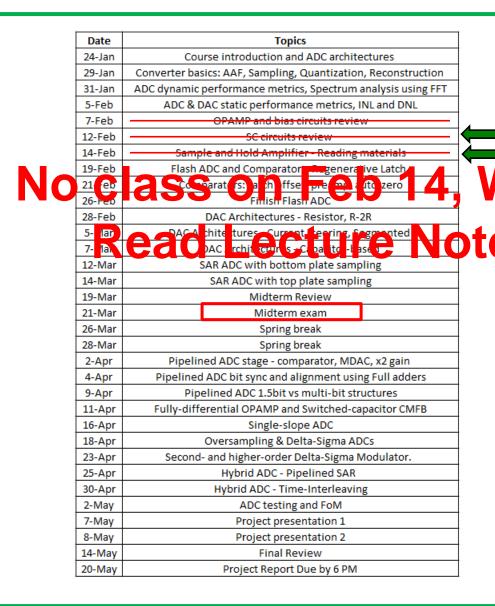
# EE288 Data Conversions/Analog Mixed-Signal ICs Spring 2018

Lecture 6: Performance Metrics

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## Course Schedule – Subject to Change



\*Midterm Exam dates are approximate and subject to change with reasonable notice.

**Performance Metrics** 

**EE223 Review** 

#### **Announcements**

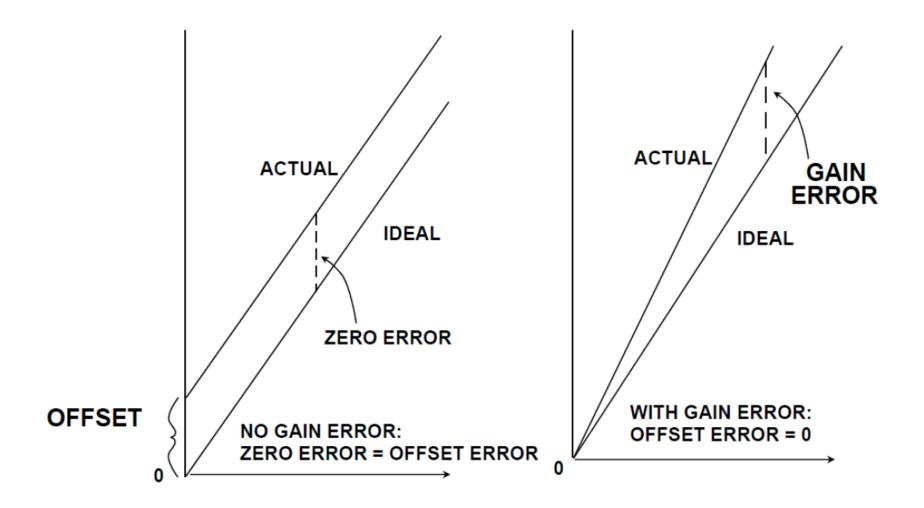
- No class this Wednesday
  - ISSCC conference

- Reading materials posted in Canvas file section
  - Lecture 7 EE223 Review
- Homework posted in Canvas Due Feb 26, Monday
  - 4-bit Flash ADC

# Agenda

- ADC Performance Metrics
  - Static
  - Dynamic

## **Offset and Gain Errors**



#### **Performance Parameters**

#### DC Performance

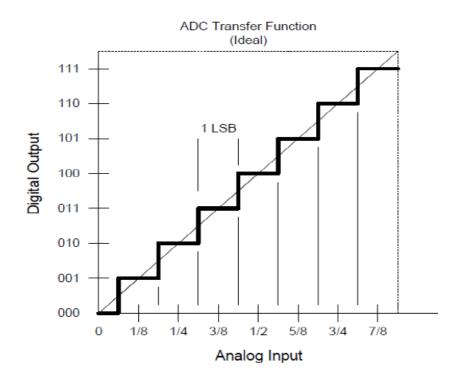
- Differential Non-Linearity (DNL)
- Integral Non-Linearity (INL)

#### AC Performance

- Harmonic Distortion
- Worst Harmonic
- Total Harmonic Distortion (THD)
- Total Harmonic Distortion Plus Noise (THD + N)
- Signal-to-Noise-and-Distortion Ratio (SINAD, or S/N +D)
- Effective Number of Bits (ENOB)
- Signal-to-Noise Ratio (SNR)
- Analog Bandwidth (Full-Power, Small-Signal)
- Spurious Free Dynamic Range (SFDR)
- Two-Tone Intermodulation Distortion
- Noise Power Ratio (NPR) or Multitone Power Ratio (MPR)

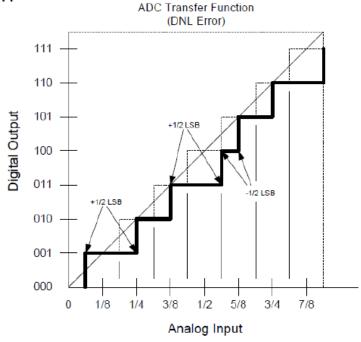
#### Ideal DC Characteristic

- Ideal ADC code transitions are exactly 1 LSB apart.
- For an N-bit ADC, there are  $2^N$  codes. (1 LSB = FS/ $2^N$ )
- For this 3-bit ADC, 1 LSB =  $(1V/2^3 = 1/8)$
- Each "step" is centered on an eighth of full scale



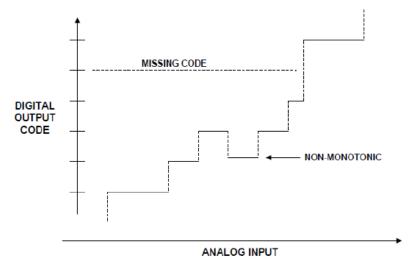
## **Differential Non Linearity (DNL)**

- The deviation of an actual code width from the ideal 1 LSB code
- DNL error is measured in LSBs
- Results in narrow or wider code widths than ideal
- Results in additive noise/spurs beyond the effects of quantization



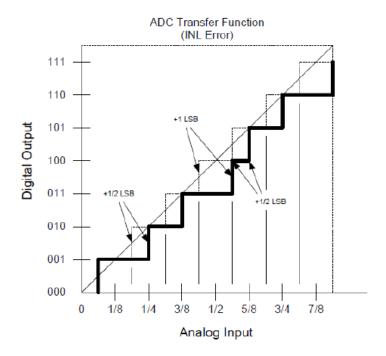
## **DNL Effects**

- Missing Codes An ADC has missing codes if an infinitesimally small change in voltage causes a change in result of two codes, with the intermediate code never being set. A DNL of -1.0 LSB indicates the ADC has missing.
- Non-Monotonicity An ADC is monotonic if it continually increases conversion result with an increasing voltage (and vice versa). A nonmonotonic ADC may give a lower conversion result for a higher input voltage, which may also mean that the same conversion may result from two separate voltage ranges.



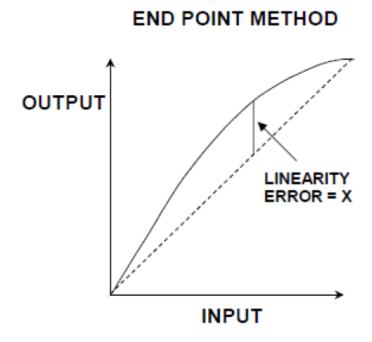
## **Integral Non Linearity (INL)**

- The deviation of an actual code transition point from its ideal position on a straight line drawn between the end points of the transfer function.
- INL is calculated after offset and gain errors are removed
- Results in additive harmonics and spurs

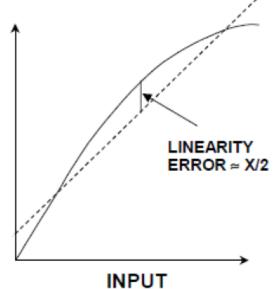


#### **INL Measurement Methods**

- End Point Method: the deviation is measured from the straight line through the origin and the full-scale point (after gain adjustment)
- Best Straight Line Method: the best fit straight line is drawn through the transfer characteristic of the device using standard curve fitting techniques, and the maximum deviation is measured from this line

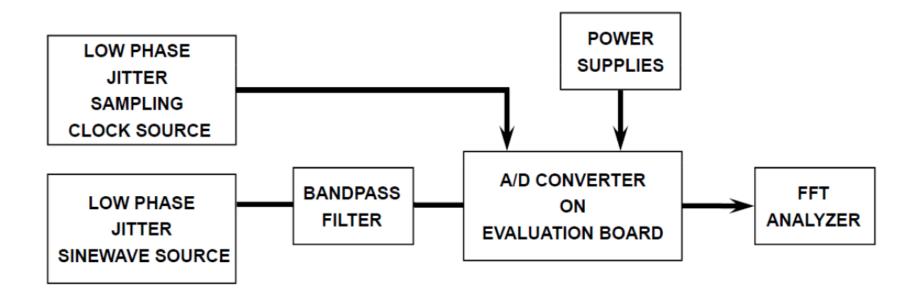


#### BEST STRAIGHT LINE METHOD



## **Dynamic Testing of ADC**

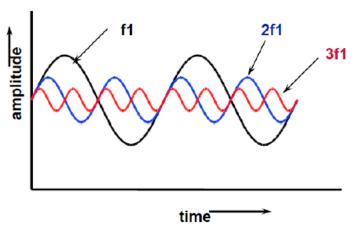
 A Fast Fourier Transform (FFT) analyzer is used to measure dynamic performance



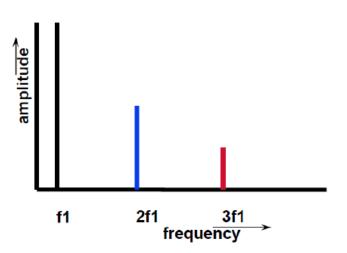
## **FFT**

■ The Fast Fourier Transform converts a signal from time

domain....



....to frequency domain



## SINAD, ENOB, SNR

#### SINAD (Signal to Noise and Distortion Ratio)

 The ratio of the rms signal amplitude to the mean value of the root-sum-squares (RSS) of all other spectral components, including harmonics, but excluding DC

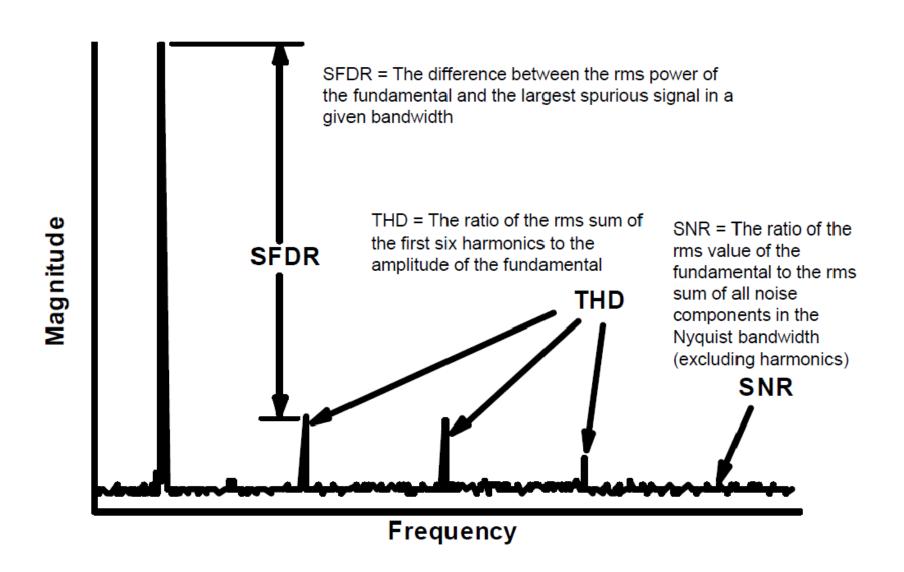
#### ENOB (Effective Number of Bits)

$$ENOB = \frac{SINAD - 1.76 dB}{6.02}$$

#### SNR (Signal to Noise Ratio)

 The ratio of the rms signal amplitude to the mean value of the root-sum-squares (RSS) of all other spectral components, excluding the first five harmonics and DC

## SFDR, THD, SNR



## **SQNR**

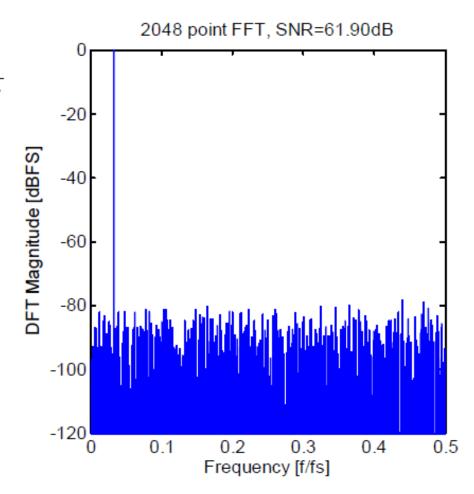
$$SQNR = \frac{Signal\ Power}{Quantization\ Noise\ Power}$$

$$= \frac{\frac{1}{2} \left(\frac{V_{FS}}{2}\right)^2}{\frac{1}{12} \left(\frac{V_{FS}}{2^N}\right)^2} = \frac{3}{2} \cdot 2^{2N}$$

$$=6.02 \cdot N + 1.76 \text{ [dB]}$$

$$=6.02 \cdot 10 + 1.76 \text{ [dB]}$$

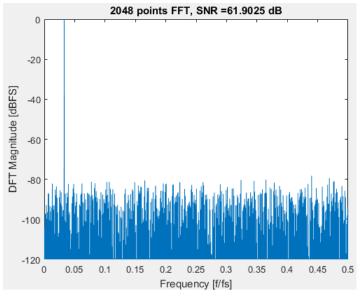
= 61.9 dB



## FFT Spectrum of Quantized Signal

```
clear all; clc; close all;
N = 2048;
                        % FFT size
                       % Signal bin
cvcles = 67;
fs = 1000;
                       % Sampling rate
                       % Signal frequency
fx = cycles*fs/N;
B = 10;
                        % ADC resolution
LSB = 2/2^B;
                        % LSB size
t = 0:N-1;
                       % time sequence
x = cos(2*pi*fx/fs*t); % Signal sequence
x = round(x/LSB)*LSB; % Quantized signal
s = abs(fft(x)) + eps;
                      % Take FFT
s = s(1:end/2)/N*2; % Take half of the spectrum and normalize
sigbin = cycles + 1;
noise = [s(l:sigbin-l), s(sigbin+l:end)];
SNR = 10*log10(s(sigbin)^2/sum(noise.^2));
fprintf('SNR = %0.4f\n',SNR);
s = 20*log10(s);
                       % dB relative to full-scale
f = [0:N/2-1]/N;
                       % frequency vector
plot(f, s);
axis([0 0.5 -120 0]);
xlabel('Frequency [f/fs]');
ylabel('DFT Magnitude [dBFS]');
```

title(strcat(num2str(N), 'points FFT, SNR = ', num2str(SNR), 'dB'));

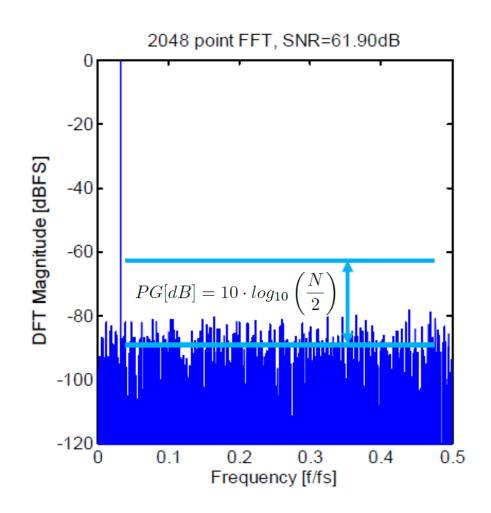


fft\_p55.m

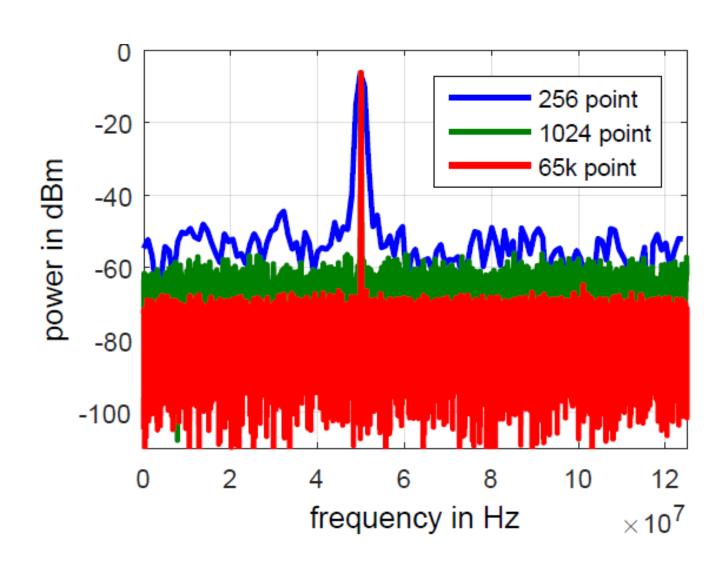
#### **FFT Noise Floor**

$$N_{floor} = -61.9 \text{ dBc} - 10 \log \left( \frac{2048}{2} \right)$$
  
= -61.9 dBc - 30.1 dB  
= -92 dBc

- Depends on FFT size
- Plot is "useless" if FFT size is not specified



## **FFT Processing Gain**



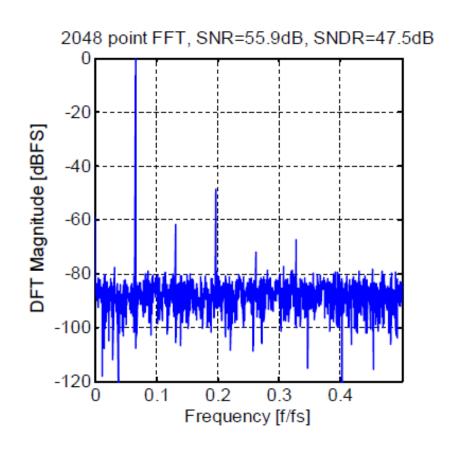
#### **SNDR and ENOB**

Definition

$$SNDR = \frac{\text{Signal Power}}{\text{Noise and Distortion Power}}$$

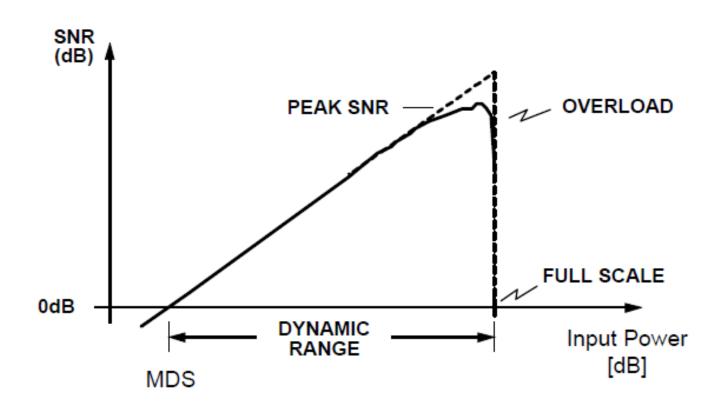
- Noise and distortion power includes all bins except DC and signal
- Effective number of bits

$$ENOB = \frac{SNDR(dB)-1.76dB}{6.02dB}$$



## **Dynamic Range**

$$\text{DR} = \frac{\text{Maximum Signal Power}}{\text{Minimum Detectable Signal}} \geq \text{SNR}_{peak}$$



#### **SDR and THD**

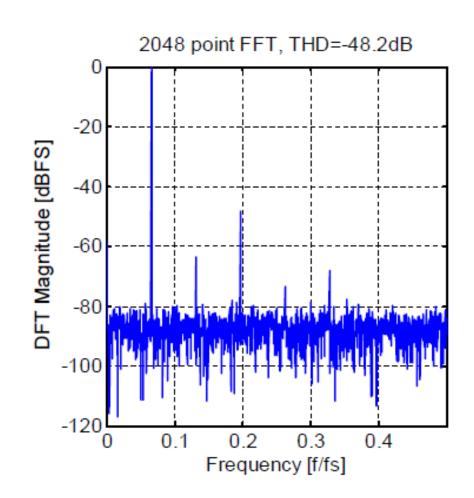
Signal-to-distortion ratio

$$SDR = \frac{Signal\ Power}{Total\ Distortion\ Power}$$

Total harmonic distortion

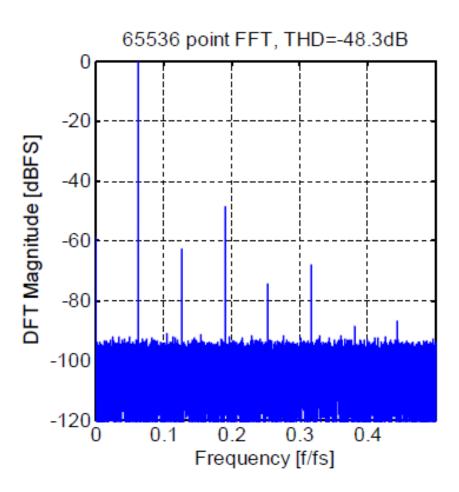
$$THD = \frac{Total \ Distortion \ Power}{Signal \ Power} = \frac{1}{SDR}$$

 By convention, total distortion power consists of 2<sup>nd</sup> through 7<sup>th</sup> harmonic



## **Lowering the Noise Floor**

 Increasing the FFT size let's us lower the noise floor and reveal low level harmonics



## **Aliasing**

 Harmonics can appear at "arbitrary" frequencies due to aliasing

$$f_1 = f_x = 0.3125 f_s$$
  
 $f_2 = 2 f_1 = 0.6250 f_s \rightarrow 0.3750 f_s$   
 $f_3 = 3 f_1 = 0.9375 f_s \rightarrow 0.0625 f_s$   
 $f_4 = 4 f_1 = 1.2500 f_s \rightarrow 0.2500 f_s$   
 $f_5 = 5 f_1 = 1.5625 f_s \rightarrow 0.4375 f_s$ 

