

LPSC GRENOBLE  
for  
CALICE Collaboration

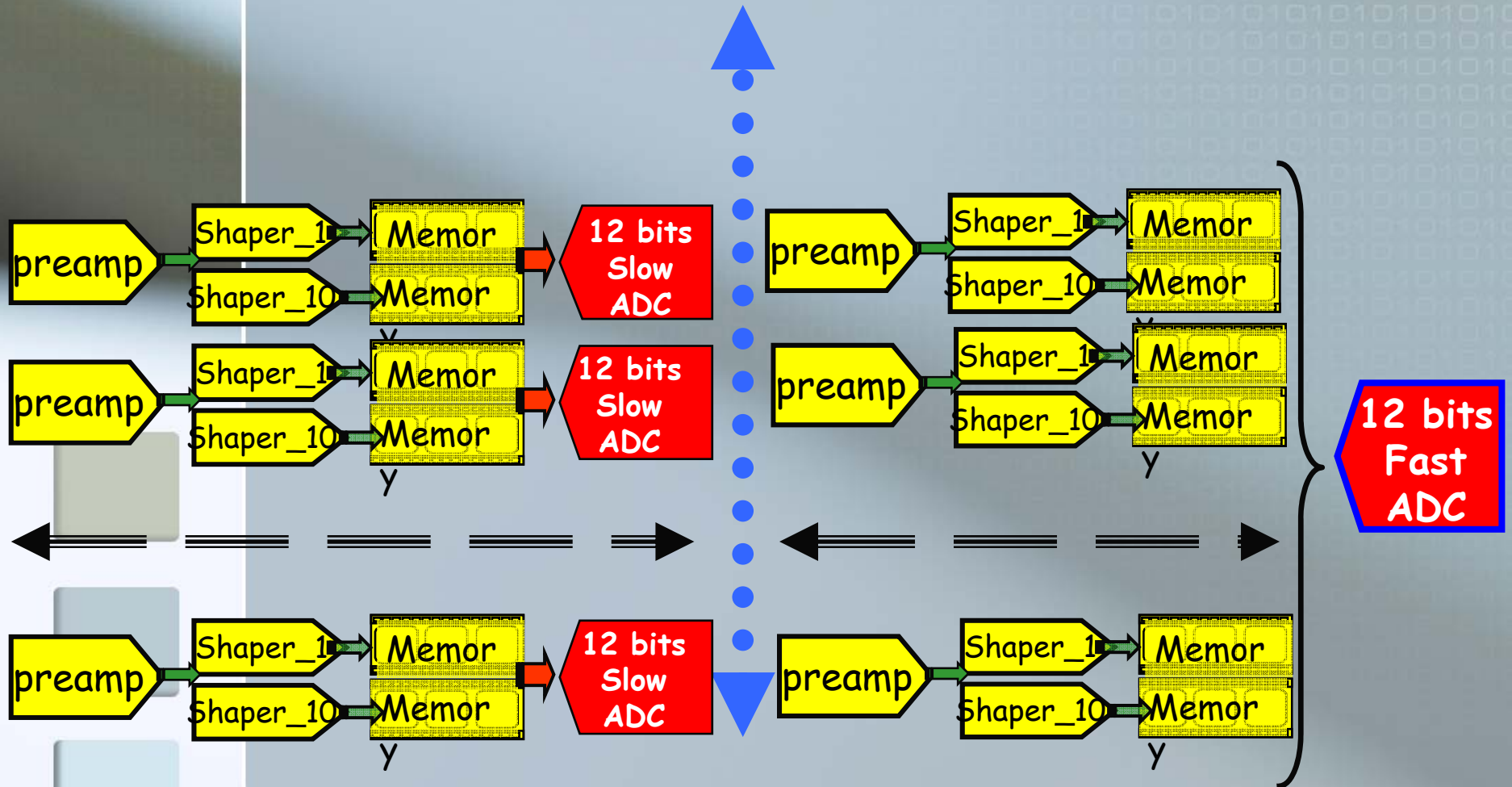
A low power 12-bits up to 30-MS/s  
**pipelined ADC** (*F.Rarbi; D. Dzahini*)

**16 bits DAC for the CALIBRATION**

*Laurent Gallin Martel / Olivier Rossetto / D. Dzahini*

# Slow versus fast digitizer for CALICE

Which one will optimize better: Power / Clock noise / cross talk



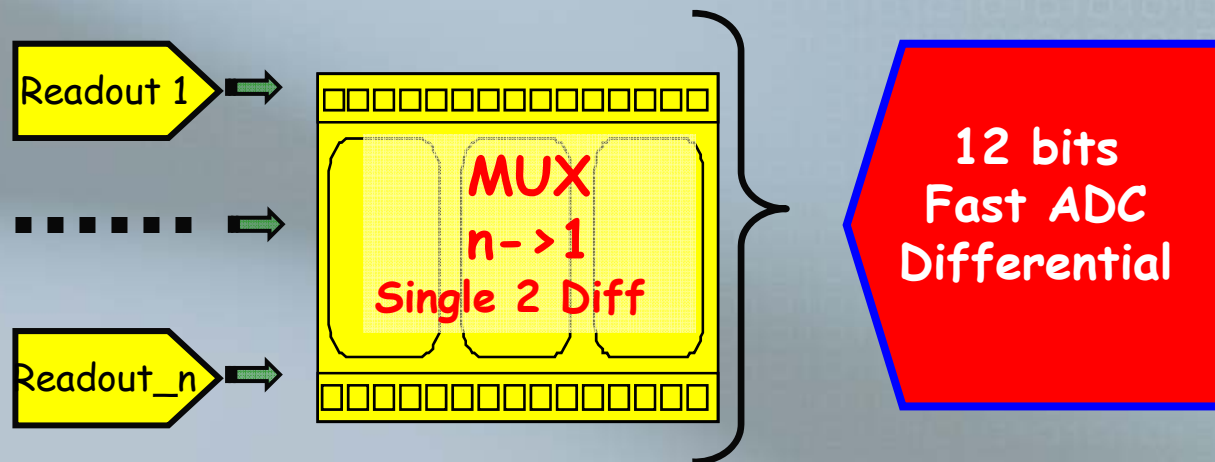
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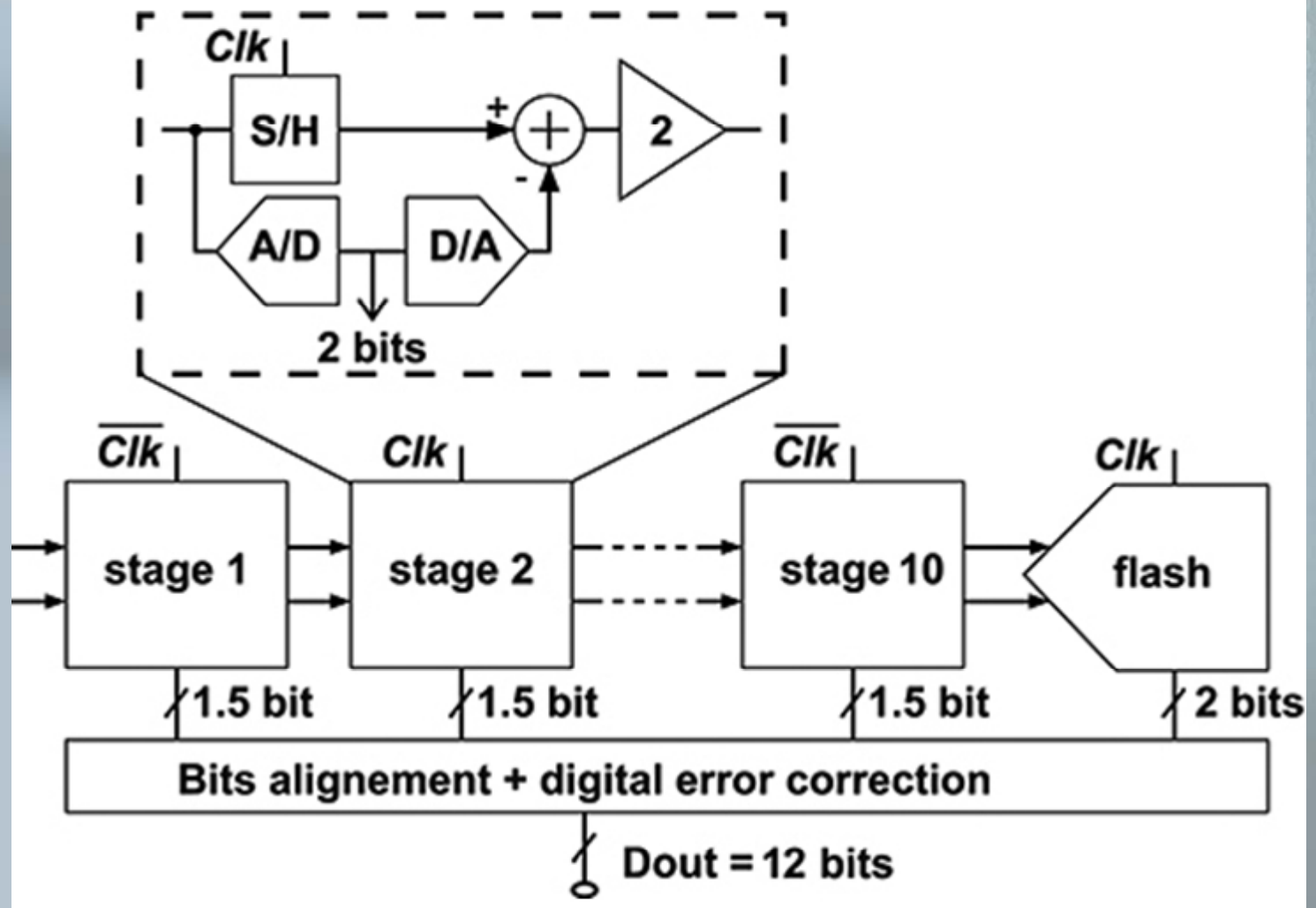
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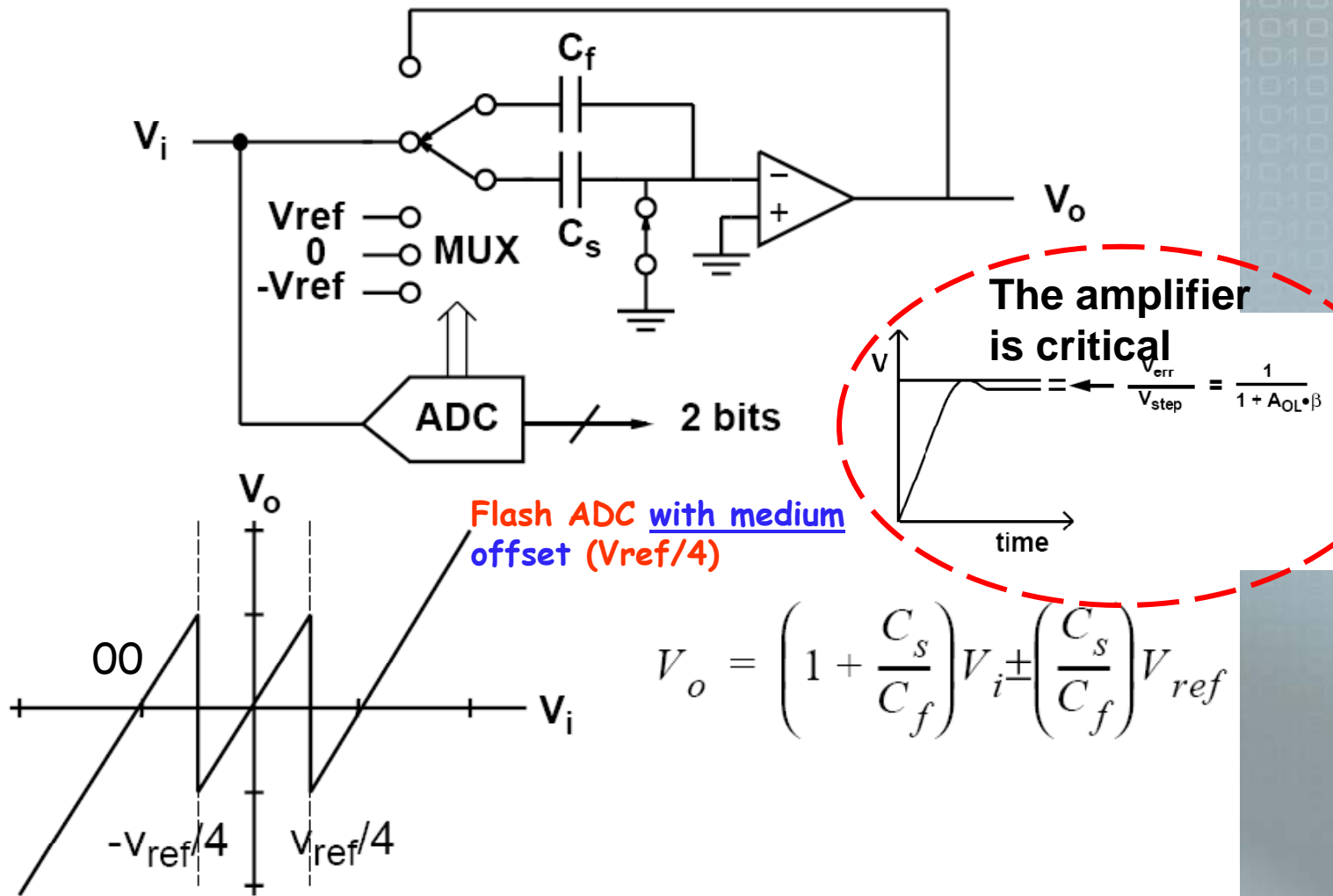
# Fast MUX + High speed ADC proposal



# ADC Pipeline



# MDAC Function: Multiplier & DAC



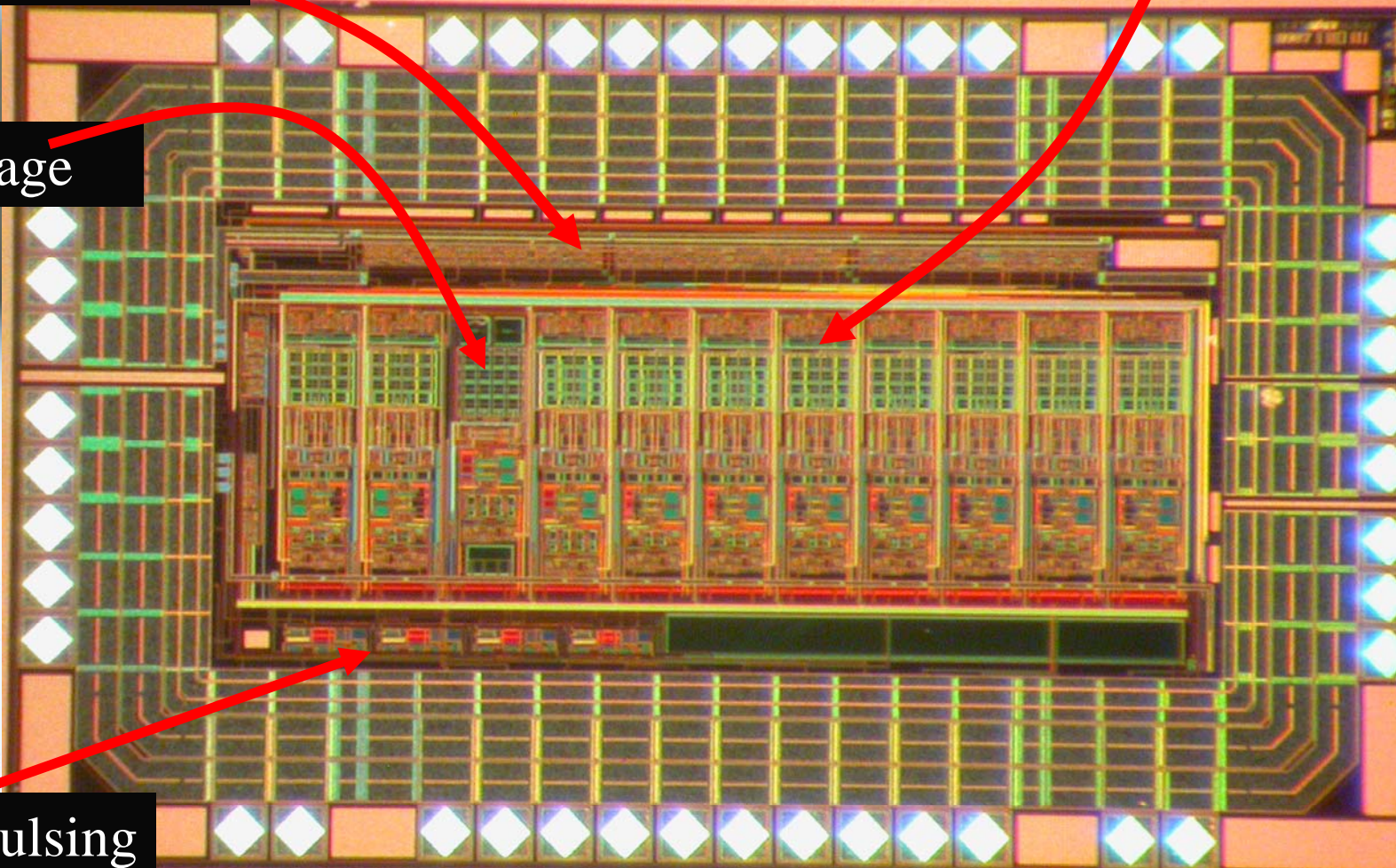
# Die of the «prototype n-2» Oct.07: S&H+ADC

Error Correction

A 1.5 bits stage

S&H stage

Power pulsing



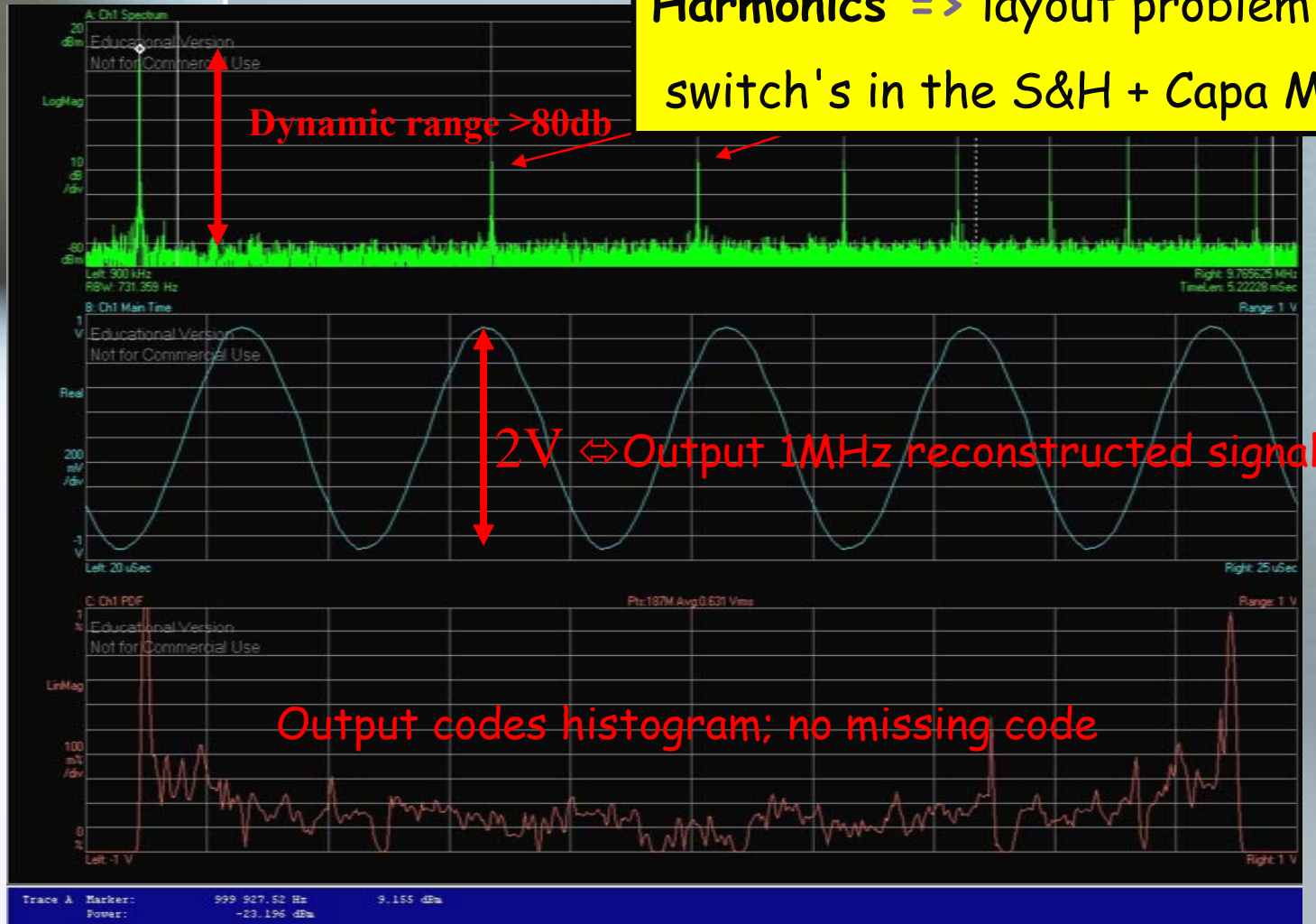
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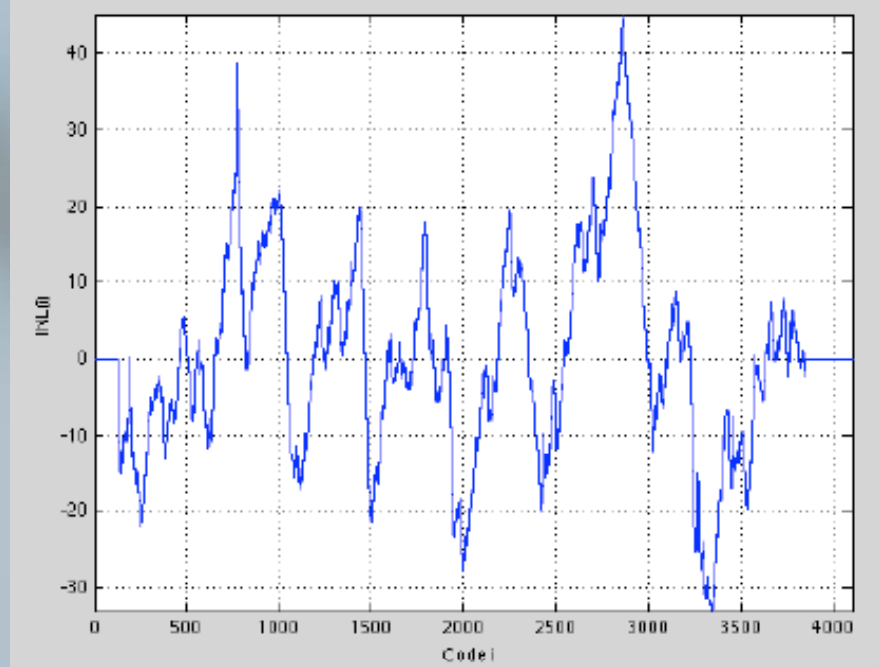
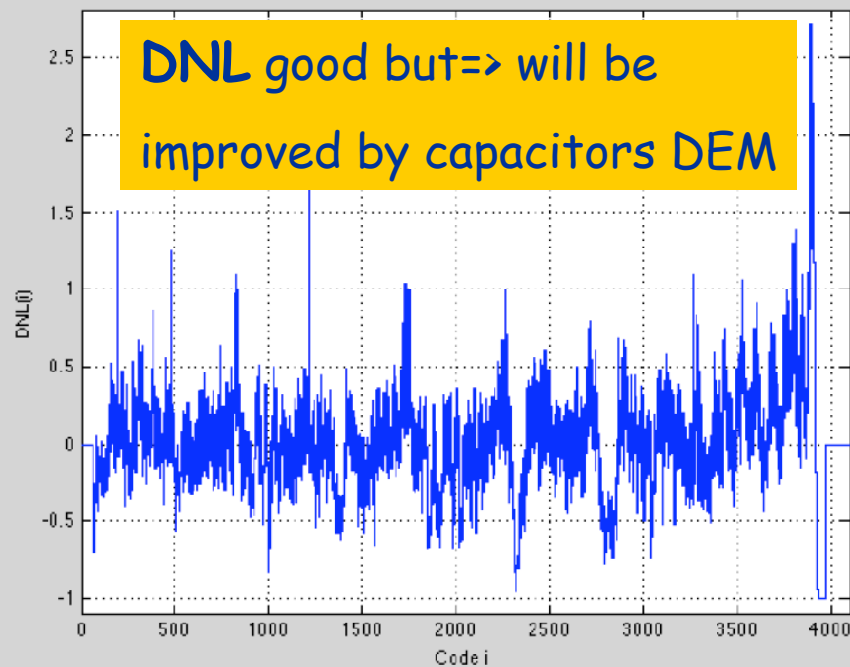
# Spectral response to an input sinus: 1Mhz, 2V pp

Harmonics => layout problem from the switch's in the S&H + Capa Matching



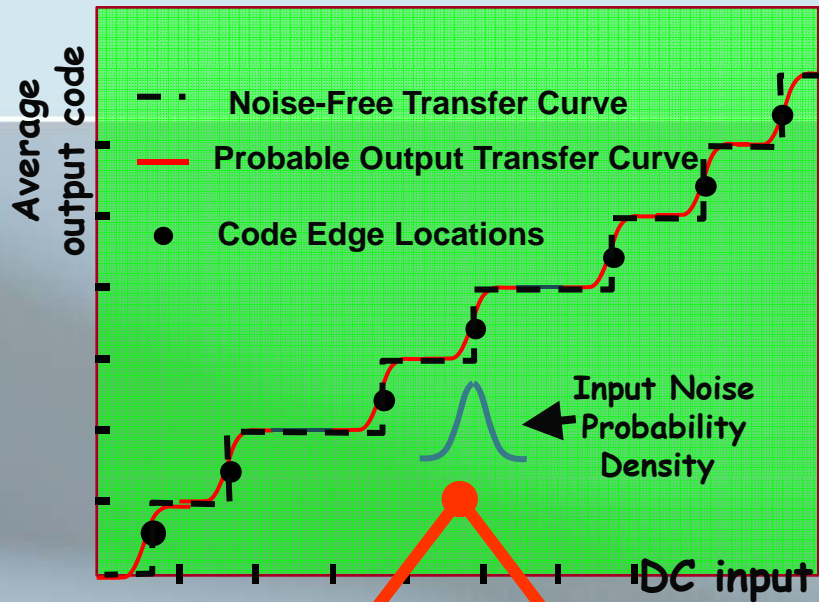
# Linearity results extracted from the histogram method (1MHz)

The Clock noise in S&H (layout problem) and capacitors Mismatch leads to distortions and high INL;

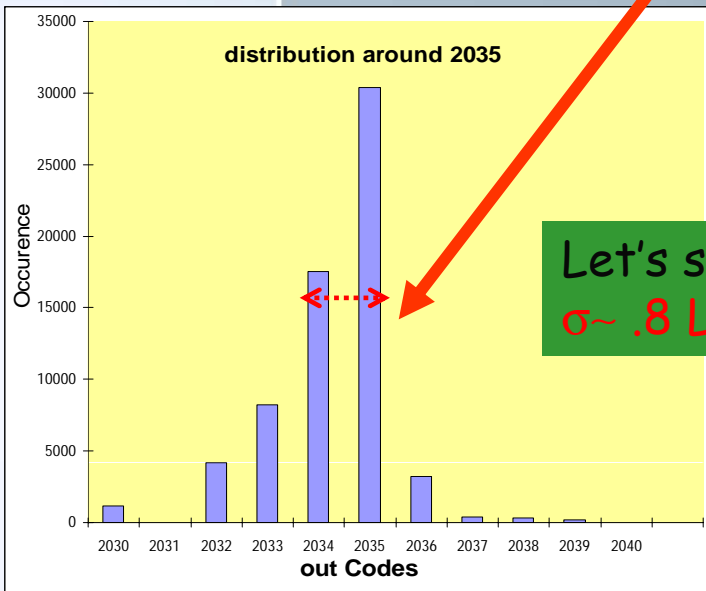




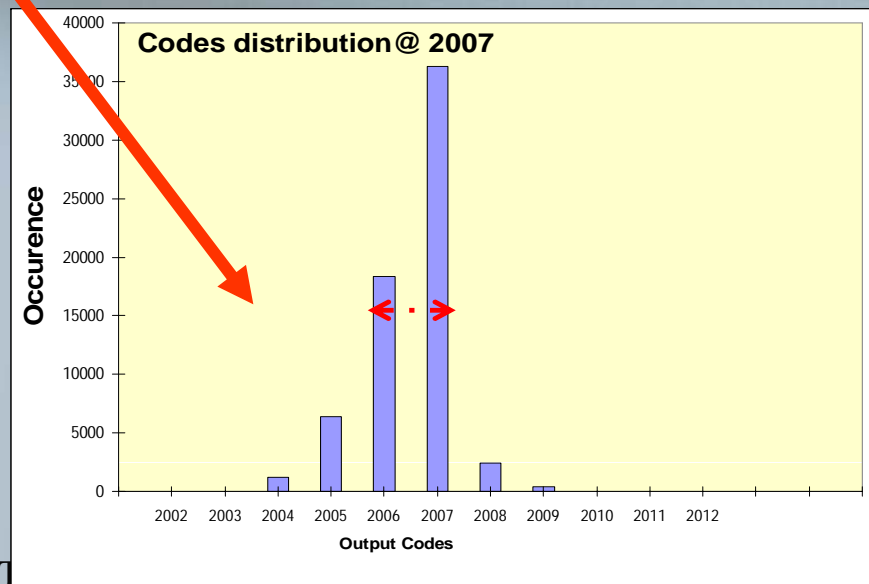
# Code-Edge Noise results @25Mhz



$\sigma \sim .8$  LSB for SH+ADC  
 Means  $.8/\sqrt{2} \approx .6$  LSB  
 for ADC alone



Let's say :  
 $\sigma \sim .8$  LSB



SC -FAT

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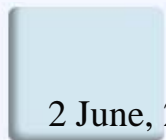
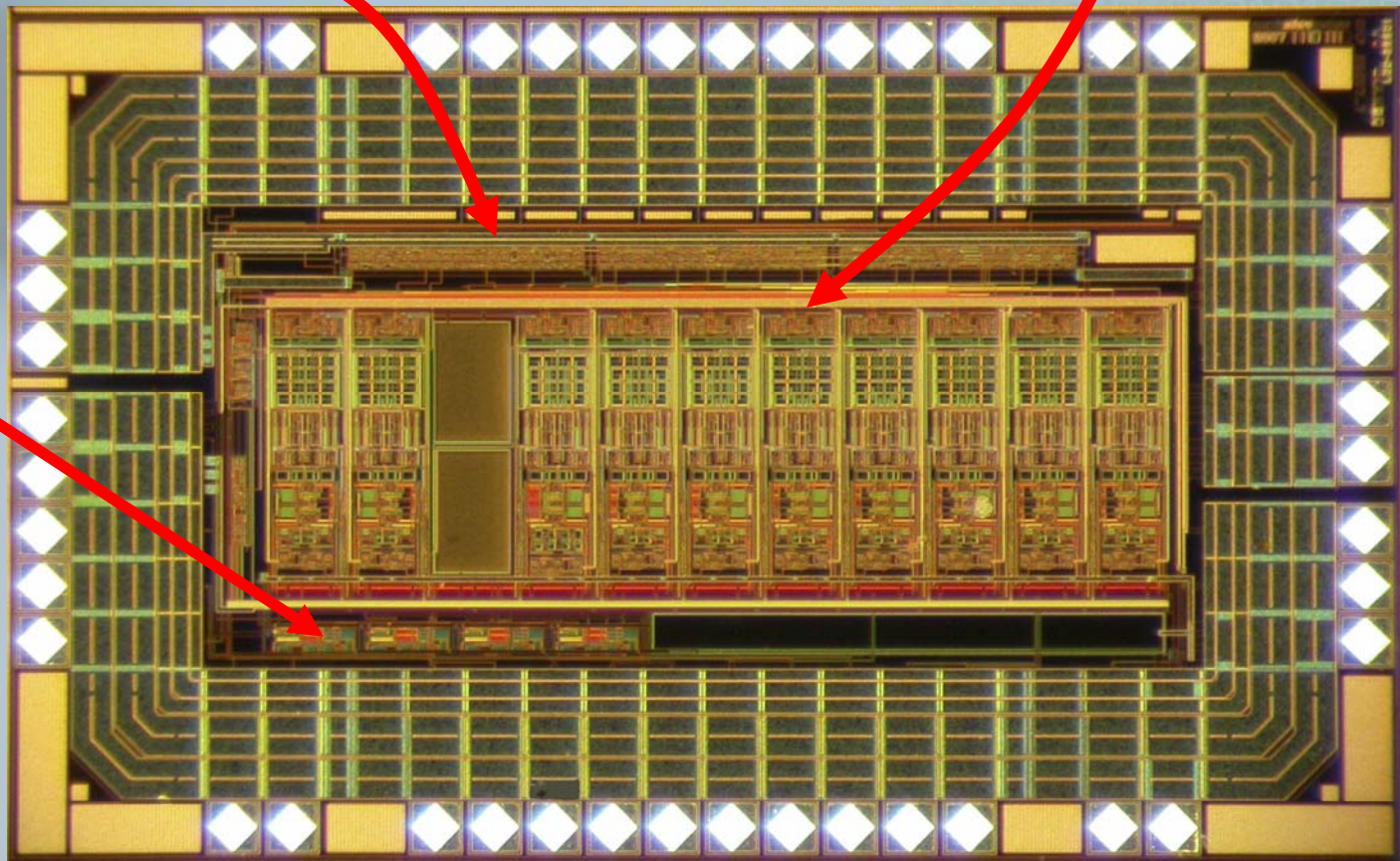
# ADC 12b Without S&Hold, 30MHz&35mW

: April 08=> prototype n-1

Error  
Correction

A 1.5 bit stage

Power  
pulsing



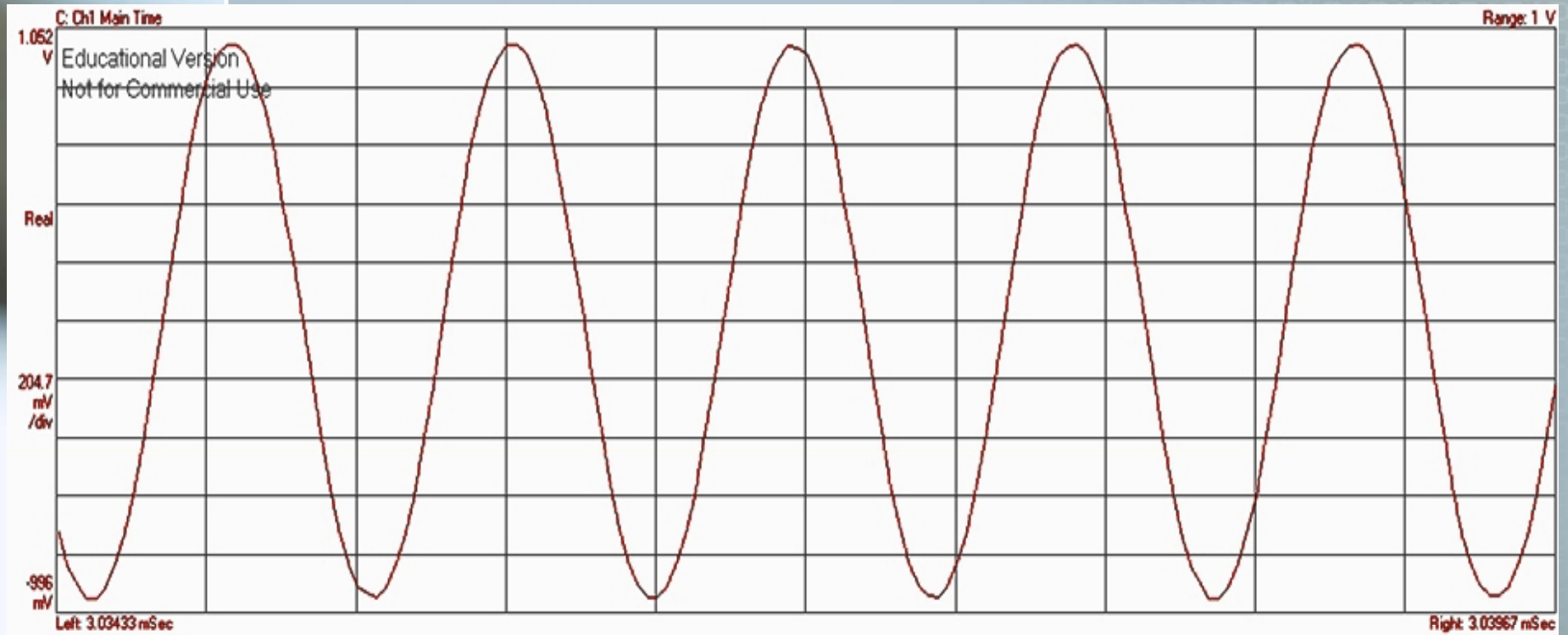
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# 2 Vpp Dynamic range at 30MHz ADC Pipeline

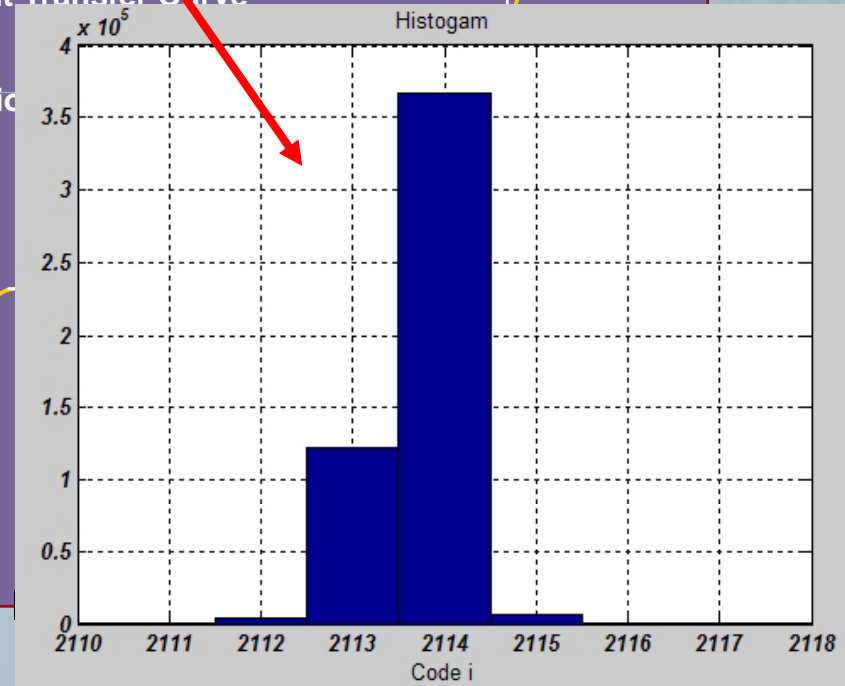
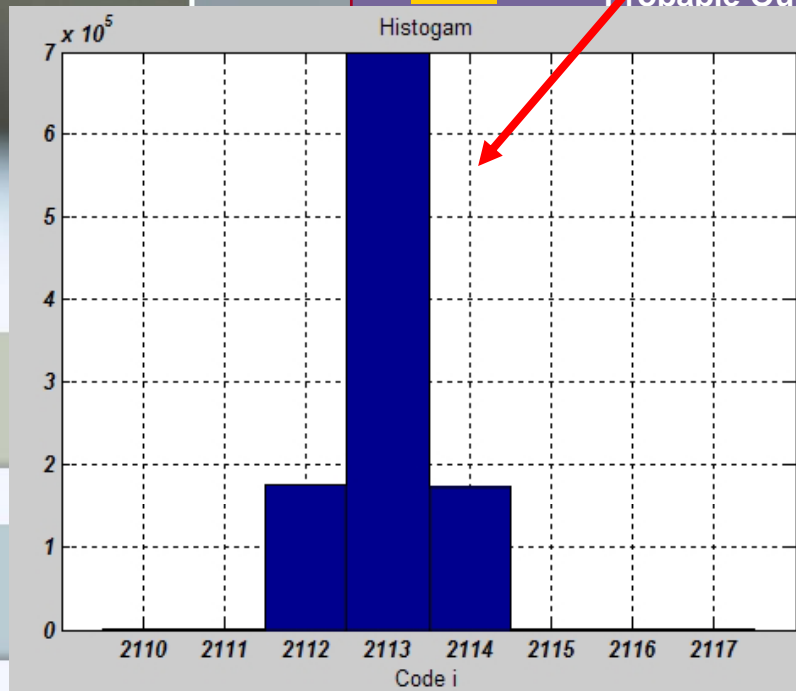
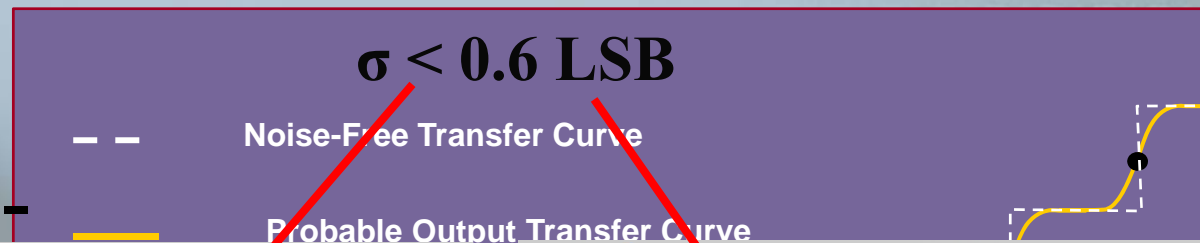


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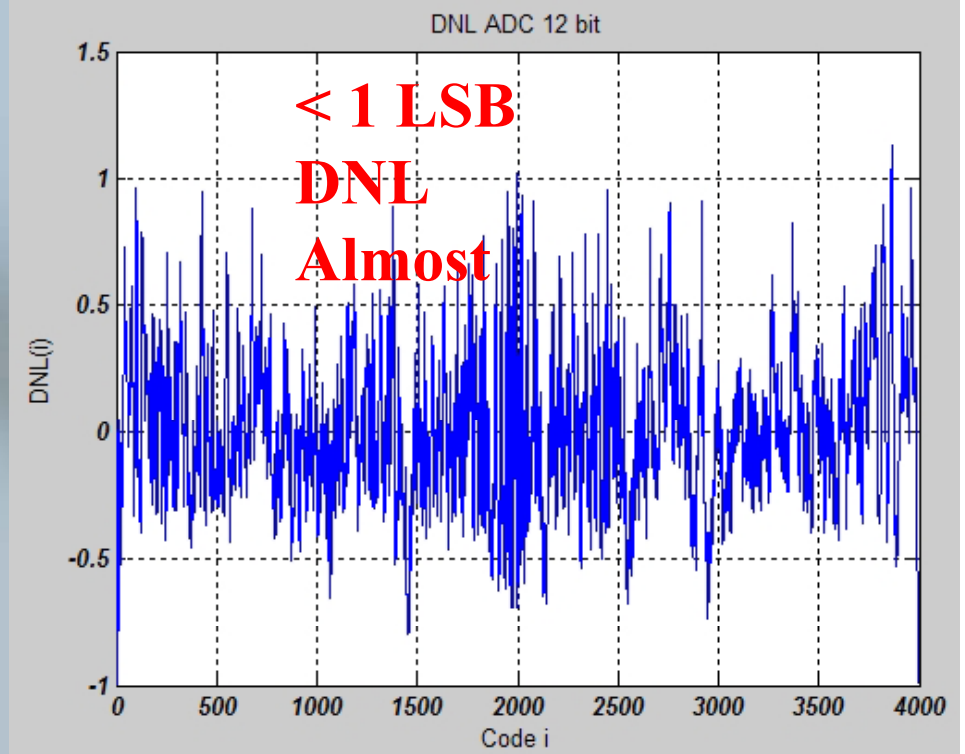
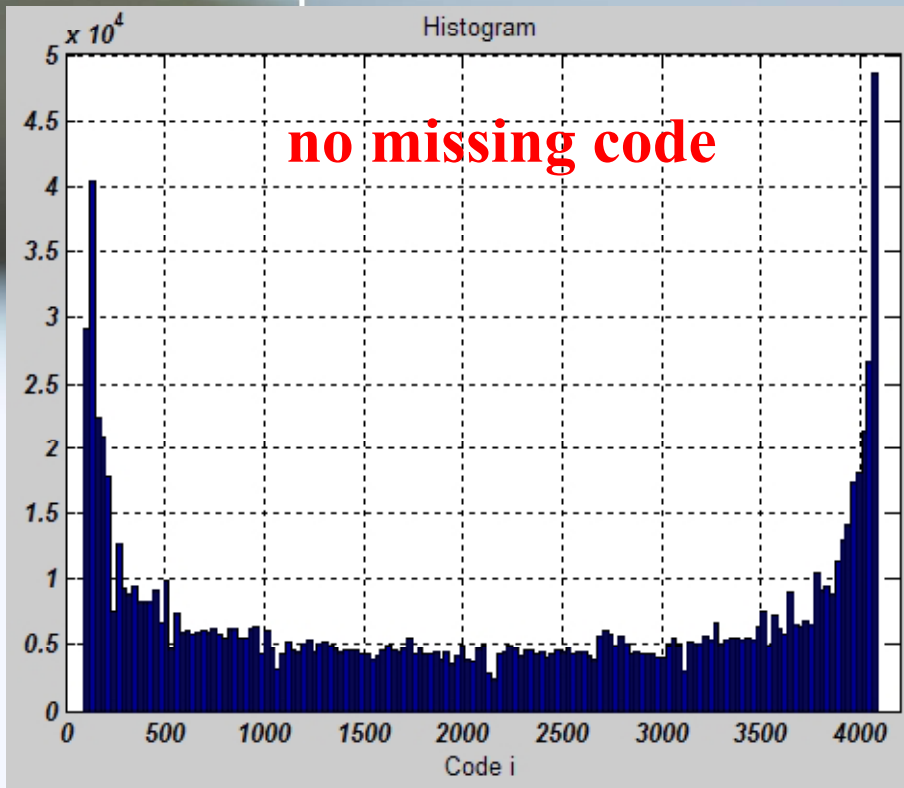
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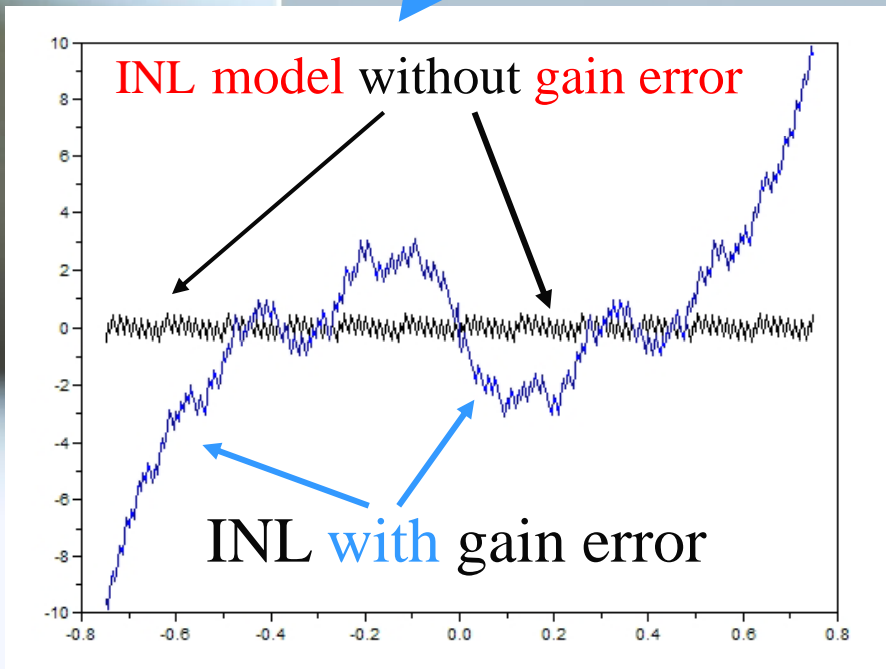
# Noise distribution at 30MHz



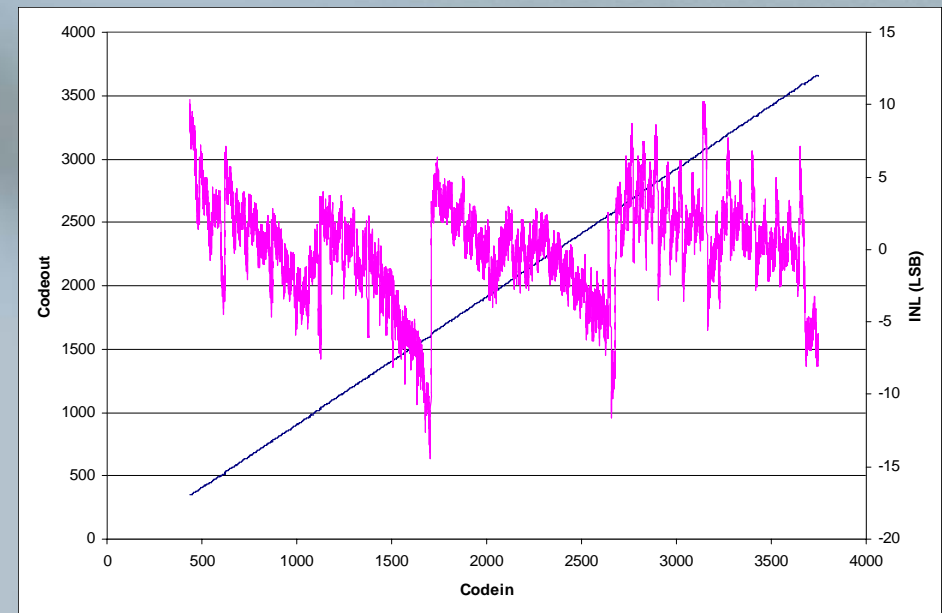
# Missing code and DNL at 30MHz on a 1MHz sinus



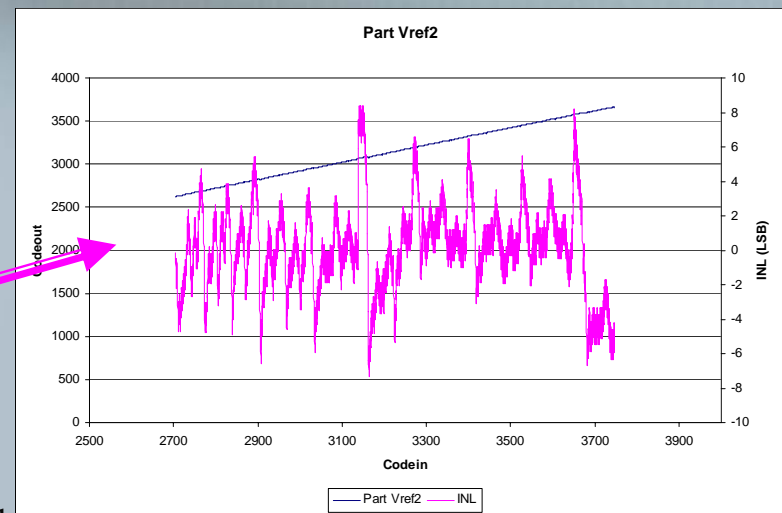
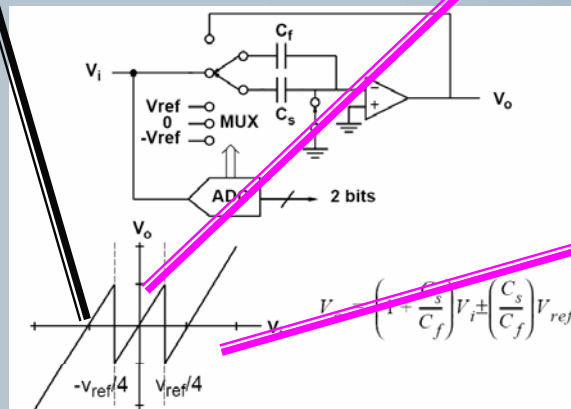
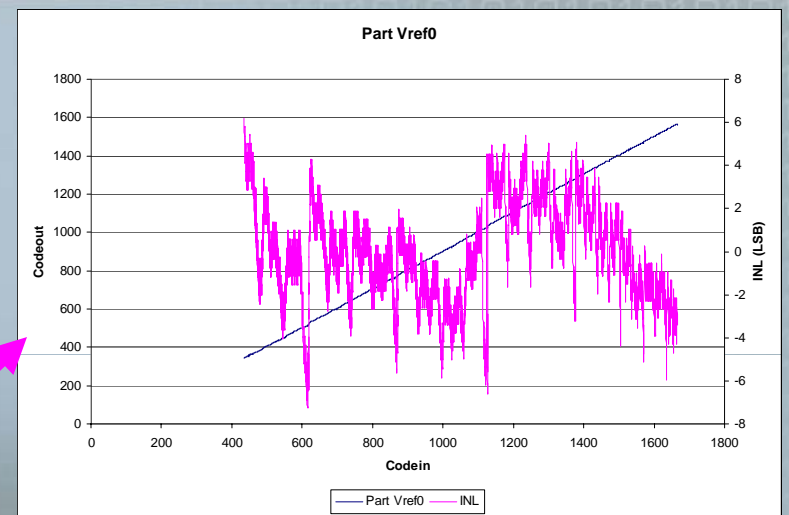
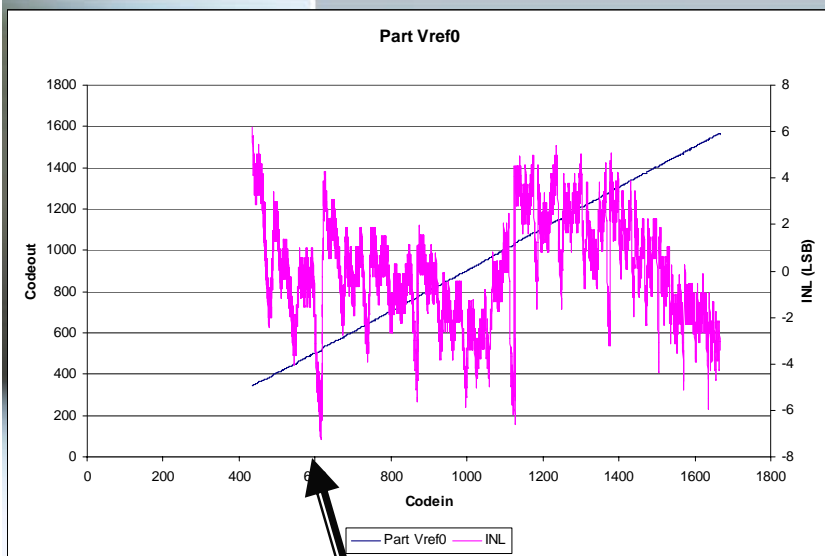
# INL from ADC Model with and without gain error versus testing results



This shape (in 3 sections)  
Corresponds to mismatch  
From the first stage.



# INL testing results in each section of the first MDAC stage



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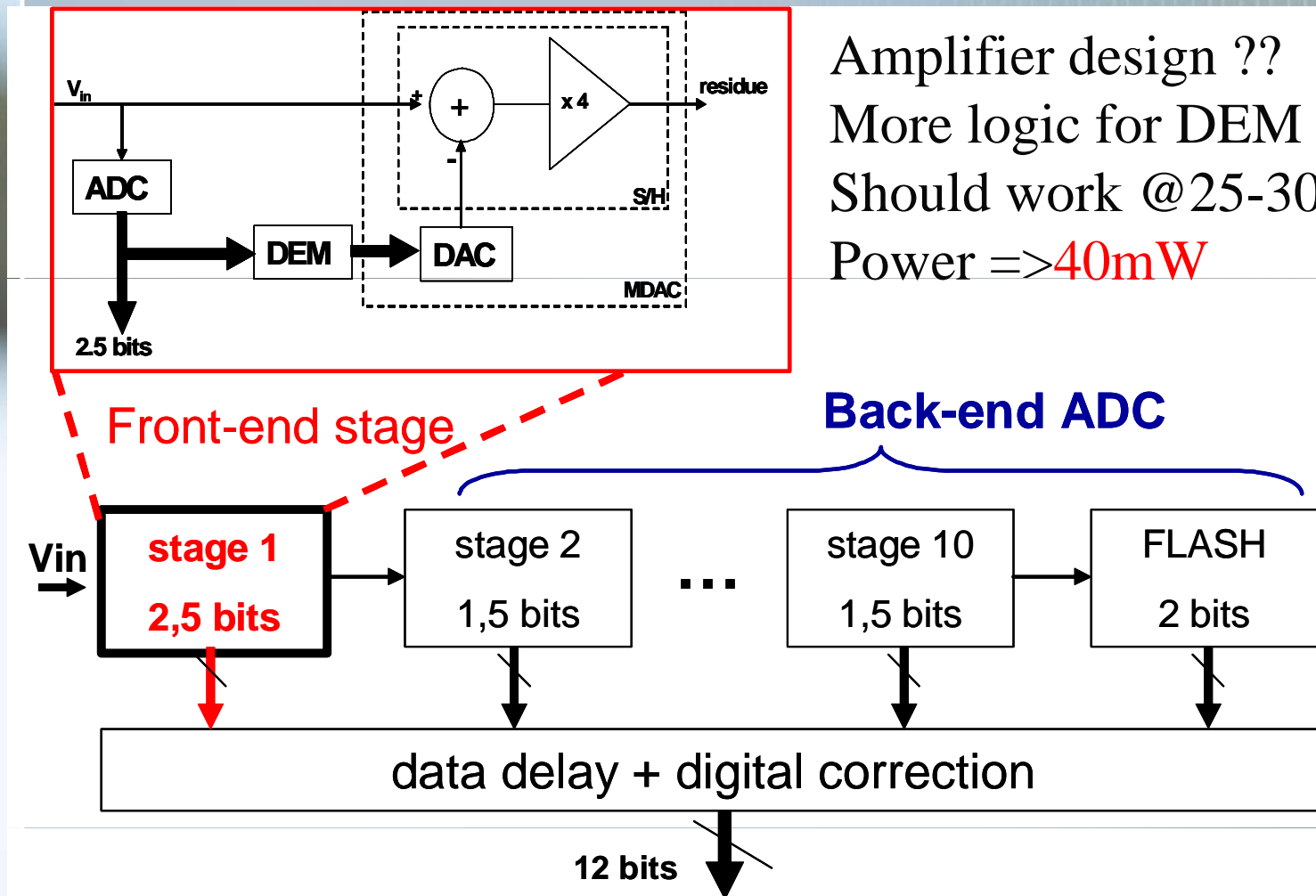
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# Trends for INL reduction in 12 bits

- Our Design reaches the limits for **Cpoly matching** in AMS CMOS process.
- Possible improvements using **Cmim** capa in SiGe.
- Increase the number of bits +DEM in the first stage=> Done.
- **Digital gain correction** algorithm very soon.
- Analog gain setting ??? *I do not like it.*

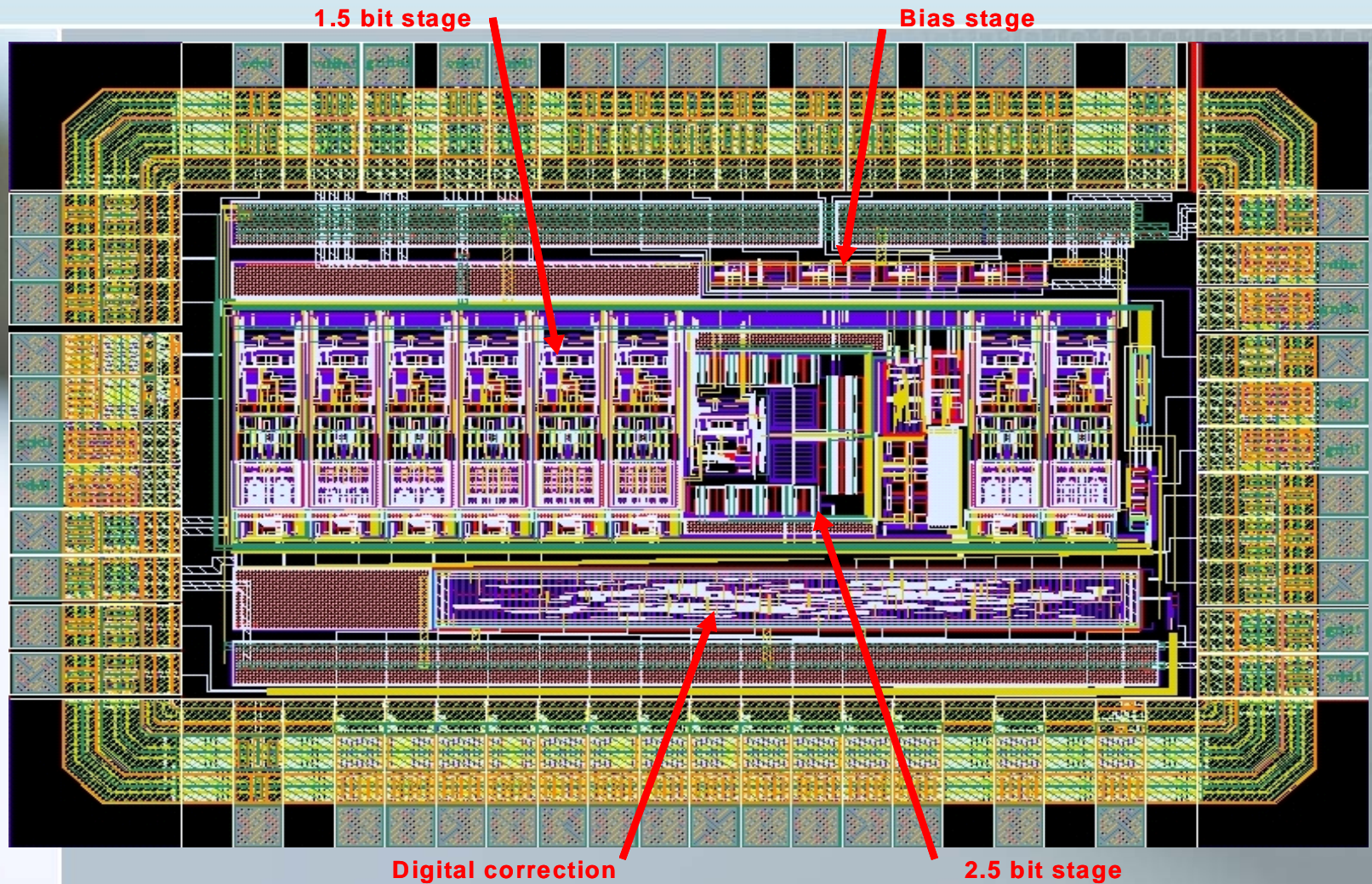


# Prototype "n" submitted April 08



Amplifier design ??  
More logic for DEM  
Should work @25-30Mhz  
Power =>40mW

# Prototype April 08 (4mm<sup>2</sup> ou .9\*1.9 mm<sup>2</sup> (coeur))



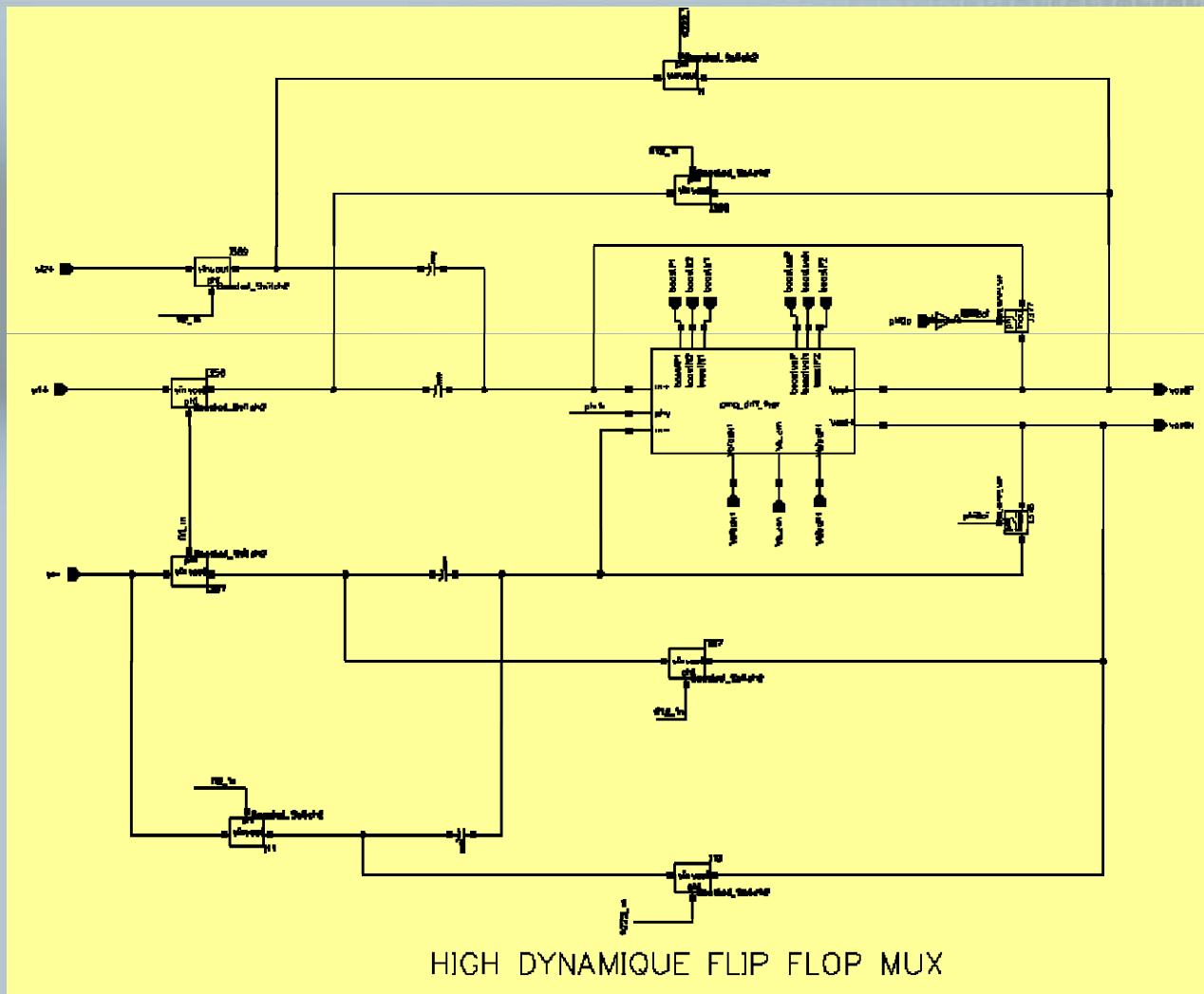
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# ANALOG FLIP FLOP MUX (12 bits)

## Next steps

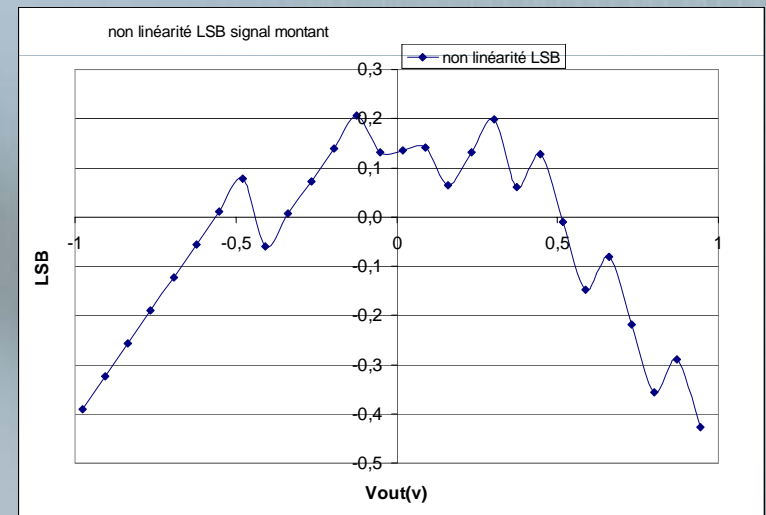
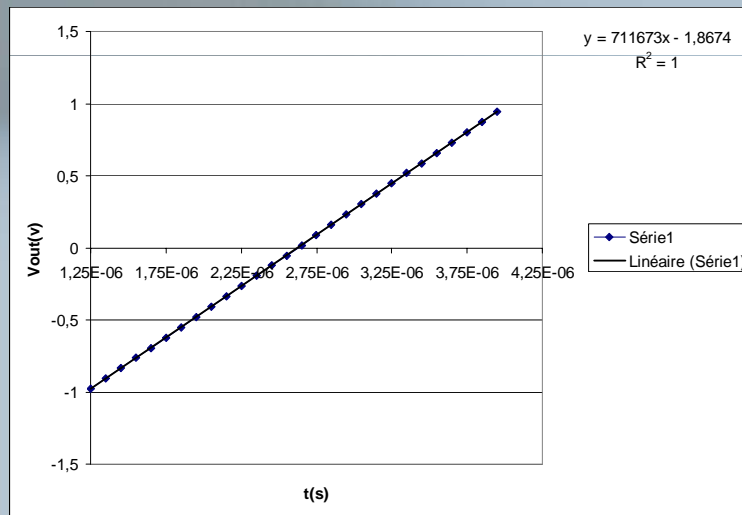


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# Beginning of simulations



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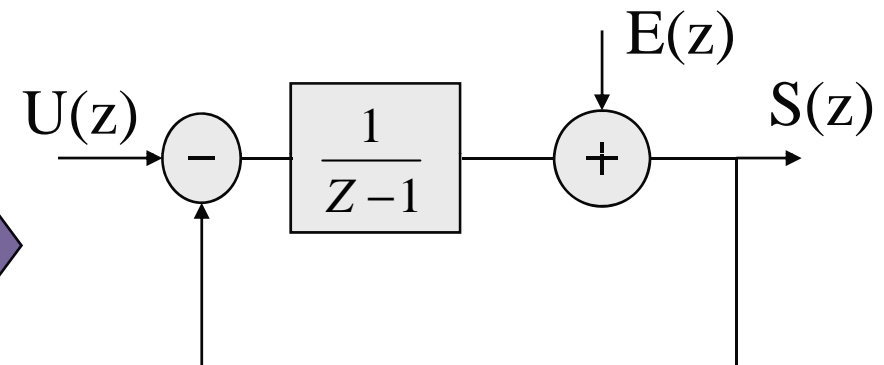
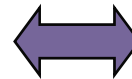
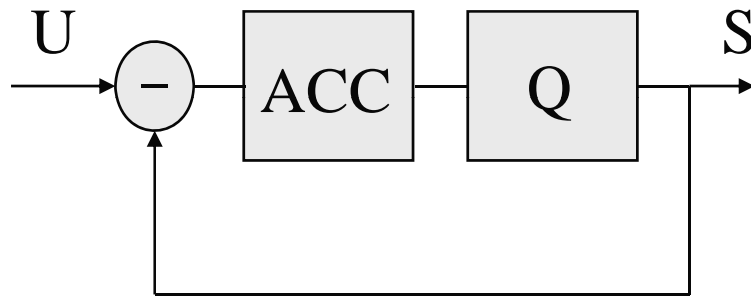
## **16 bits DAC for the CALIBRATION**

*Laurent Gallin Martel / Olivier Rossetto /D. Dzahini*

# Modulation Sigma Delta

Modulateur "single stage" du premier ordre

$E(z)$  : quantization noise



ACC somme les erreurs de quantification et agit comme un filtre passe haut du point de vue du bruit.

C'est le nombre d'intégrateurs qui détermine l'ordre du modulateur.

Le quantificateur peut avoir de 2 à m niveaux de quantification.

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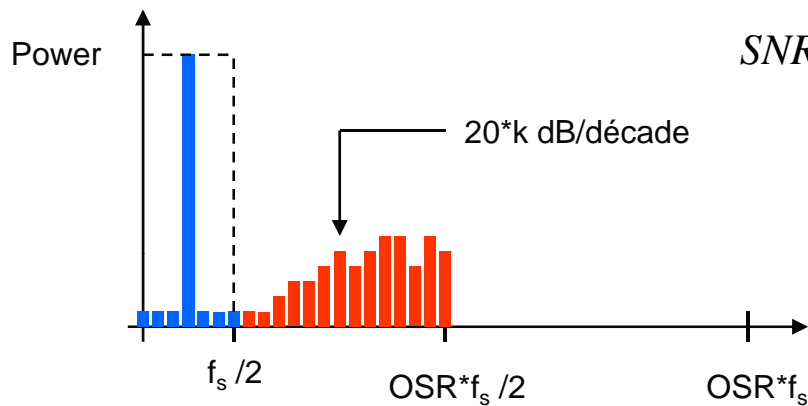
$$\left\{ \begin{array}{l} S(z) = z^{-1}U(z) + (1 - z^{-1})E(z) \\ STF = Z^{-1} \\ NTF = 1 - Z^{-1} \text{ (high pass filter)} \end{array} \right.$$

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# Modulation Sigma Delta

SNR d'un convertisseur Sigma Delta

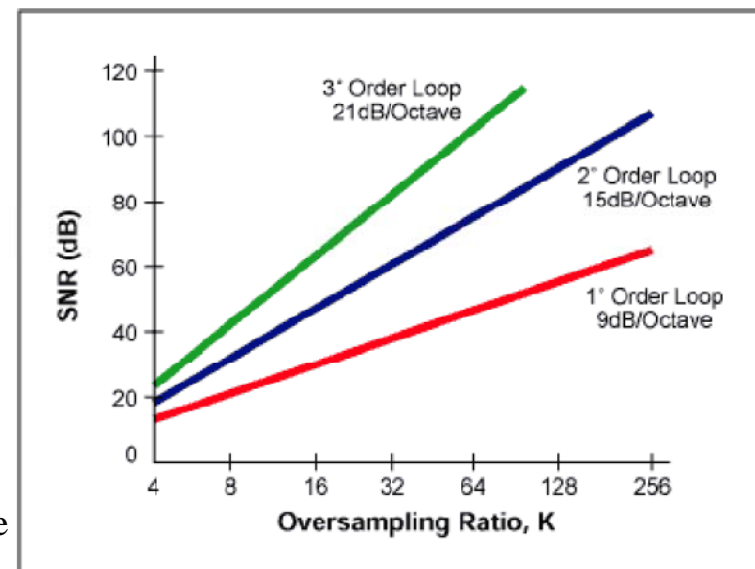


$$SNR_{dB} = 6.02N + 1.76 + 10 \log \left( \frac{2k+1}{\pi^{2k}} \right) + 10(2k+1) \log(OSR)$$

- N : nombre de bits du quantificateur
- k : ordre du modulateur
- OSR : taux de sur échantillonnage

SNR d'un modulateur 1 bit

[http://www.maxim-ic.com/appnotes.cfm/an\\_pk/1870](http://www.maxim-ic.com/appnotes.cfm/an_pk/1870)

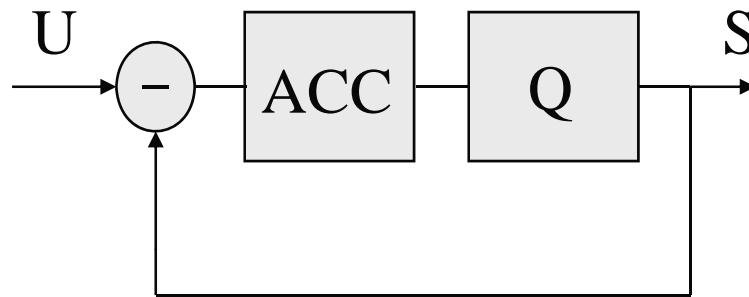


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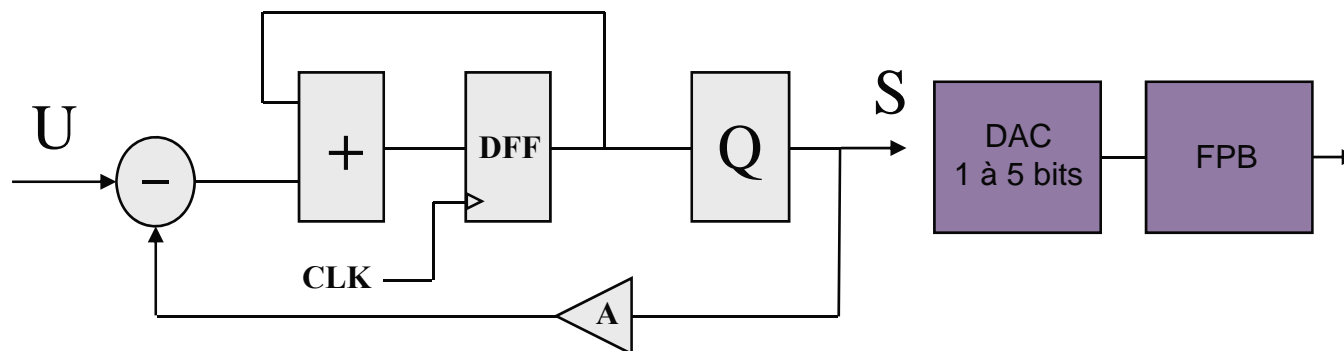
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# Modulation sigma delta

Modulateur "single stage" du premier ordre – Implémentation pour un DAC



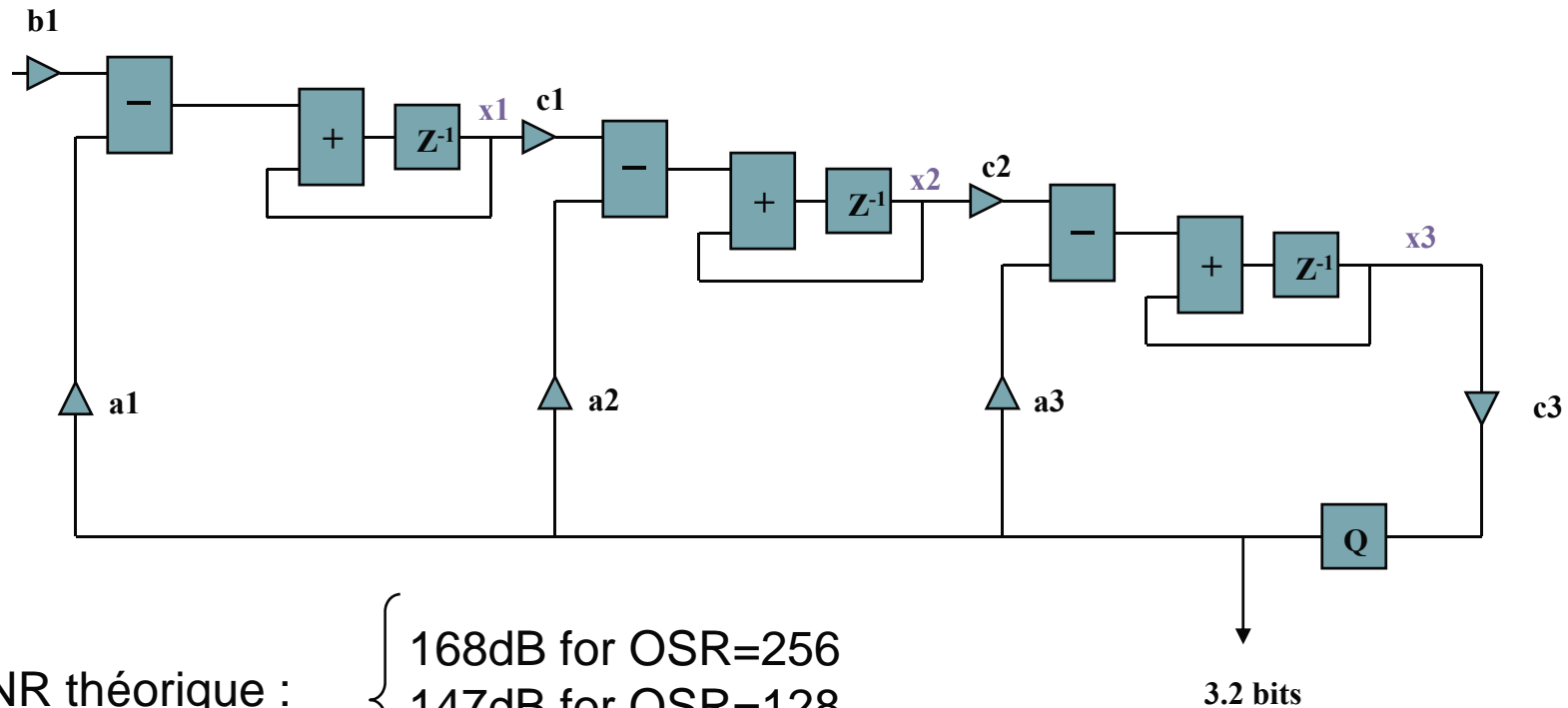
Quantificateur à m niveaux => nb de bits  $n = \log(m)/\log(2)$   
=> de 2 à 32 niveaux de quantification





# DAC sigma delta 16 bits

3<sup>e</sup> ordre – 9 niveaux de quantification



SNR théorique :

- 168dB for OSR=256
- 147dB for OSR=128
- 126dB for OSR=64

La toolbox permet de déterminer la valeur des coefficients ainsi que le taille des bus xn pour atteindre (approcher) le SNR théorique.

Une fois le DAC mis au point avec la toolbox on peut le simuler avec n'importe quel langage

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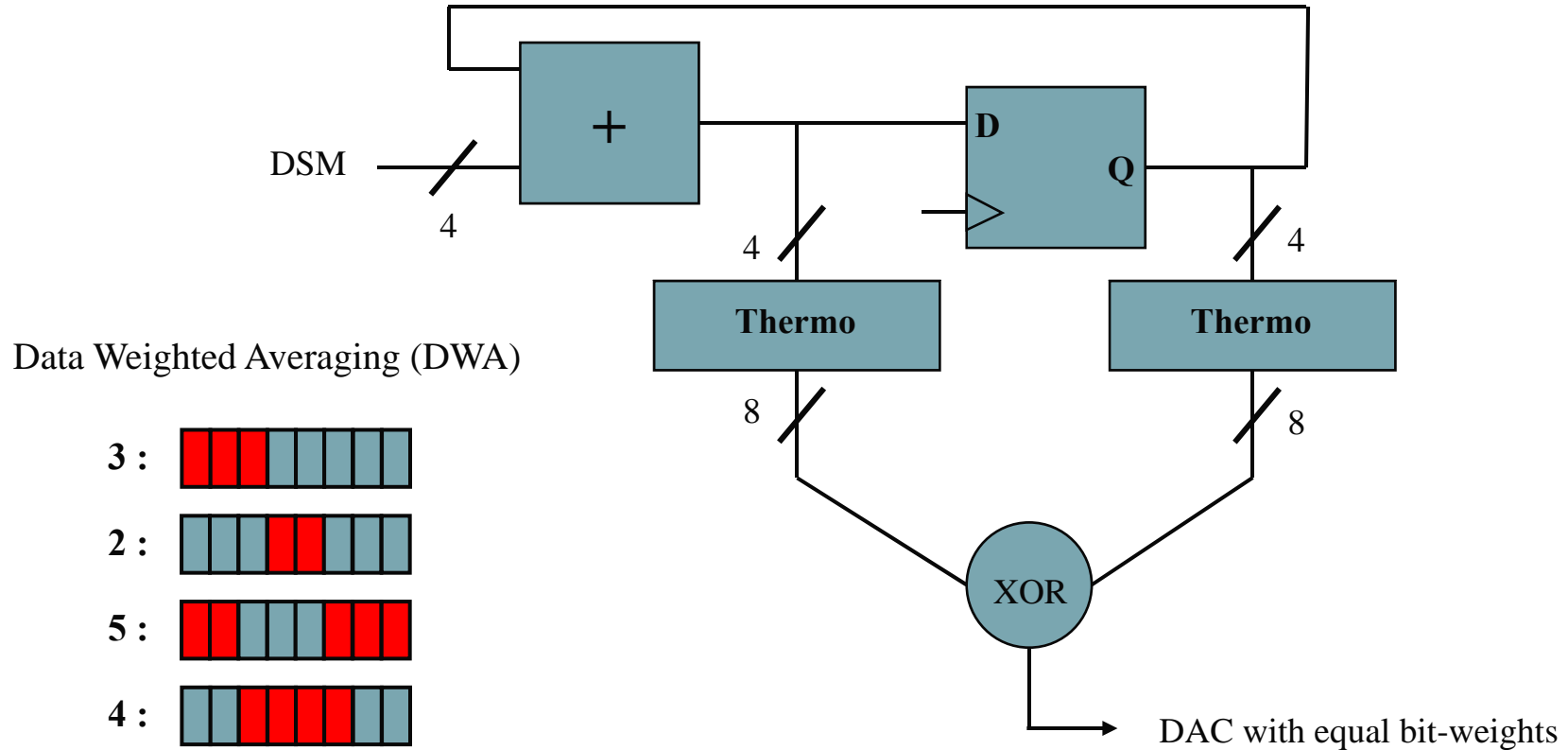
```

S=c3*x3;
if (S>4) S=4;
if (S<-4) S=-4;
x3=x3 + c2*x2 - a3*S;
x2=x2 + c1*x1 - a2*S;
x1=x1 + U - a1*v;
    
```

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# DAC sigma delta 16 bits

Dynamic Element Matching



Random re-mapping turns elements mismatch into white noise.

DWA re-mapping shapes DAC errors noise.

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# DAC sigma delta 16 bits

DAC 9 niveaux

Based on Direct Charge Transfert (DCT) :

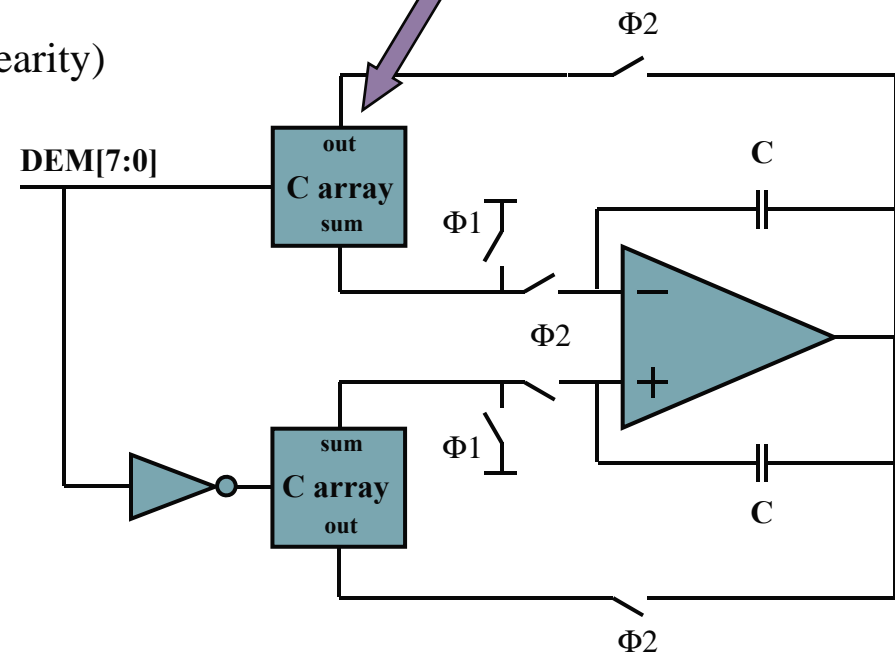
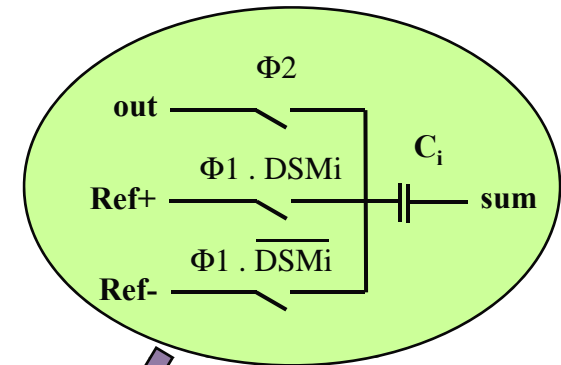
- DAC operation is performed during  $\Phi 1$  (capacitor arrays charging)
- Charges are shared with the feedback capacitor  $C$  during  $\Phi 2$  (DCT)
- Ref voltage is limited by the OTA dynamic range (linearity)

Low pass DCT :

$$H(z) = \frac{1}{1 + a - a \cdot z^{-1}}$$

$$a = \frac{C}{\sum c_i}$$

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# DAC sigma delta 16 bits

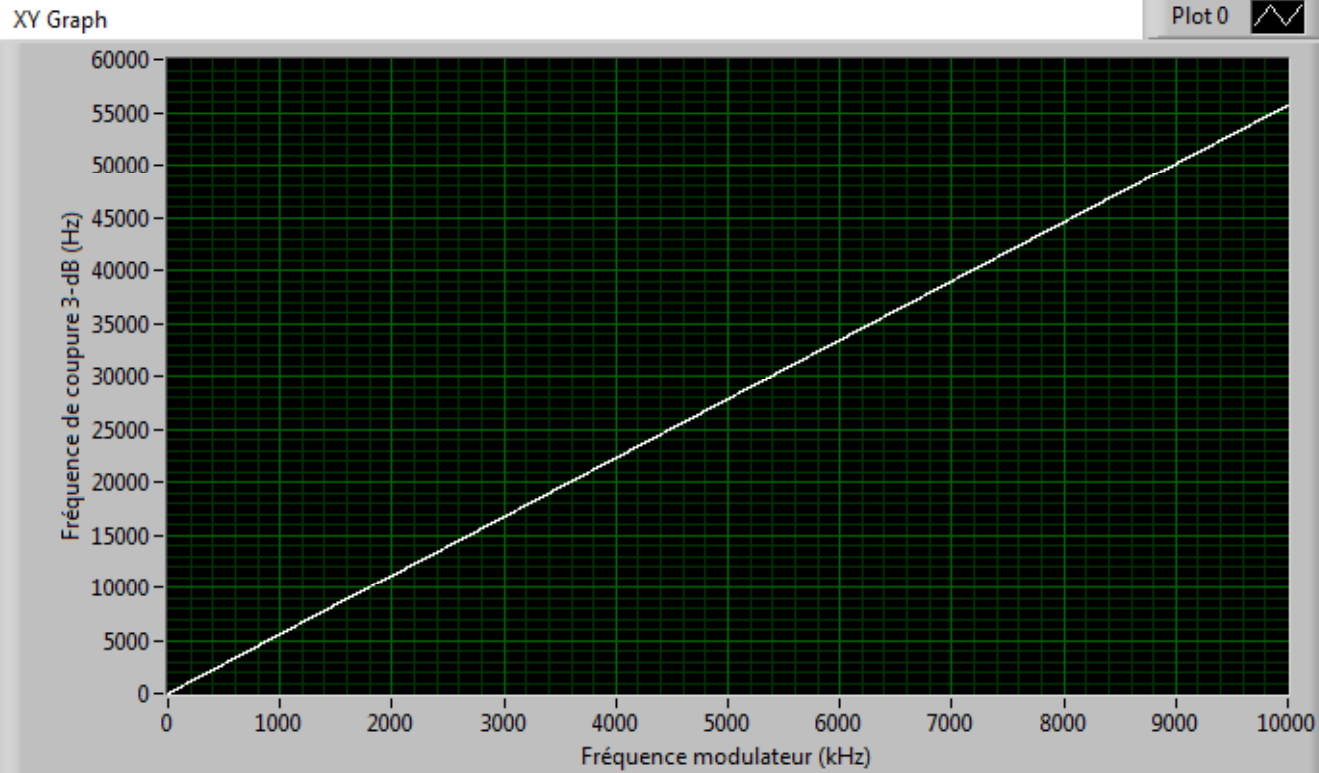
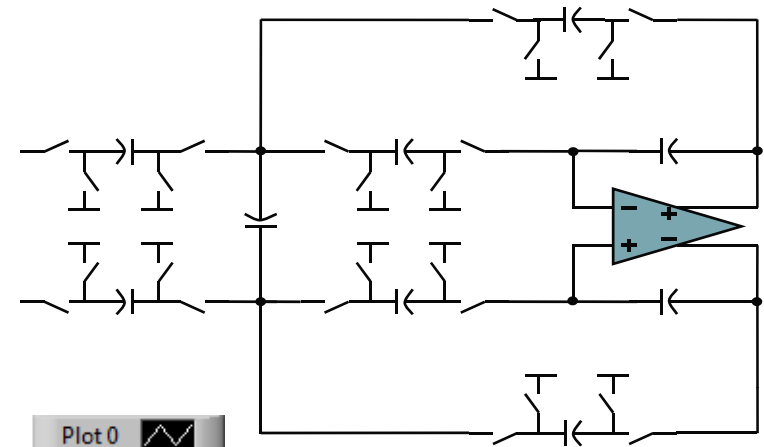
Filtre passe bas du 4<sup>e</sup> ordre (capa com)

Architecture Multi FeedBack (MFB)

2 filtres du second ordre (identiques) cascades

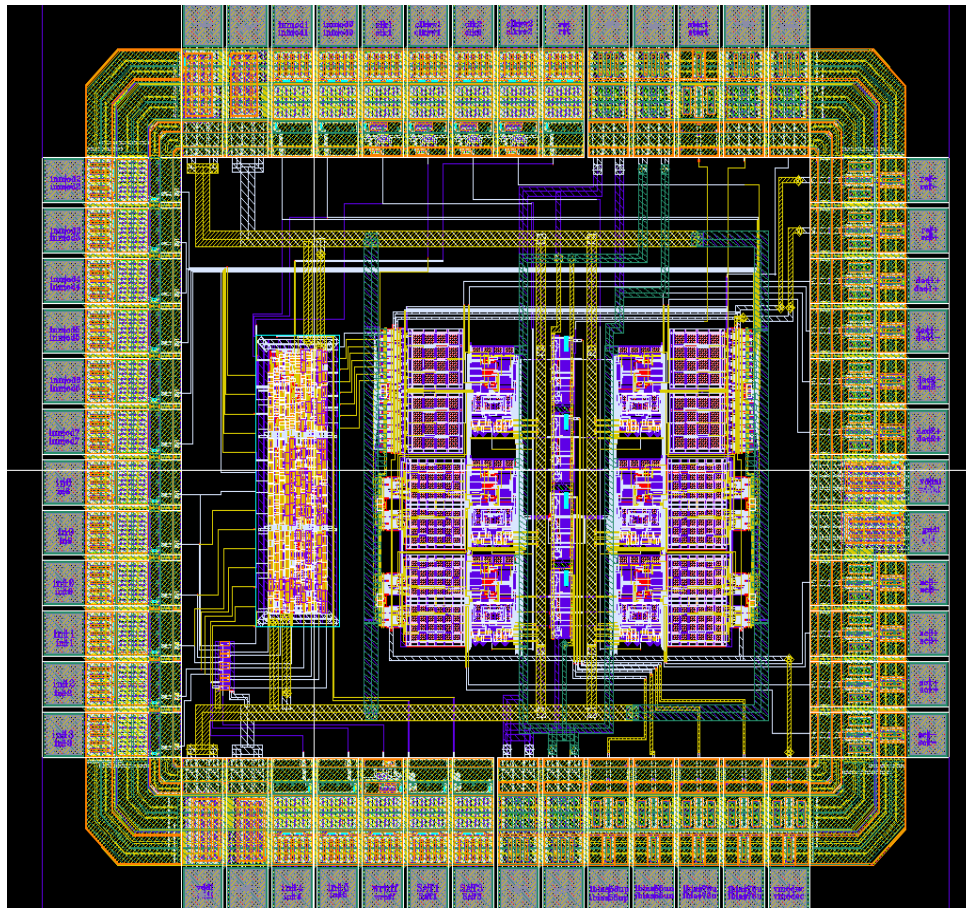
Simulations trop longues avec SPECTRE (> à 1 semaine !!!)

2 x =>



# DAC sigma delta 16 bits

Layout



AMS CMOS 0.35 $\mu$

Surface puce : 2000 x 2200  $\mu\text{m}^2$

Surface DSM : 650 x 200  $\mu\text{m}^2$

Surface 1 analogique : 700 x 500  $\mu\text{m}^2$

Boîtier PLCC52

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Le chip contient un DAC 16 bits complet et une deuxième partie analogique (DAC 3 bits + FPB) qui peut être connectée à modulateur externe (FPGA).

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# DAC sigma delta 16 bits

Caractéristiques (provisoires)

La carte de test utilise un ADC 16 bits (AD7621) qui présente un SNRmax de 90dB

Le bruit RMS obtenu avec cette carte est de 0.8 LSB =>  $20 \log \left( \frac{32768}{0.8 * \sqrt{2}} \right) = 89dB$

Le SNR obtenu pour le DAC 16 bits est de :

70dB (bruit RMS # 5 LSB)

THD :

85dB

INL :

+/- 3LSB

pour

Fdsm = 3 MHz  
Fsinus = 12kHz

Consommation partie numérique :

70µW @ 250kHz (simulation)

Consommation partie analogique :

11mW (simulation)

Il reste à mesurer la DNL et surtout à comprendre pourquoi le SNR est si mauvais.

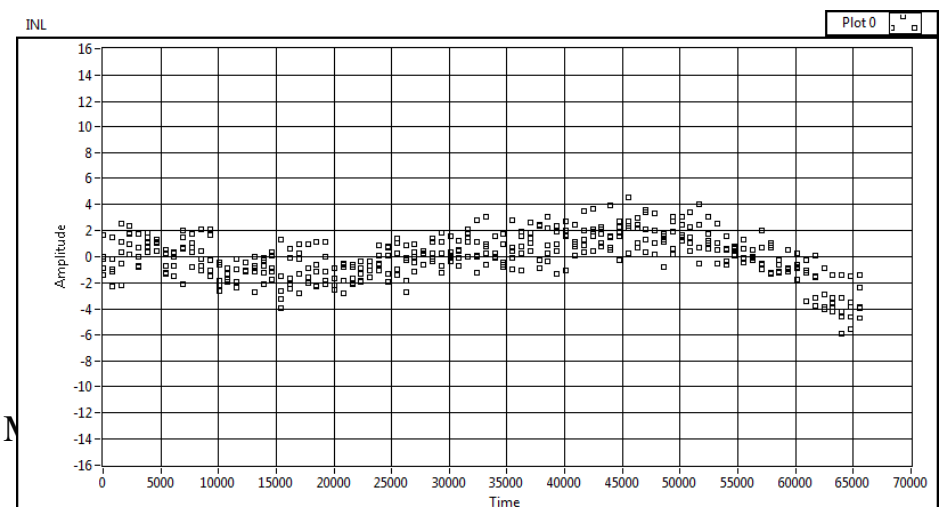
Pour un OSR de 250 : SNR théorique # 160dB !!!

D'autres DSM peuvent être testés dans le FPGA

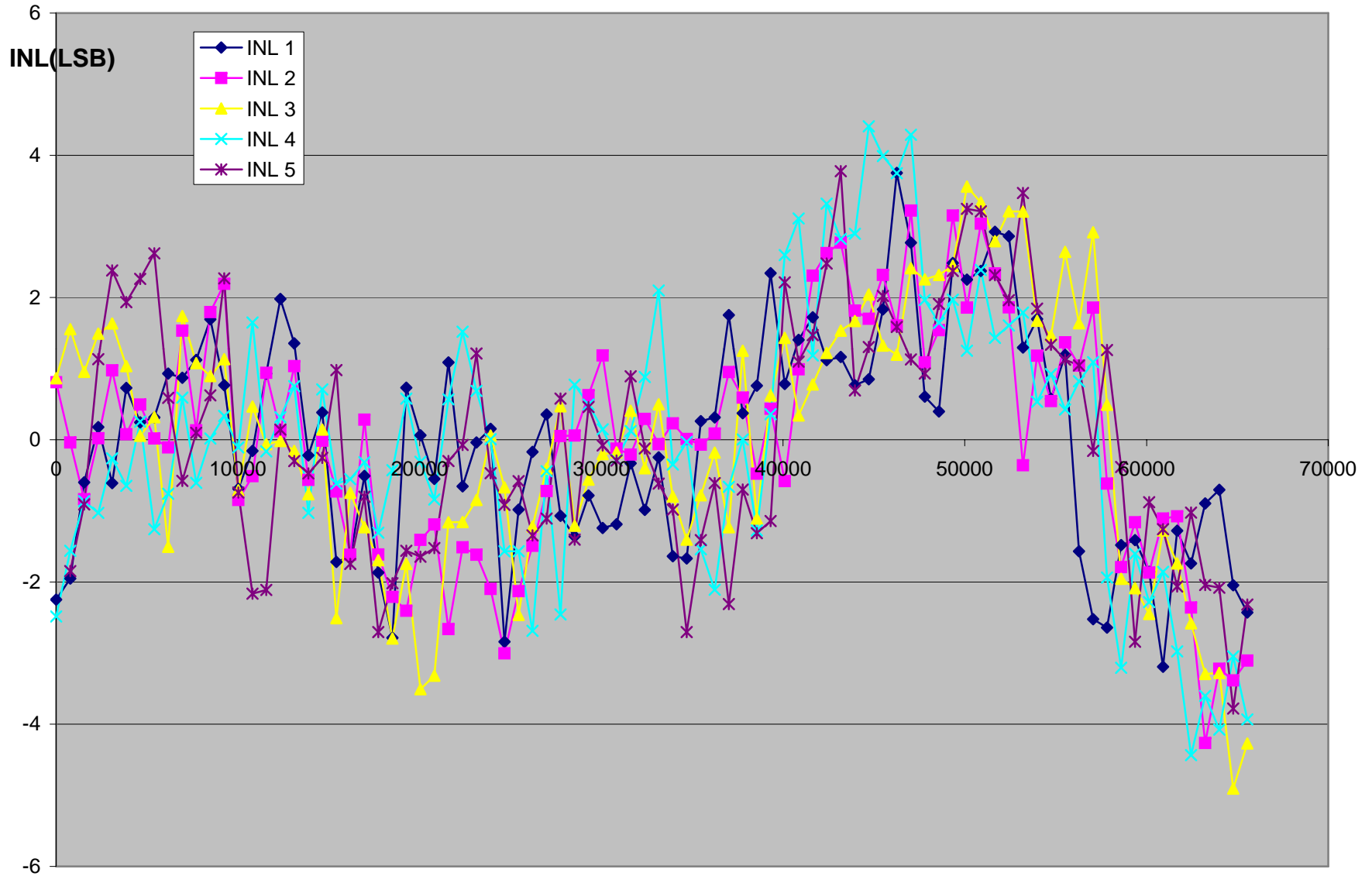
=> Encore beaucoup de travail

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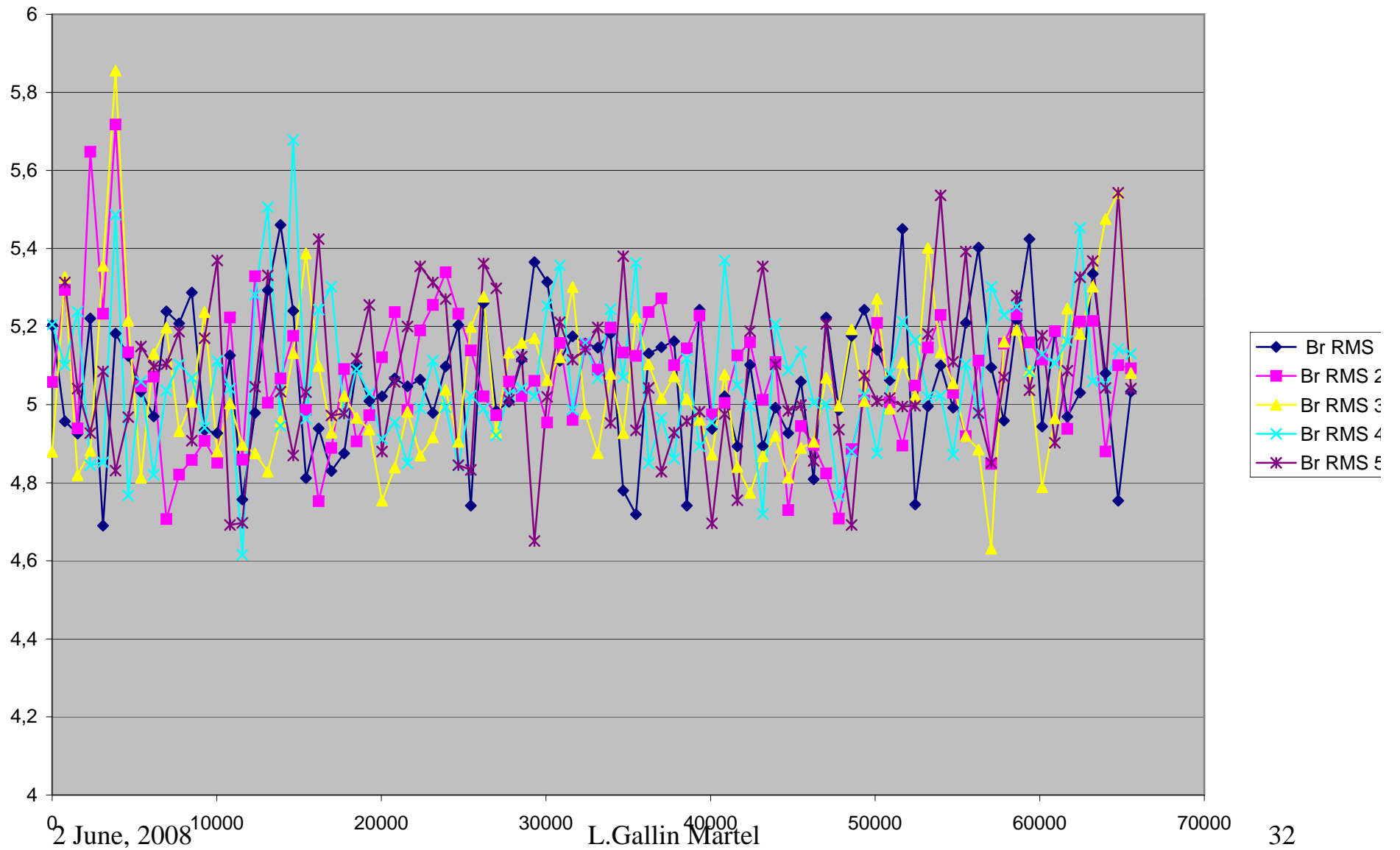
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# Testing results2



# OUTPUT Codes rms noise



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