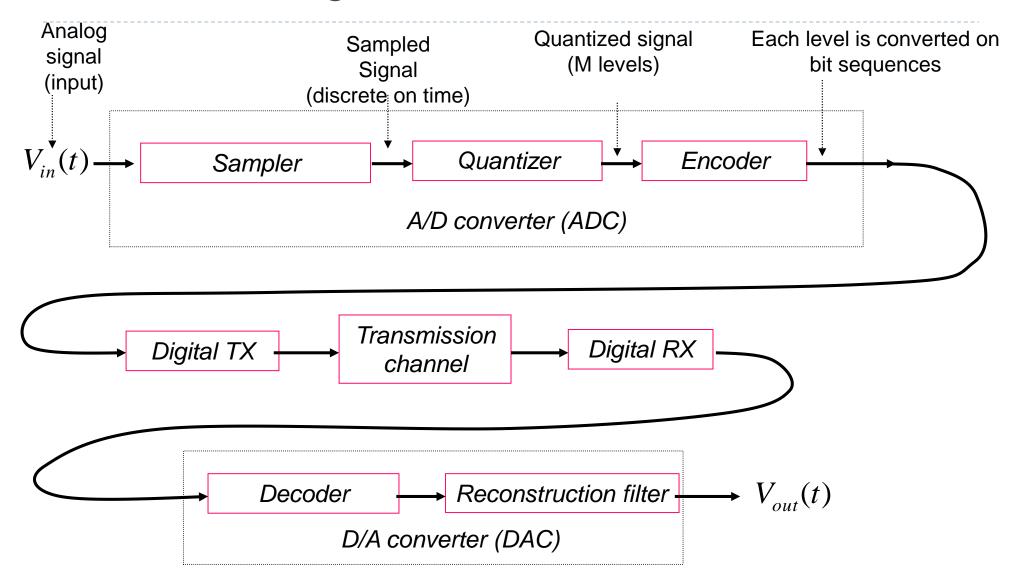
LAB#I

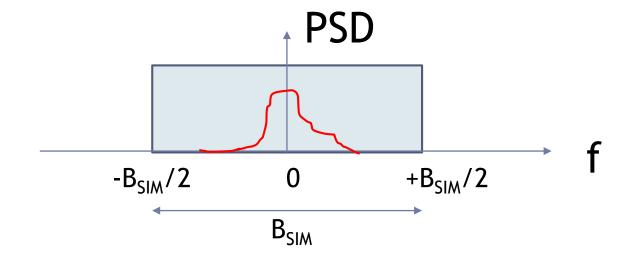
PCM block diagram



SAMPLING AND SIMULATION BANDWIDTH

Rule: satisfy sampling theorem $f_c > 2B_{signal}$

$$T_C = \frac{1}{f_c} \qquad \Longrightarrow \qquad B_{SIM} = \frac{1}{T_c}$$



WHERE IS THE SAMPLER?

- You will start with a sampled signal
 - Software generated by Matlab (uniform distribution)
 - Taken from an audio file

SNR MEASUREMENT

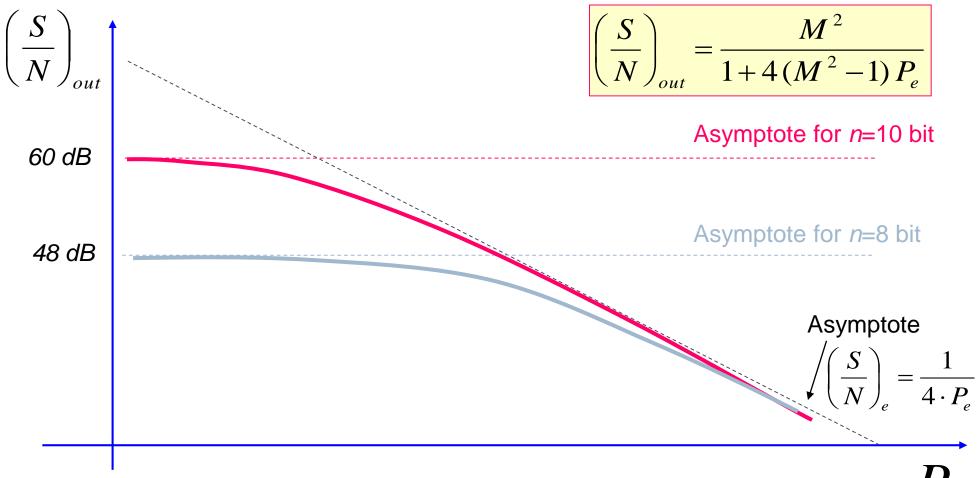
- Let's use the SNR definition applied to our system:
 - ratio between signal power and noise (error) power
- It can be obtained as the ratio between variances:

$$\left(\frac{S}{N}\right) = \frac{\sigma_{V_{in}}^2}{\sigma_e^2}$$

where

$$e = V_{out} - V_{in}$$

PCM performance (signal with uniform pdf)



 P_{e}

MATLAB COMMAND - I

- SIGNAL WITH UNIFORM PDF GENERATOR
 - sig=rand(sz1,sz2);
- QUANTIZER
 - [index,quants]=quantiz(sig,partition,codebook);
- ENCODER
 - words=de2bi(index,nbit);
- BINARY SYMMETRIC CHANNEL
 - outdata=bsc(indata,probability);
- DECODER
 - index=bi2de(words);

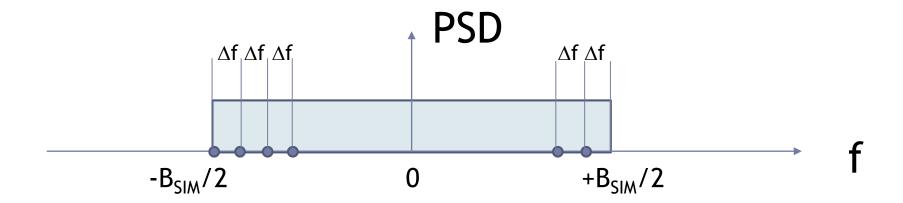
MATLAB COMMAND - II

- PDF plot
 - histogram(x,nbin,'Normalization', 'pdf');
- PSD plot
 - psd=abs(fft(x)).^2;
 - plot(f,fftshift(psd));

FFT

N_{FFT} frequency points

Frequency spacing: $\Delta f = B_{SIM}/N_{FFT}$



$$f=[B_{SIM}/2:\Delta f:B_{SIM}/2-\Delta f];$$