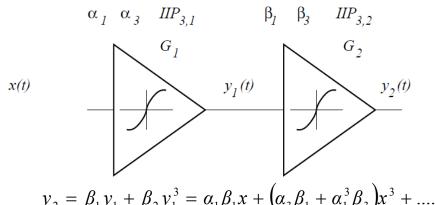
Let P be the IMD (intermodu lation distortion measured in dB or - dBc). From the graph below:

$$\frac{IIP \, 3}{dBV} = \Delta P \Big|_{dB} / 2 + A_{in} \Big|_{dBV}$$

$$IIP \ 3 \Big|_{\underline{dBm}} = \Delta P \Big|_{\underline{dB}} / 2 + P_{\underline{in}} \Big|_{\underline{dBm}}$$



IP3 of Cascaded Non-Linear Stages



$$y_1 = \alpha_1 x + \alpha_3 x^3$$
 $y_2 = \beta_1 y_1 + \beta_2 y_1^3 = \alpha_1 \beta_1 x + (\alpha_3 \beta_1 + \alpha_1^3 \beta_3) x^3 + \dots$

The total IP3 is then:

$$\frac{1}{A_{\text{in, IP3, total}}^{2}} = \frac{3}{4} \left| \frac{\left(\alpha_{3}\beta_{1} + \alpha_{1}^{3}\beta_{3}\right)}{\alpha_{1}\beta_{1}} \right| < \frac{3}{4} \frac{\left|\alpha_{3}\beta_{1}\right| + \left|\alpha_{1}^{3}\beta_{3}\right|}{\left|\alpha_{1}\beta_{1}\right|} = \frac{1}{A_{in, IP3, 1}^{2}} + \frac{\alpha_{1}^{2}}{A_{in, IP3, 2}^{2}}$$

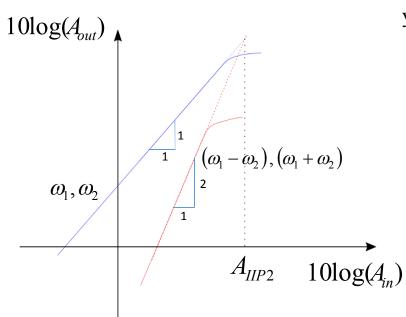
In terms of power P = $A_{in}^2/(2R)$ and power gain G = α^2 this is written (using also

$$P_{OIP3} = P_{IIP3} * G \text{ and } G_{total} = G_1G_2$$

$$\frac{1}{P_{IIP\,3,total}} = \frac{1}{P_{IIP\,3,1}} + \frac{G_1}{P_{IIP\,3,2}} \quad \text{or} \quad \frac{1}{P_{OIP\,3,total}} = \frac{1}{G_2 P_{OIP\,3,1}} + \frac{1}{P_{OIP\,3,2}}$$

 α_2 and β_2 are neglected because they contribute via higher harmonics which are filtered out.

IP2: 2nd Order Intercept Point



Let
$$x(t) = A\cos\omega_1 t + A\cos\omega_2 t$$

$$y(t) = \alpha_1 x(t) + \alpha_2 x^2(t)$$

$$= \alpha_1 A (\cos \omega_1 t + \cos \omega_2 t) + \alpha_2 A^2 \cos(\omega_1 + \omega_2) t$$

$$+ \alpha_2 A^2 \cos(\omega_1 - \omega_2) t + \frac{\alpha_2 A^2}{2} \cos(2\omega_1) t + \frac{\alpha_2 A^2}{2} \cos(2\omega_2) t$$

Power at these harmonics is half that at $(\omega_1 - \omega_2)$ or $(\omega_1 - \omega_2)$

$$\alpha_1 A_{IIP2} = \alpha_2 A_{IIP2}^2$$

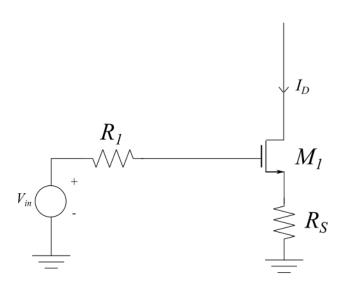
$$\Rightarrow A_{IIP2} = \frac{\alpha_1}{\alpha_2}$$

In practice, the $(\omega_1 - \omega_2)$ terms may originate from some other part of the circuit (feed through, leakage). In that case the eqn above is modified as follows,

$$\alpha_{1}A_{IIP2} = \underbrace{k}_{\substack{\text{coupling} \\ \text{factor}}} \alpha_{2}A_{IIP2}^{2}$$

$$\Rightarrow A_{IIP2} = \frac{\alpha_{1}}{k\alpha_{2}}$$

Linearity of degenerated CS stage



$$y = \alpha_1 x + \alpha_2 x^2 + \alpha_3 x^3$$

$$\alpha_1 = \frac{\partial y}{\partial x}\Big|_{x=0}, \alpha_2 = \frac{1}{2} \frac{\partial^2 y}{\partial x^2}\Big|_{x=0}, \alpha_3 = \frac{1}{6} \frac{\partial^3 y}{\partial x^3}\Big|_{x=0}$$

$$I_{D} = K(V_{GS} - V_{TH})^{2},$$

$$V_{GS} = V_{in} - R_{S}I_{D} \Rightarrow I_{D} = K(V_{in} - R_{S}I_{D} - V_{TH})^{2}$$

$$\Rightarrow \frac{\partial I_{D}}{\partial V_{in}} = \underbrace{2K(V_{in} - R_{S}I_{D} - V_{TH})}_{g_{m}} \left(1 - R_{S}\frac{\partial I_{D}}{\partial V_{in}}\right)$$

$$\Rightarrow \alpha_{I} = \frac{\partial I_{D}}{\partial V_{in}}\Big|_{V_{in} = 0} = \frac{g_{m}}{1 + g_{m}R_{S}}, \text{ where } g_{m} = \frac{\partial I_{D}}{\partial V_{GS}} = 2K(V_{GS} - V_{TH})$$

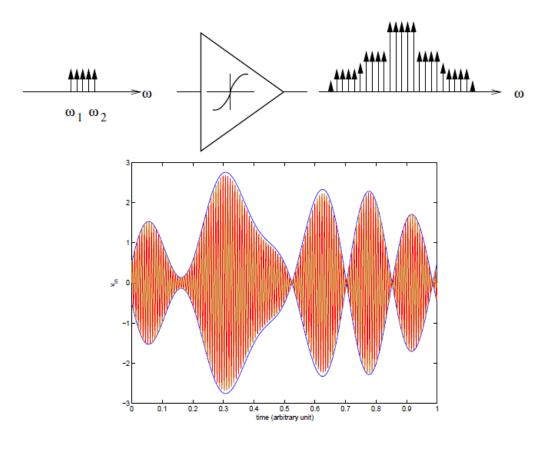
Similarly, we can find,

$$\frac{\alpha_{2}}{2} = \frac{1}{2} \frac{\partial^{2} I_{D}}{\partial V_{in}^{2}} \bigg|_{V_{in}=0} = \frac{K}{\left(1 + g_{m} R_{S}\right)^{2}}, \frac{\alpha_{3}}{\alpha_{3}} = \frac{1}{6} \frac{\partial^{2} I_{D}}{\partial V_{in}^{2}} \bigg|_{V_{in}=0} = \frac{-2K^{2} R_{S}}{\left(1 + g_{m} R_{S}\right)^{5}}$$

From these equations, IIP3 and IIP2 can be obtained as

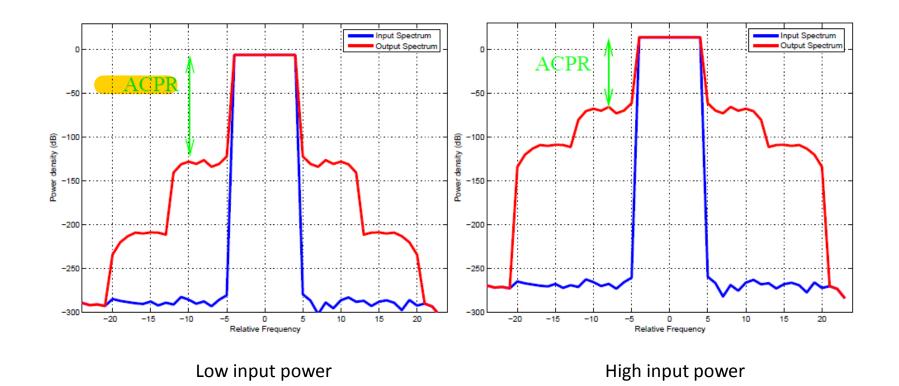
$$A_{IIP3} = \sqrt{\frac{4}{3} \left| \frac{\alpha_1}{\alpha_3} \right|} = \frac{(1 + g_m R_S)^2}{K} \sqrt{\frac{2}{3} \frac{g_m}{R_S}}, \quad A_{IIP2} = \frac{\alpha_1}{\alpha_2} = \frac{g_m}{K} (1 + g_m R_S)$$

Multitone excitation: Multisine



Input and Output Multisine

Spectral Regrowth



Intersymbol Interference

The problem:

Distortion of a signal can also arise from insufficient bandwidth. For digital signals this leads to intersymbol interference" (ISI) (see next slide).

The solutions:

- Equalization in the receiver
- Pulse shaping (Nyquist signaling)

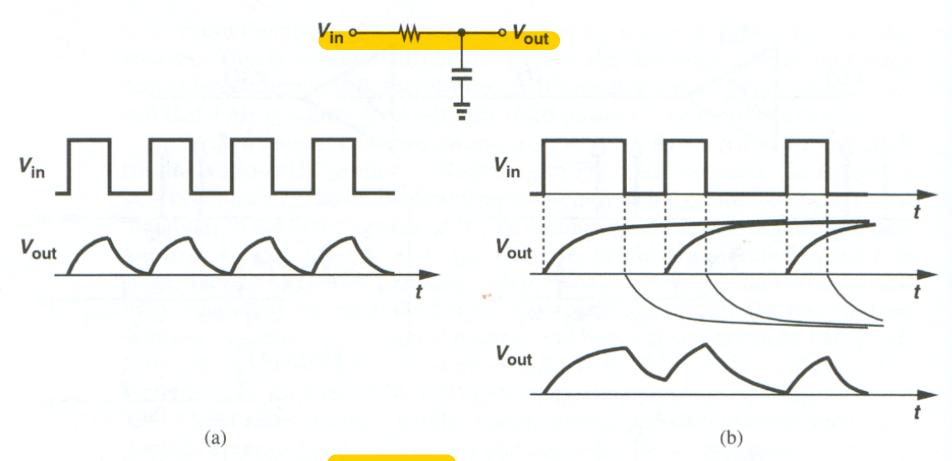


Figure 2.13 Response of a low-pass filter to (a) a periodic square wave, (b) a random sequence of ONEs and ZEROs.

Nyquist Signaling

In Nyquist signaling, each pulse p(t) is allowed to overlap with past and future pulses but the shape is selected such that the pulse is zero at the sampled time $t = kT_s$ so that no ISI exists:

$$p(kT_S) = \begin{cases} 1 & \text{if } k = 0 \\ 0 & \text{if } k \neq 0 \end{cases}$$

Example #1: sinc function (rectangul ar spectrum calls for complex filter):

0.8-0.6-0.4-0.2--0.2-

$$p(t) = \operatorname{sinc}(t - kT_{S})$$

Example #2 : raised - cosine pulse (smoother but trunca ted spectrum) :

$$p(t) = \frac{\sin(\pi t/T_S)}{\pi t/T_S} \frac{\cos(\pi \alpha t/T_S)}{1 - 4\alpha^2 t^2/T_S^2}$$

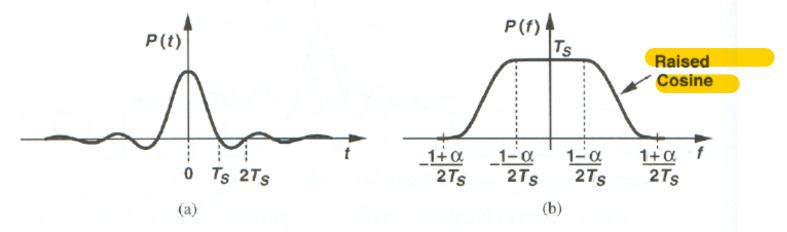


Figure 2.16 Raised-cosine pulse: (a) in time domain, (b) in frequency domain.

and

$$P(f) = T_S \quad 0 < |f| < \frac{1-\alpha}{2T_S}$$

$$= \frac{T_S}{2} \left[1 + \cos \frac{\pi T_S}{\alpha} \left(|f| - \frac{1-\alpha}{2T_S} \right) \right] \quad \frac{1-\alpha}{2T_S} < |f| < \frac{1+\alpha}{2T_S}$$

$$= 0 \quad |f| > \frac{1+\alpha}{2T_S},$$

$$(2.54)$$

Dynamic Nonlinearity (α 's change dynamically in the circuit)

$$V_{in} \overset{R_1}{\circ - } V_{out}$$

$$C_1 = C_0 \left(1 + \alpha V_{out} \right)$$

$$V_{out}(t) = \sum_{n=1}^{\infty} a_n \cos(n\omega_1 t + \theta_n) + \sum_{n=1}^{\infty} b_n \cos(n\omega_2 t + \phi_n)$$
$$+ \sum_{n=-\infty}^{\infty} \sum_{m=-\infty}^{\infty} c_{m,n} \cos(n\omega_1 t + m\omega_2 t + \phi_{n,m})$$

Assuming weak non linearity,

$$V_{out}(t) = a_1 \cos(\omega_1 t + \phi_1) + b_1 \cos(\omega_2 t + \phi_2) + c_1 \cos[(\omega_1 + \omega_2)t + \phi_3]$$

$$+ c_2 \cos[(\omega_1 - \omega_2)t + \phi_4]$$

Only these non linear

effects have been considered

$$R_1 C_0 (1 + \alpha V_{out}) \frac{dV_{out}}{dt} + V_{out} = V_{in}$$

Substitute here and equate LHS and RHS frequency wise to find a's, b's and c's.

This method is known as harmonic balance

Dynamic Nonlinearity

- The number of equations will equal the number of unknowns.
- Problem is that there are too many equations to solve.
- A better method is using Volterra series.