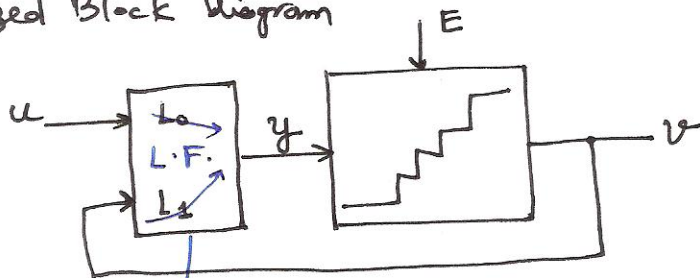


Describing Function Analysis

Generalized Block Diagram



$$Y(z) = L_0(z)U(z) + L_1(z)V(z)$$

where,

$$L_0(z) = \frac{STF(z)}{NTF(z)}$$

$$L_1(z) = \frac{NTF(z) - 1}{NTF(z)}$$

for the whole modulator:

$$\Rightarrow V(z) = Y(z) + E(z)$$

$$= L_0(z)U(z) + L_1(z)V(z) + E(z)$$

$$\Rightarrow V(z) = \underbrace{\frac{L_0(z)}{1 - L_1(z)}}_{STF} \cdot U(z) + \underbrace{\frac{L_1(z)}{1 - L_1(z)}}_{NTF} \cdot E(z)$$

$$= STF(z) \cdot U(z) + NTF(z) \cdot E(z)$$

So far a linear model. But how to model the quantizer so as to understand the effects of its non-linearity.

- Overload (or saturation) of the quantizer causes instability:
 - ↳ when input exceeds the range of the quantizer, the output of the quantizer doesn't change at all.
 - ↳ feedback breaks down!

Definition of stability:
for LTI systems,

Bounded input \rightarrow Bounded output (BIBO)
 $\Rightarrow \sum_n |h(n)| < \infty$

• So far we have assumed Linear Model of noise.

Loop-filter
↳ 2 inputs & 1 output

* for second-order DSM
 $L_0(z) = \frac{1}{(1-z^{-1})^2}$
 $L_1(z) = \frac{-z^{-1}}{1-z^{-1}} - \frac{z^{-1}}{(1-z^{-1})^2}$

* for special cases:
 $L_0(z) = L_1(z) = L(z)$
 Ex. first-order DSM

Definition for a DSM (non-linear feedback system):

* If the state variables become unbounded for a bounded input, the system is unstable.

• State variables $\hat{=}$ integrator outputs in the loop filter.

• for higher-order modulators

• for small input amplitudes \rightarrow system is stable

• when the amplitude (u) is increased, the output of the quantizer either gets clipped at the maximum or the minimum output, or wildly oscillates between the two.

• the output of the loop filter, (y_n) hits ∞ or $-\infty$ when unstable.

• Non-linear system with feedback:

↳ can't use linear feedback theory directly.

↳ sampled non-linear feedback system $\hat{=}$ DSM

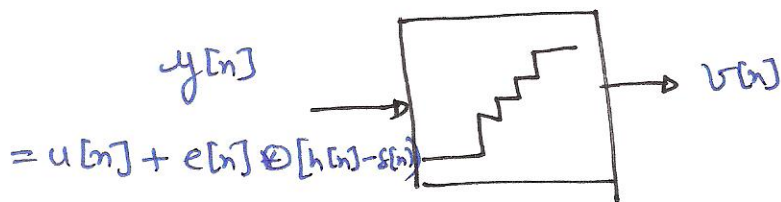
↳ simulation?

↳ need to develop some intuition.

* Approximate the non-linear (NL) system by a linear one.

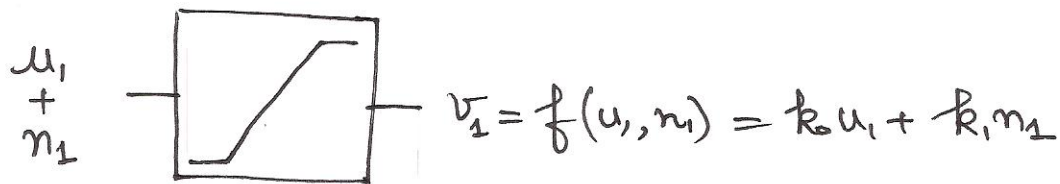
↳ Additive white quantization noise was a good working model.

↳ Quantizer overload is the primary reason for instability.

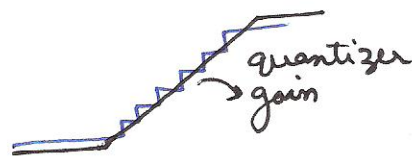


the sum of input and ~~step~~ quantization noise, shaped by the filter ($h[n] - s[n]$) is quantized by the quantizer.

Describing function Method (by Asdalan & Paulos)



Approximating the quantizer by a two input linear system.



$\Rightarrow v_1 = k_0 u_1 + k_1 n_1$

the quantizer output is a linear function of the input (u_1) and quantization noise (n_1).

find the gains k_0 and k_1

$v_1[n] = k_0 u_1[n] + k_1 n_1[n]$ for N samples

$$\begin{matrix} [u_1 & n_1] \\ N \times 2 \end{matrix} \begin{matrix} \begin{bmatrix} k_0 \\ k_1 \end{bmatrix} \\ 2 \times 2 \end{matrix} = \begin{matrix} v_1 \\ N \times 2 \end{matrix}$$

let $A = [u_1 \ n_1]$ be $R^{N \times 2}$, then

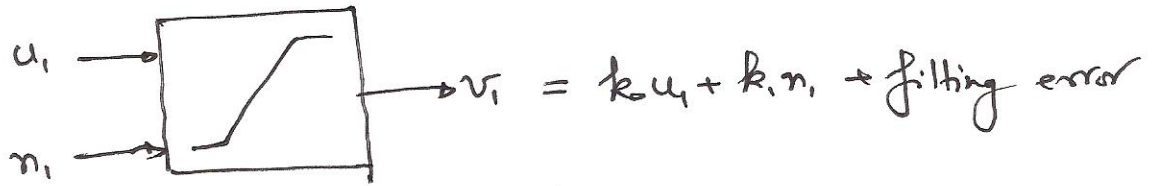
$$A \begin{bmatrix} k_0 \\ k_1 \end{bmatrix} = v_1$$

$$\Rightarrow A^T A \begin{bmatrix} k_0 \\ k_1 \end{bmatrix} = A^T v_1$$

$$\Rightarrow \boxed{\begin{bmatrix} \hat{k}_0 \\ \hat{k}_1 \end{bmatrix} = (A^T A)^{-1} A^T v_1} \leftarrow \text{optimal curve fitting}$$

curve fitting using

error $\|v - [A] \begin{bmatrix} \hat{k}_0 \\ \hat{k}_1 \end{bmatrix}\|_2 = 0$ if the system is linear.

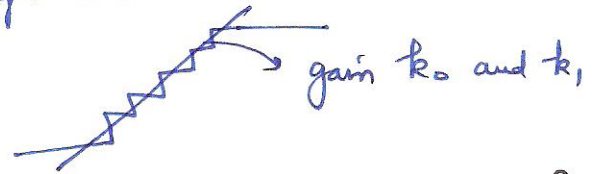


⇒ least square fit between the inputs and the output.
 ↳ k_0 & k_1 will be different for different inputs.

See Matlab file: "Describing_function_Analysis.m".

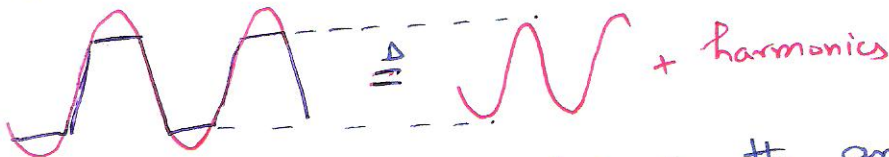
for the toolbox $\Delta = \text{LSB} = 2$ & step-height = 2

* when the filtering error is small



(a) Input 'u' is ~~not~~ within the quantizer range (no clipping):
 $k_0 = k_1 = 1$.

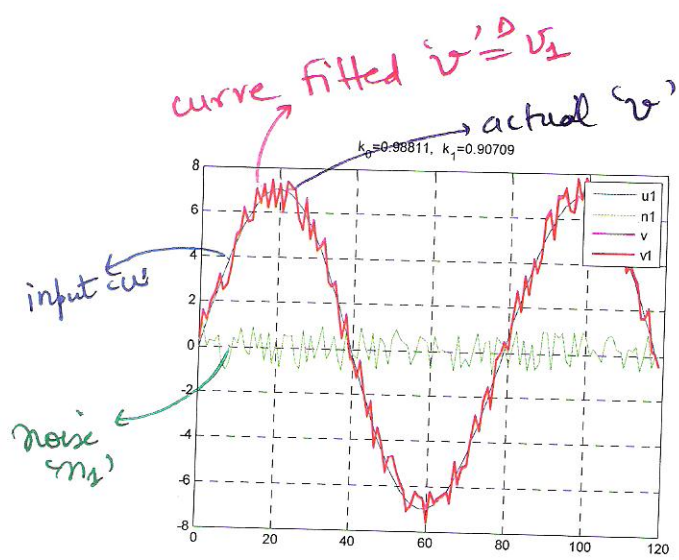
(b) Quantizer output clips:
 gains k_0 and k_1 are very different



* the gain for the input (u) is the amplitude of the first harmonic of the clipped output. ($k_0 \approx 0.9$) for $A = 1.2 \cdot FS$

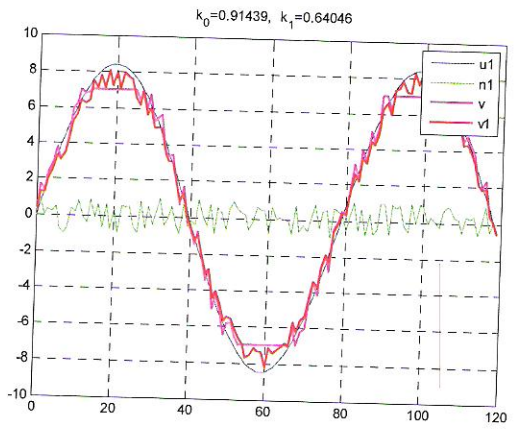
* since the noise is riding on the inputs, during the clipping the noise is eliminated in the output ⇒ noise gain is zero during clipping
 ⇒ on average k_I is much smaller than k_0 .

⇒ for $A = 1.2 \cdot FS$
 $k_1 \approx 0.7$



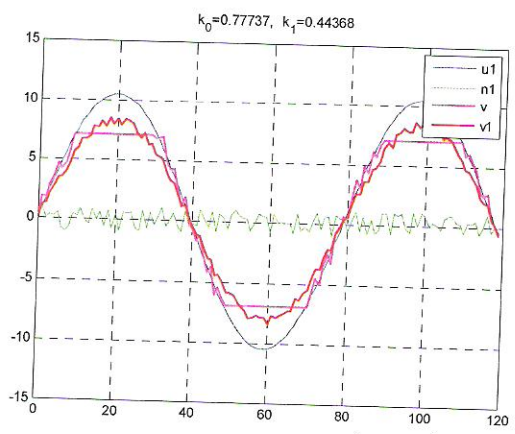
input amplitude $A = FS$ ← full scale range of the quantizer

$k_0 = 0.99, k_1 = 0.91$



$A = 1.2 FS$

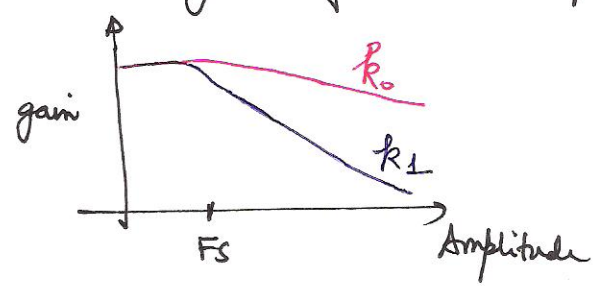
$k_0 = 0.91, k_1 = 0.64$



$A = 1.5 FS$

$k_0 = 0.78, k_1 = 0.44$

With overloading,
 \Rightarrow gain for noise, k_1 , falls much faster than the gain for the input signal u , i.e. k_0 .

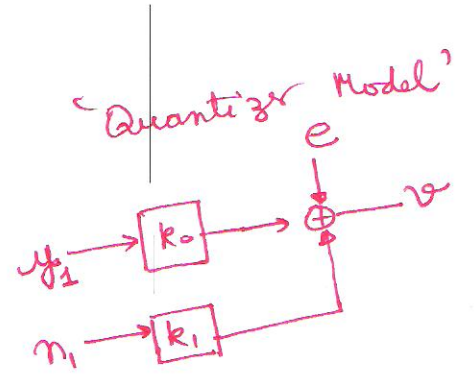
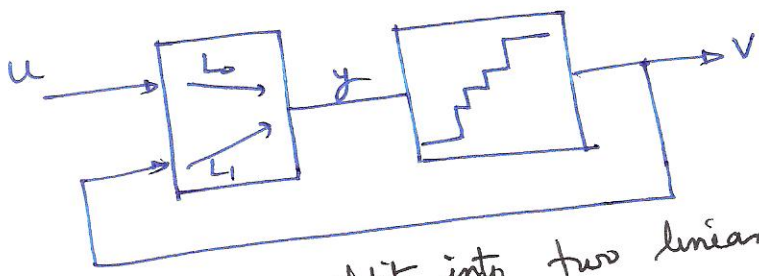


⇒ Signal and noise gains through the 'quantizer'
 $\Rightarrow 1$, when there is no overload

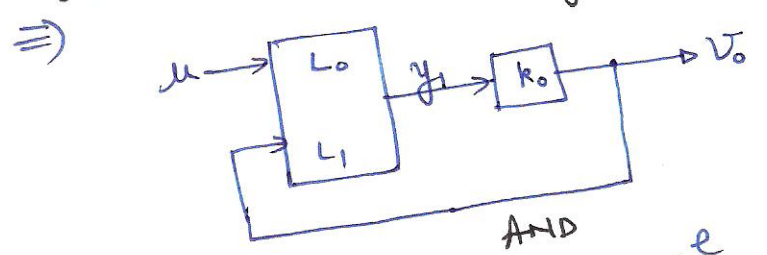
• ~~During~~ with overload,
 Signal gain: $k_0 < 1$
 Noise gain: $k_1 \ll k_0 < 1$

⇒ Noise gain (k_1) depends upon the signal 'u'.
 when signal amplitude is high \Rightarrow noise gain (k_1) is smaller.

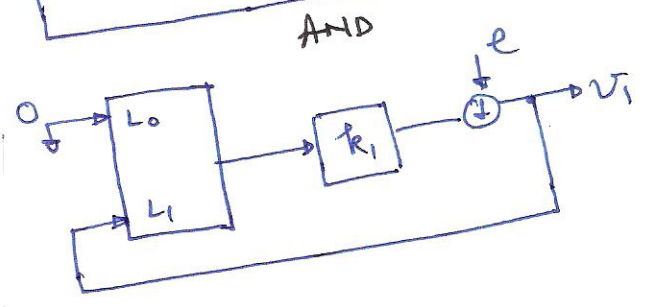
⇒ Introduce gain into the DSM model



The modulator splits into two linear systems.



'Signal loop'

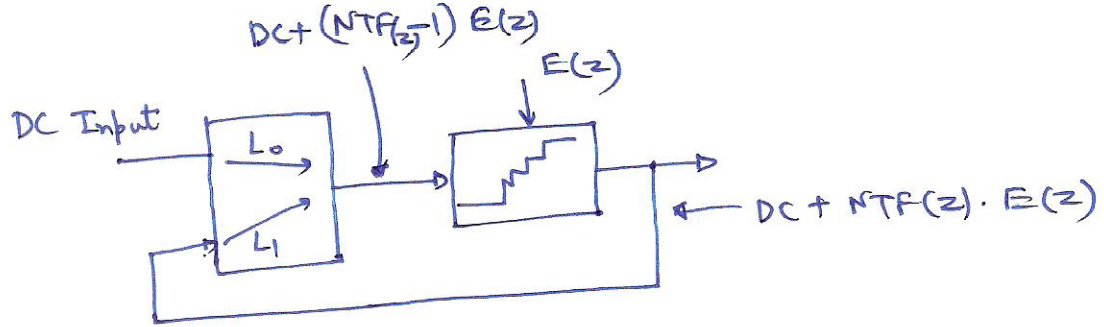


'noise loop'

from these 'linear' loops.

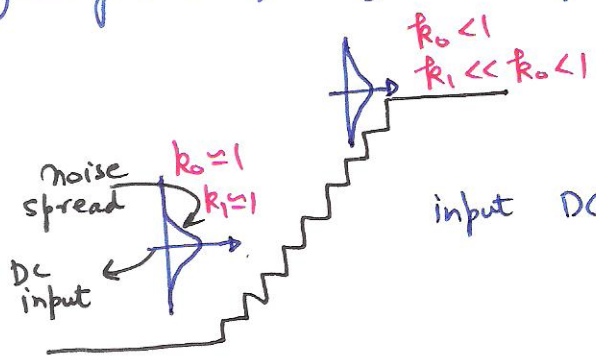
$$NTF'(z) = \frac{V_1(z)}{E(z)} = \frac{L_1(z)}{1 - k_1 L_1(z)} = \frac{NTF(z)}{k_1 + (1 - k_1) NTF(z)}$$

$$STF'(z) = \frac{V_0(z)}{U(z)} = \frac{\cancel{L_0(z)}}{\cancel{1 - k_0 L_1(z)}} = \frac{L_0(z)}{1 - k_0 L_1(z)}$$



• Gain of the quantizer is dependent upon the input level.

* Here \$\ll\$ is of the order of \$1/4^{th}\$



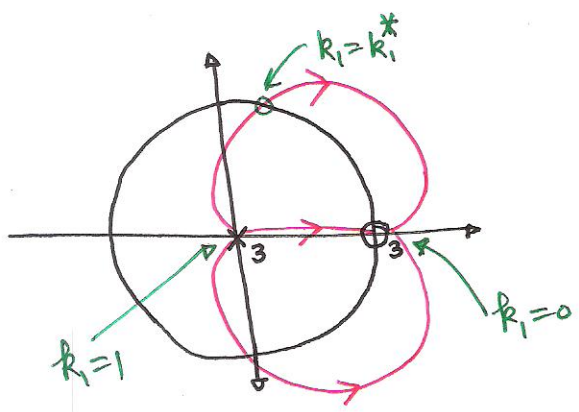
input DC level up \$\Rightarrow k_1 \downarrow \Rightarrow\$ noise gain' decreases
 \$\Rightarrow\$ poles of the NTF(z) start moving towards the unit-circle.

$$NTF(z) = \frac{NTF(z)}{k_1 + (1-k_1)NTF(z)} = \frac{L_1(z)}{1 - k_1 L_1(z)}$$

* At what point the poles move out of the unit circle?
 \$\hookrightarrow\$ Plot root locus of NTF(z) w.r.t. '\$k_1\$'

Example:

$$NTF(z) = (1-z^{-1})^3 \Rightarrow \begin{cases} 3 \text{ zeros at } z=1 \\ 3 \text{ poles at } z=0 \end{cases}$$



at \$k_1=0\$, poles move to \$z=1\$
 \$\Rightarrow\$ complete instability.

At \$k_1=k_1^*\$, the NTF becomes unstable.

• poles leave the unit circle only for NTF order 3 or higher.

• \$rlocus(NTF)\$ works ok in MATLAB for plotting root-locus even though the system description (model) is different.
 slightly

Higher-order Modulators:

- Signal dependent stability
- Gain falls off as the quantizer overloads (or saturates)
 - ↳ when the input levels start to become larger, we hit a point where the modulator becomes unstable.

∴ Noise at the quantizer input = $(NTF(z)-1)E(z)$

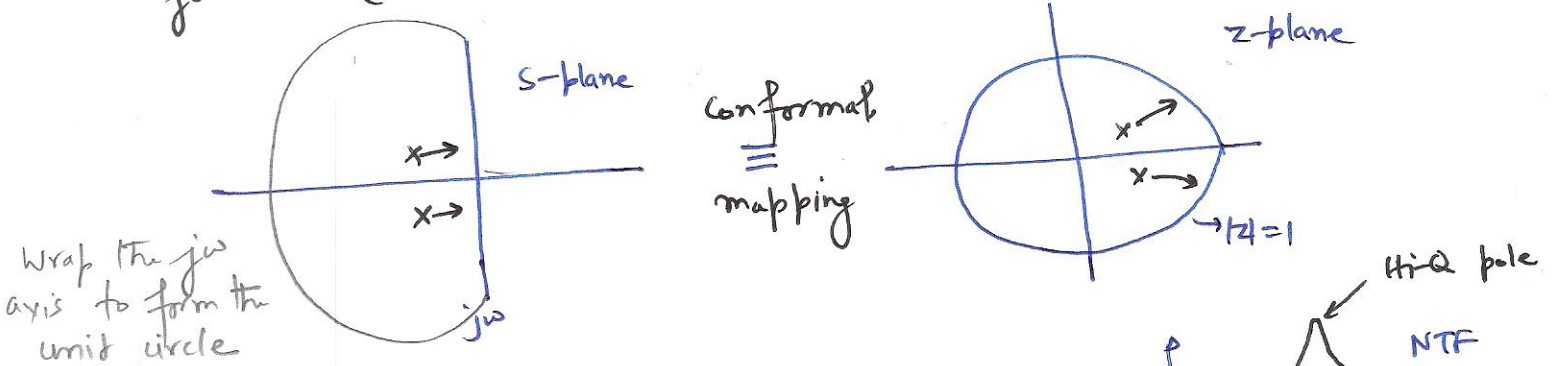
↳ noise mean square value = $\frac{\Delta^2}{12} [\|h[n]\|_2 - 1]$

↳ $\sum_n h^2[n]$

By Parseval's Theorem

* The modulator 'destabilization' sequence:

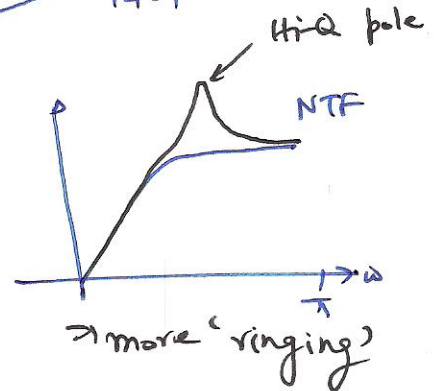
- ① As the input DC level is increased, the gain for the noise falls.
- ② As, $\langle R_1 \rangle$ drops, poles of the NTF start moving towards the $j\omega$ axis (unit-circle) \Rightarrow Q-factor of the poles increases.



- ③ As the Q factor of the poles increases
 - ↳ length of $h[n]$ increases

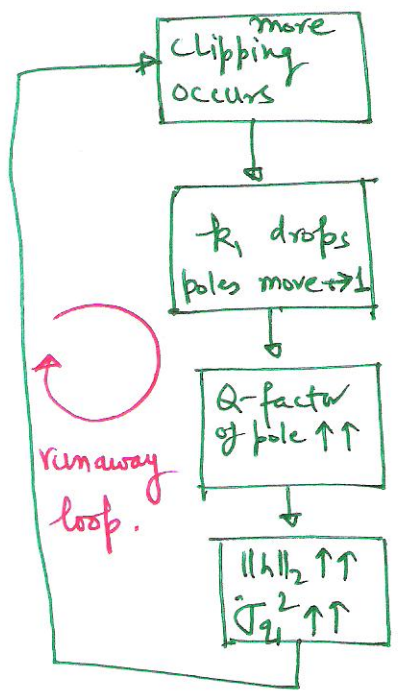
↳ $\sigma_q^2 = \frac{\Delta^2}{12} (\|h\|_2 - 1)$ increases

\Rightarrow quantization noise variance at Y' increases



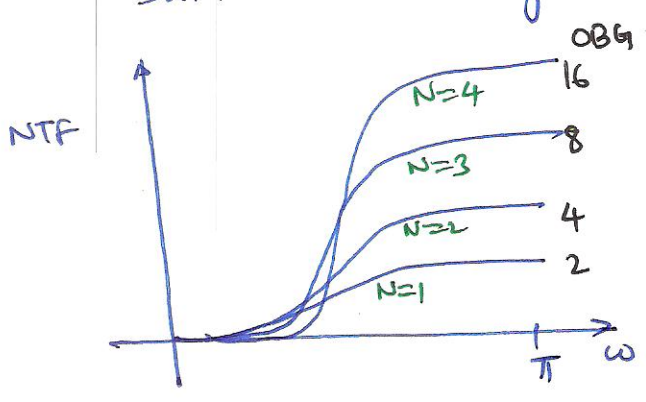
④ Clipping of the output waveform occurs more frequently \Rightarrow 'k' drops further
 \Rightarrow poles move faster towards unit circle
 runaway loop \Rightarrow noise variance increases further

⑤ System becomes unstable
 \hookrightarrow State variable become unbounded.
 \hookrightarrow SNR drops down



• Very rapid runaway to instability.

\Rightarrow Modulator can not be stable for the signal level upto the Saturation level of the quantizer. (MSA \Rightarrow maximum stable amplitude)



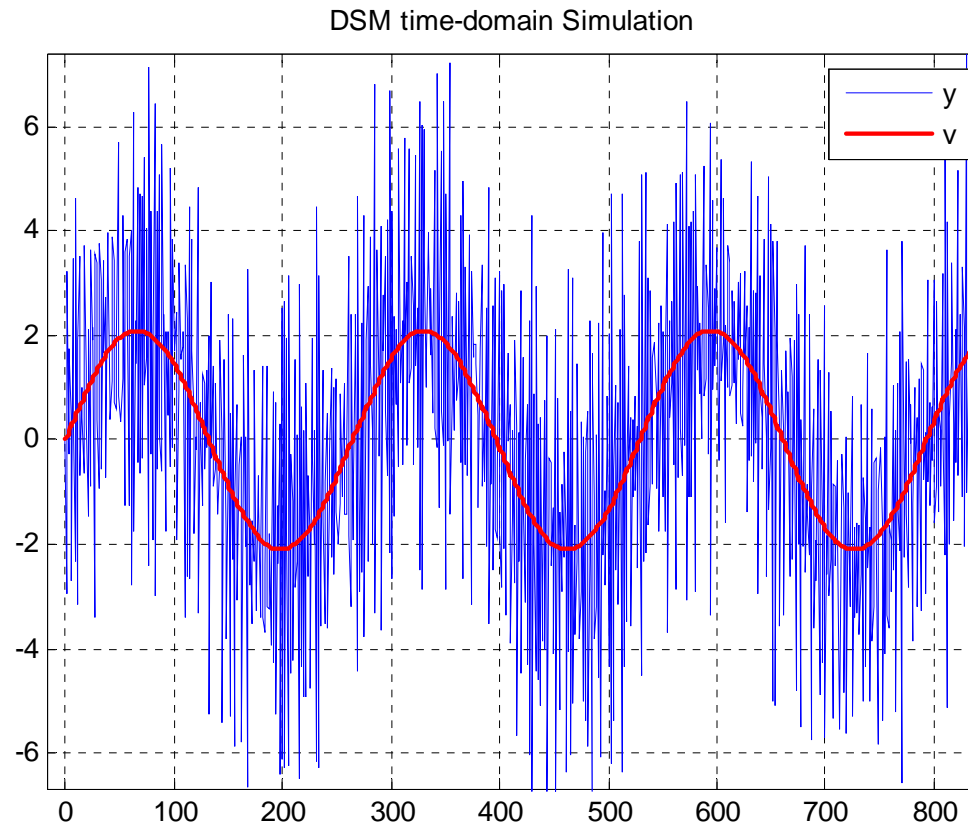
As OBG increases, the variance of the noise is higher.
 \Rightarrow Quantizer overloads more often
 \Rightarrow MSA starts decreasing rapidly.
 • $OBG = H(-1) = ||h||_1$

ECE 697 Delta-Sigma Converters Design

Lecture#10 Slides

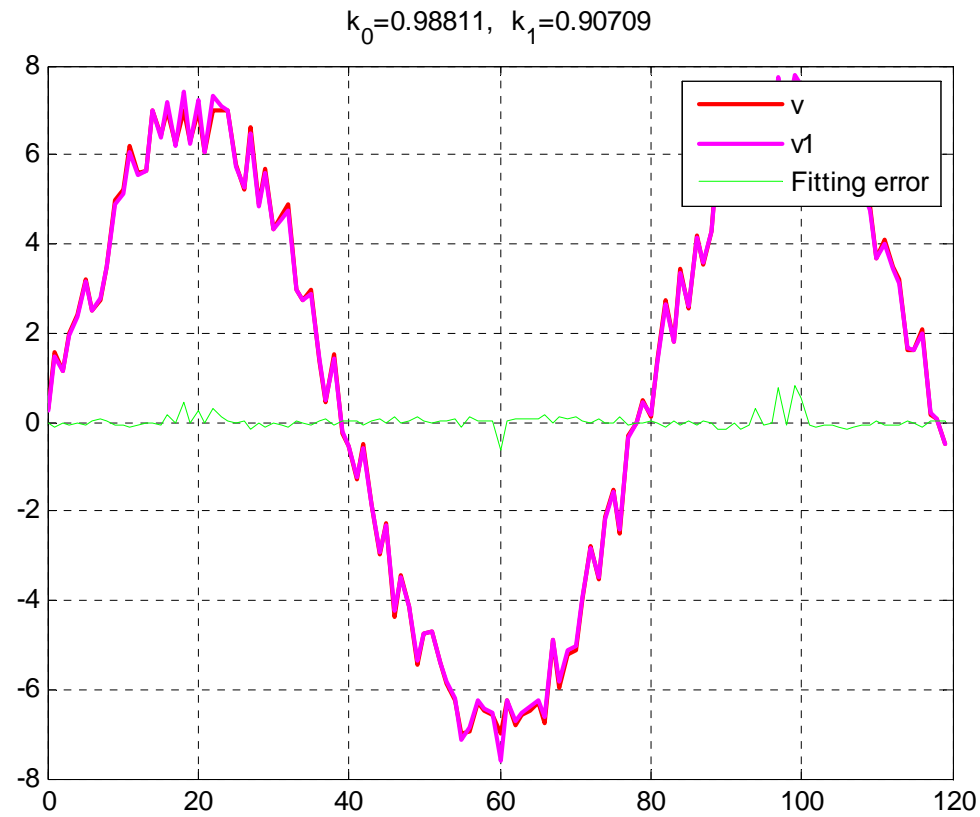
Vishal Saxena
(vishalsaxena@u.boisestate.edu)

Accumulated Noise at the Quantizer Input



File:Third_Order_DSM_1.m

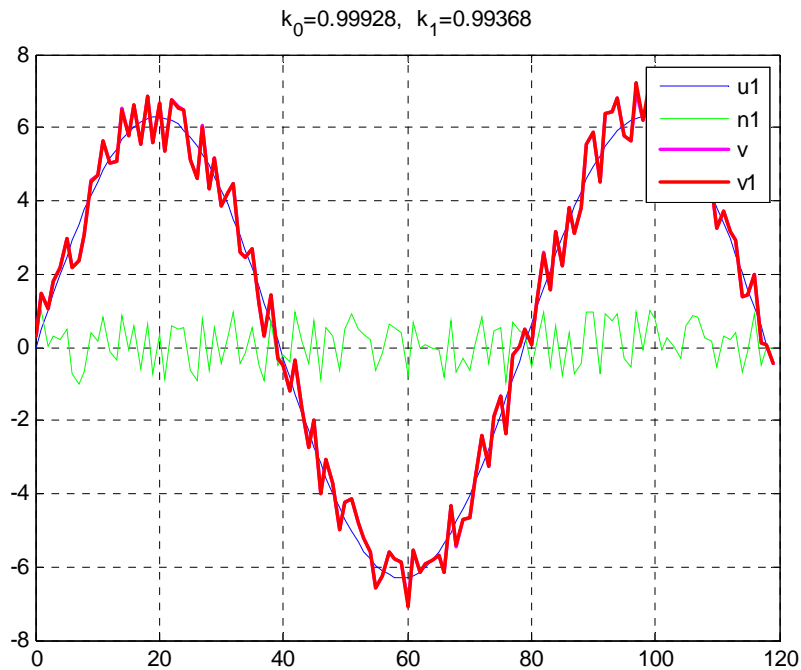
Describing Function Analysis: Curve Fitting



File: Describing_Function_Analysis.m

Describing Function Analysis

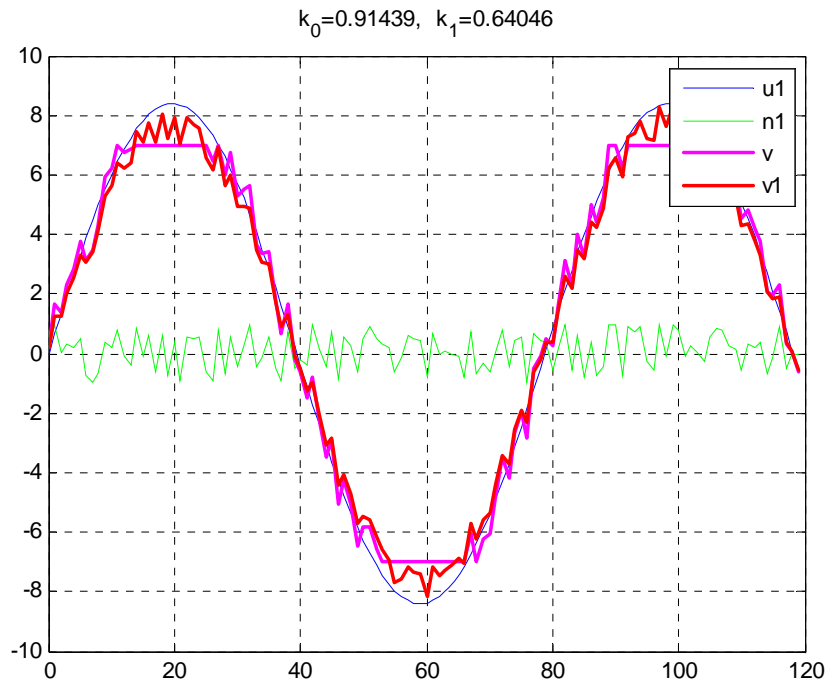
- Amplitude = 0.9 FS
- $k_0 \approx 1, k_1 \approx 1.$



File: Describing_Function_Analysis.m

Describing Function Analysis contd.

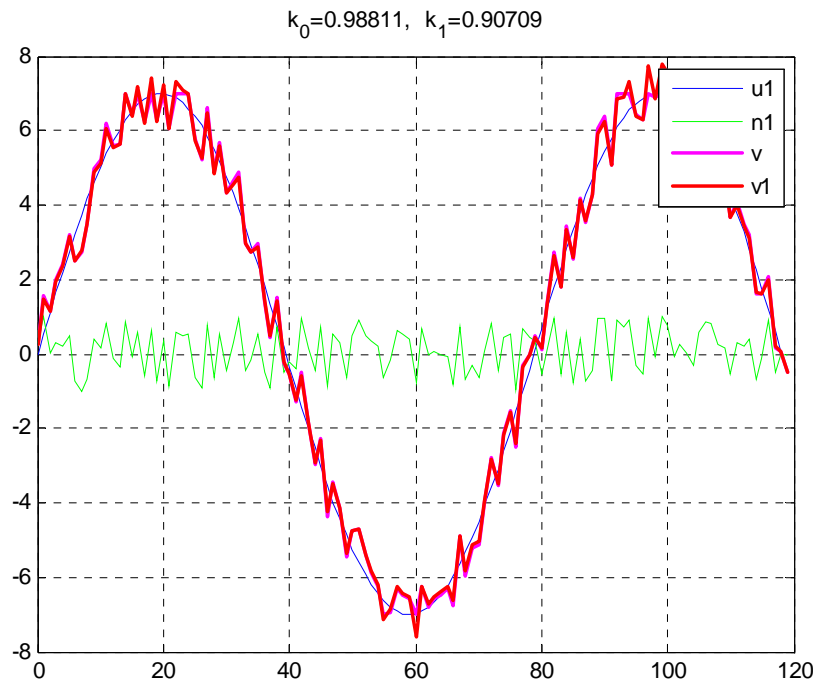
- Amplitude = 1.2 FS
- $k_0 = 0.91, k_1 = 0.64$.



File: Describing_Function_Analysis.m

Describing Function Analysis contd.

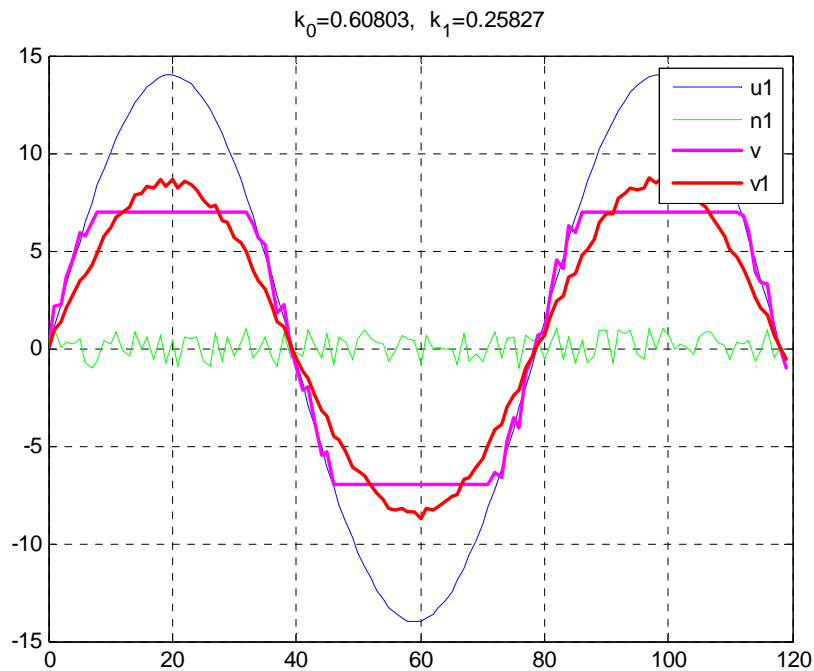
- Amplitude = 1.5 FS
- $k_0 = 0.78, k_1 = 0.44.$



File: Describing_Function_Analysis.m

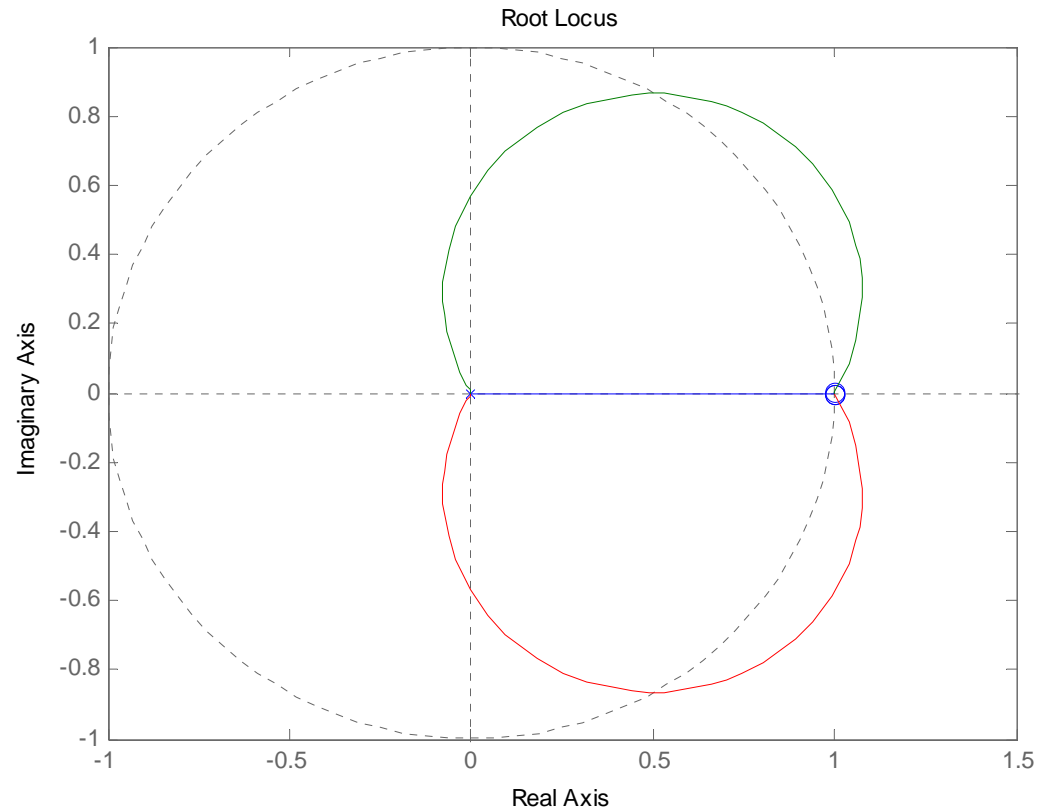
Describing Function Analysis contd.

- Amplitude = 1.5 FS
- $k_0 = 0.61, k_1 = 0.26.$



File: Describing_Function_Analysis.m

3rd-Order DSM: NTF Root Locus



File: Third_Order_DSM_Root_Locus.m