# **EECS 242: Analysis of Memoryless Weakly Non-Lineary Systems**

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### **Review of Linear Systems**

$$
y(t) = \int_{-\infty}^{\infty} h(t, \tau) x(\tau) d\tau
$$
  
\nLinear:  
\n
$$
\hat{x} = \alpha x_1(t) + \beta x_2(t)
$$
  
\n
$$
y(t) = \alpha \int_{-\infty}^{\infty} h(t, \tau) x_1(\tau) d\tau + \beta \int_{-\infty}^{\infty} h(t, \tau) x_2(\tau) d\tau
$$
  
\n
$$
y_1(t) = \alpha y_1(t) + \beta y_2(t) \quad \text{Linear}
$$

**Complete description of a general time-varying linear** system. Note output cannot have a DC offset!

### **Time-invariant Linear Systems**

- Time-invariant Linear Systems has  $h(t,\tau)=h(t-\tau)$
- Relative function of time rather than absolute
- The transfer function is "stationary"

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

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

convolution in time is product in frequency

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 $\infty$ 

### **Stable Systems**

■ Linear, time invariant (LTI) system cannot generate frequency content not present in input



#### **Memoryless Linear System**

 $N \cap$  "DC"

$$
y(t) = \alpha x(t)
$$
  
No Delay  
\nIf function is continuous at x<sub>0</sub>, then we can do a  
\nTaylor Series expansion about x<sub>0</sub>:  
\n
$$
x \longrightarrow f(x) \longrightarrow y = f(x)
$$
\n
$$
y_0 \longrightarrow y_0
$$
\

#### **Taylor Series Expansion**

**This expansion has a certain radius of** convergence. If we truncate the series, we can compute a bound on the error  $y - \hat{y}$ 

$$
S_o(x) = \hat{y}(x) - y_o = a_o + a_1 x + a_2 x^2 + a_3 x^3 + \dots
$$



Maximum excursion must be less than radius of convergence. Certainly the max  $A_k$  has to be smaller than the radius of convergence.

#### **Sinusoidal Exciation**

$$
y_m = \left(\sum_{n=1}^{N} A_n \cos \omega_n t\right)^m = \left(\sum_{n=1}^{N} \frac{A_n}{2} e^{j\omega_n t} + e^{-j\omega_n t}\right)^m
$$
  

$$
= \left(\sum_{n=-N}^{N} \frac{A_n}{2} e^{j\omega_n t}\right)^m \qquad A_o = 0
$$
  

$$
= \sum_{k_1=-N}^{N} \sum_{k_2=-N}^{N} \frac{A_{k_1} A_{k_2} ... A_{k_m}}{2^m} e^{j(\omega_{k_1} + \omega_{k_2} + ... + \omega_{km})t}
$$

### **General Mixing Product**

We have frequency components:  $\omega_{k_1} + \omega_{k_2} + ... + \omega_{k_m}$ where  $k_p$  ranges over 2*N* values Terms in summation:  $(2N)^n$ Example: Take m=3, N=2  $\begin{cases} \omega_1, \omega_1 \\ \omega_2, \omega_3 \end{cases}$   $(2N)^n = 4^3 = 64$ 64 Terms in summation!  $\omega \in \{-\omega_2, -\omega_1, \omega_1, \omega_2\}$  $HD<sub>3</sub>$  $IM<sub>3</sub>$ 64 Terms gain expression or compression

#### **Vector Frequency Notation**

Define 
$$
\vec{k} = (k_{-N}, ..., k_{-1}, k_1, ..., k_N)
$$

2*N*-vector where  $k_j$  denotes the number of times a particular frequency appears in a give summation:

$$
\omega_2 + \omega_1 + \omega_2
$$
\n
$$
\omega_1 + \omega_2 + \omega_2
$$
\n
$$
\omega_2 + \omega_2 + \omega_1
$$
\n
$$
\overline{k} = \begin{pmatrix} 0 & 0 & 0 \end{pmatrix} \qquad \begin{pmatrix} 2 & 0 & 0 \end{pmatrix}
$$
\n
$$
\overline{l} = \begin{pmatrix} 0 & 0 & 0 \end{pmatrix} \qquad \overline{l} = \begin{pmatrix} 2 & 0 & 0 \end{pmatrix}
$$

$$
\omega_1 + \omega_1 - \omega_2 \qquad \vec{k} = (1 \quad 0 \quad 2 \quad 0)
$$
  
\n
$$
\sum_{k=-N}^{k=+N} k_j = k_{-N} + ... + k_{-1} + k_1 + ... + k_N = m
$$
  
\n
$$
\sum_{k=-N}^{k=-N} k_j = k_{-N} + ... + k_{-1} + k_1 + ... + k_N = m
$$

## **Multinomial Coefficient**

For a fixed vector  $k_{o}$ , how many different sum vectors are there?

*m* frequencies can be summed *m*! different ways, but order is immaterial.

Each coefficient  $k_j$  can be ordered  $k_j!$  ways. Therefore, we have:

$$
(m; k) = \frac{m!}{(k_{-N})! \cdots (k_{-1})! (k_{+1})! \cdots (k_{N})!}
$$

Multinomial coefficient

## **Game of Cards (example)**

3 Cards: 3! or six ways to order cards



# **Making Conjugate Pairs**

Usually, we only care about a particular frequency mix generated by certain order non-linearity

Since our signal is real, each term has a complex conjugate. Hence, there is another:

$$
\vec{k} = (k_{N}, ..., k_{1}, k_{-1}, ..., k_{-N})
$$

reverse order

Taking the complex conjugates in pairs:

$$
2 \operatorname{Re}\left(e^{j(\omega_{k1} + \omega_{k2} + \dots + \omega_{km})t}\right) = 2 \cos\left(\omega_{k_1} + \omega_{k_2} + \dots + \omega_{k_m}\right)
$$

## **Amplitude of Mix**

Thus the amplitude of any particular frequency component is:

$$
\frac{2\times(m;\vec{k})}{2^m}=\frac{(m;\vec{k})}{2^{m-1}}
$$

 $Ex: IM<sub>3</sub>$  product generated by the cubic term  $-2$   $-1$   $1$   $0$  $IM_{3}$ : *m*=3  $(m;\vec{k}) = \frac{3!}{1!2!} = 3$   $2^{m-1} = 2^2 = 4$ *N*=2

Amplitude of  $IM<sub>3</sub>$  relative to fundamental:

$$
IM_3 = \frac{3}{4} \frac{a_3 s_i^3}{a_1 s_i} = \frac{3}{4} \frac{a_3}{a_1} s_i^2
$$

## **Gain Compression/Expansion**

■ How much gain compression occurs due to cubic and pentic  $(x^5)$  terms?

cubic:m=3, N=1 amp. of  $\frac{3}{4}$   $\frac{3}{4}$   $\frac{3}{8}$  appear This to appear anywhere fund: appear twice  $\begin{array}{ccc} \text{m-1} & 2^2 & 4 & \overline{3} & 4 & \overline{3} \\ \text{App. Gain:} & a_i + \frac{3}{4} a_3 s_i^2 & \text{Gain depends on signal amplitude} \end{array}$ anywhere pentic: m=5, N=1  $\phi_1^1 - \phi_1^1 + \phi_1^2 - \phi_1^2 + \omega_1 = \omega_1$  $\vec{k} = (2 \quad 3)$   $(m; \vec{k}) = \frac{5!}{3! \cdot 2!} = \frac{5 \cdot 4}{2} = 10$   $2^{m-1} = 2^4 = 16$  $1<sub>0</sub>$ 

App. Gain: 
$$
= a_1 + \frac{3}{4} a_3 s_i^2 + \frac{10}{16} a_5 s_i^4
$$

## **Who wins? Pentic or Cubic?**

R=
$$
\frac{\text{Gain Reduction due to Cubic}}{\text{Gain Reduction due to Pentic}} = \frac{\frac{3}{4}a_3s_i^2}{\frac{5}{8}a_5s_i^4}
$$
  
R= $\frac{1}{s_i^2} \frac{a_3}{a_5} \frac{6}{5}$ 

Take an exponential transfer function and consider gain compression:  $I_C = I_S e^{V_{BE}/V_T}$   $V_{BE} = V_O + v_S$ 

$$
I_C = \underbrace{I_S e \frac{V_Q}{V_T} e \frac{v_s}{V_T}}_{= I_Q (1 + \left(\frac{v_s}{V_T}\right) + \frac{\left(\frac{v_s}{V_T}\right)^2}{2!} + \dots)}
$$

### **Compression for Exp (BJT)**

$$
a_{3} = \frac{\left(\frac{1}{V_{T}}\right)^{3}}{3!}
$$
\n
$$
a_{5} = \frac{\left(\frac{1}{V_{T}}\right)^{5}}{5!}
$$
\n
$$
R = \frac{1}{s_{i}^{2}} \frac{a_{3}}{a_{5}} \frac{6}{5} = \frac{1}{s_{i}^{2}} \left(\frac{1}{V_{T}}\right)^{3} \frac{1}{3!} \frac{5!}{\left(\frac{1}{V_{T}}\right)^{5}} \frac{6}{5} = \frac{\frac{1}{s_{i}^{2}}}{\left(\frac{1}{V_{T}}\right)^{2}} \frac{5 \cdot 4 \cdot 6}{5} = 24 \left(\frac{V_{T}}{s_{i}}\right)^{2}
$$

When  $R=1$ , pentic non-linearity contributes equally to gain compression…

$$
R=1 \quad \Rightarrow \quad s_i = V_T \sqrt{24} \approx \underline{127mV}
$$

### **Summary of Distortion**

$$
x(t) \t f(x) \t f(x)
$$
  
f(x) = a<sub>1</sub>x + a<sub>2</sub>x<sup>2</sup> + a<sub>3</sub>x<sup>3</sup> + ...

 Due to non-linearity, y(t) has frequency components not present in input. For sinusoidal excitation by *N* tones, we *M* tones in output:

$$
M = \frac{(2N + m - 1)!}{m! (2N - 1)!}
$$

m: Order of highest term in non-linearity (Taylor exp.)

## **Amplitude of Frequency Mix**

Particular frequency mix  $\overline{k}$  has frequency

$$
f_{\overline{k}} = \sum_{\substack{j=-N \ j \neq 0}}^{N} k_j f_j = (k_1 - k_{-1}) f_1 + ... + (k_N - k_{-N}) f_N
$$
  

$$
\sum_{j=-N}^{+N} k_j = k_{-N} + ... + k_{-1} + k_1 + ... + k_N = m
$$
  

$$
\theta_{\overline{k}} = \sum \theta_j k_j = (k_1 - k_{-1}) \theta_1 + ... + (k_N - k_{-N}) \theta_N
$$

The amplitude of any particular frequency mix

$$
\left(\frac{\left(m;\vec{k}\right)}{2^{m-1}}\Bigg|A_1\Big|^{k_1+k_{-1}}\dots\Big|A_N\Big|^{k_N+k_{-N}}\right)\times\cos(\omega_k t+\theta_k)
$$
\namplitude

## **Harmonic Distortion**



For an input frequency  $\omega_j$ , each order non-linearity (power) produces a jth order harmonic in output

$$
HD_3 = \frac{1}{4} \frac{a_3}{a_1} s_i^2
$$

$$
= \frac{1}{4} \frac{a_3}{a_1^3} s_o^2
$$

2 dB increase for 1 dB signal increase

#### **Intermodulation**

 For a two-tone input to a memoryless non-linearity, output contains  $(2\omega_1 - \omega_2)$  &  $(2\omega_2 - \omega_1)$  due to cubic power and  $(\omega_2 - \omega_1)$  $\&$   $(\omega_2 + \omega_1)$  due to second order power.



## **Filtering Intermodulation**



- $\blacksquare$  IM<sub>2</sub> products fall at much lower (DC) and higher frequencies  $(2\omega_0)$ . These signals appear as interference to others, but can be attenuated by filtering
- $\blacksquare$  IM<sub>3</sub> products cannot be filtered for close tones.
- In a direct conversion receiver, IM2 is important due to DC.

### **IM/Harmonic Relations**

$$
IM_2 = \frac{a_2}{a_1} s_i = \frac{a_2}{a_1^2} s_o = HD_2(dB) + 6dB
$$
  

$$
IM_2 = 2HD_2
$$
  

$$
IM_2 \propto \text{Signal level}
$$

$$
IM_3 = \frac{3}{4} \frac{a_3}{a_1} s_i^2 = \frac{3}{4} \frac{a_3}{a_1^3} s_o^2 = HD_3(dB) + 9.5dB
$$
  

$$
IM_3 = 3HD_3
$$
  

$$
IM_3 \propto \text{ (Signal level)}^2
$$

## **Triple Beat**

**Triple Beat:** Apply three sine waves and observe effect of cubic non-linearity



$$
a_{k_p} = \frac{\left(\frac{1}{3}, \overline{k}_{TB}\right)}{2^{3-1}} s_i^3 = \frac{3! s_i^3}{4} = \frac{3}{2} s_i^3
$$

$$
TB = \frac{3}{2} \frac{a_3}{a_1} \frac{s_i^3}{s_i} = \frac{3}{2} \frac{a_3}{a_1} s_i^2
$$

### **Intercept Point**

**Intercept Point:** Apply a two tone input and plot output power and IM powers. The intercept point in the *extrapolated* signal power level which causes the distortion power to equal the fundamental power.



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## **Intercept/IM Calculations**

- Say an amplifier has an  $IIP3 = -10$  dBm. What is the amplifier signal/distortion (IM3) ratio if we drive it with -25 dBm?
	- Note:  $IM3 = 0$  dB at  $Pin = -10$  dBm
	- If we back-off by 15 dB, the IM3 improves at a rate of  $2:1$
	- For  $Pin = -25$  dBm (15 dB back-off), we have therefore IM3  $= 30dBc$

$$
IM_3 = \frac{3}{4} \frac{a_3}{a_1} V_{IIB}^2 = 1 \longrightarrow V_{IIB} = \sqrt{\frac{4}{3} \frac{a_1}{a_3}}
$$

intercept signal level

#### **Gain Compression and Expansion**

■ To regenerate the fundamental for the *N*<sup>th</sup> power, we need to sum *k* positive frequencies with *k*-1 negative frequencies, so  $N = 2k-1 \rightarrow N$  must be an odd power

$$
\underbrace{\omega_1 + \omega_1}_{k} + \dots + \omega_1}_{k} - \underbrace{\omega_1 - \omega_1 - \dots - \omega_1}_{k-1} = \omega_1
$$



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## **P1dB Compression Point**

■ An important specification for an amplifier is the 1dB compression point, or the input power required to lower the gain by 1dB

$$
G(s_i) = a_1 + \frac{3}{4}a_3s_i^2 = a_1 \left(1 + \frac{3}{4}\frac{a_3}{a_1}s_i^2\right)
$$
  
\n
$$
G_0 = \lim_{s_i \to 0} G(s_i) = a_1 \qquad \frac{G_0 \cdot 10^{-0.05}}{G_0} = \left(1 + \frac{3}{4}\frac{a_3}{a_1}s_i^2\right) = 10^{-0.05}
$$
  
\n
$$
s_i^2 = \left(1 - 10^{-0.05}\right) \frac{4}{3} \left| \frac{a_1}{a_3} \right| \qquad \text{Assume } a_3/a_1 < 0
$$
  
\n
$$
s_i = \sqrt{\left(1 - 10^{-0.05}\right) \frac{4}{3} \left| \frac{a_1}{a_3} \right|} = \sqrt{\left(1 - 10^{-0.05}\right)} V_{IIB} \qquad \text{About 9.6dB}
$$
  
\nlower than IIP3

## **Dynamic Range**

 $P_{-1}$ <sub>dB</sub> is a convenient "maximum" signal level which sets the upper bound on the amplifier "linear" regime. Note that at this power, the  $IM3 \sim 20$  dBc.

 The lower bound is set by the amplifier noise figure.



## **Blocking (or Jamming)**

- Blocker: Any large interfering signal
	- $P_{BL}$  = Blocking level. Interfering signal level in dBm which causes a +3dB drop in gain for small desired signal



## **Jamming Analysis**



## **Jamming Analysis (cont)**

Count the ways:  $(m; \vec{k}_B)$ 

$$
1\n\textrm{Imp little} = \frac{s_1 S_2^{-2} (3; \vec{k}_B)}{2^{m-1}} = \frac{3}{2} s_1 S_2^{-2}
$$
\n
$$
\vec{k}_B = (1 \quad 0 \quad 1 \quad 1)
$$

Append gain = 
$$
a_1 \left( 1 + \frac{3}{2} \frac{a_3}{a_1} S_2^2 \right)
$$

\ngain w/o blocker

\ngain reduction or expansion due to  $\alpha S_{BLOCKER}$ 

## **Blocking Power ~ P<sub>1dB</sub>**

$$
20\log\left(1+\frac{3}{2}\frac{a_3}{a_1}S_2^2\right) = -3dB
$$
  

$$
S_2 = \sqrt{\frac{4}{3}\left|\frac{a_1}{a_3}\right|} = \frac{\sqrt{0.293}}{\sqrt{2}}
$$
  

$$
P_{BL} = P_{IIB} - 8.3dB
$$
  

$$
= P_{-1dB} + 1dB
$$

#### **Effect of Feedback on Disto**



$$
S_o = a_1 S_{\varepsilon} + a_2 S_{\varepsilon}^2 + a_3 S_{\varepsilon}^3 + \cdots
$$
  
=  $a_1 (s_i - f g) + a_2 (s_i - f g)^2 + a_3 (s_i - f g)^3 + \cdots$   
=  $b_1 s_i + b_2 s_i^2 + b_3 s_i^3 + \cdots$ 

### **New Non-Linear Coefficients**

$$
b_1 = \frac{a_1}{1 + a_1 f} \qquad b_1 \approx \frac{1}{f} (T \rightarrow \infty)
$$
  
Loop gain T

$$
b_2 = \frac{a_2}{\left(1+T\right)^3}
$$

For high loop gain, the distortion is very small. Even though the gain drops, the distortion drops with loop gain since b2 drops with a higher power.

$$
b_3 = \frac{a_3(1+T) - 2a_2^2 f}{(1+T)^5}
$$

The cubic term has two components, the original cubic and a second order interaction term. If an amplifier does not have cubic, FB creates it (MOS with  $R_s$ )

#### **Series Inversion**



#### **Series Cascade**



### **IIP2 Cascade**

 The cascade IIP2 is reduced due to the gain of the first stage:

$$
\frac{c_2}{c_1} = \frac{b_1 a_2 + b_2 a_1^2}{b_1 a_1} = \frac{a_2}{a_1} + \frac{b_2}{b_1} a_1
$$

$$
\frac{1}{IIP2} = \frac{1}{IIP2^A} + \frac{a_1}{IIP2^B}
$$

- To calculate the overall IIP2, simply input refer the second stage IIP2 by the *voltage gain* of the first stage.
- The overall IIP2 is a parallel combination of the first and second stage.

### **IIP3 Cascade**

■ Using the same approach, we can calculate the IIP3 of a cascade. To simplify the result, neglect the effect of second order interaction:

$$
\frac{c_3}{c_1} = \frac{b_1a_3 + b_2a_1a_22 + b_3a_1^3}{b_aa_1} = \left(\frac{a_3}{a_1} + \frac{b_3}{b_1}a_1^2 + \frac{b_2}{b_1}2a_2\right)
$$

$$
\frac{1}{IIP3^2} = \frac{1}{IIP3_A^2} + \frac{a_1^2}{IIP3_B^2}
$$

Input refer the IIP3 of the second stage by the *power gain* of the first stage.

#### **References**

- **UCB EECS 242 Class Notes**, Robert G. Meyer, Spring 1995
- **Sinusoidal analysis and modeling of weakly nonlinear circuits : with application to nonlinear interference effects**, Donald D. Weiner, John F. Spina. New York : Van Nostrand Reinhold, c1980.