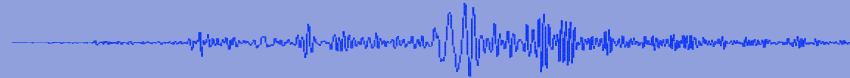


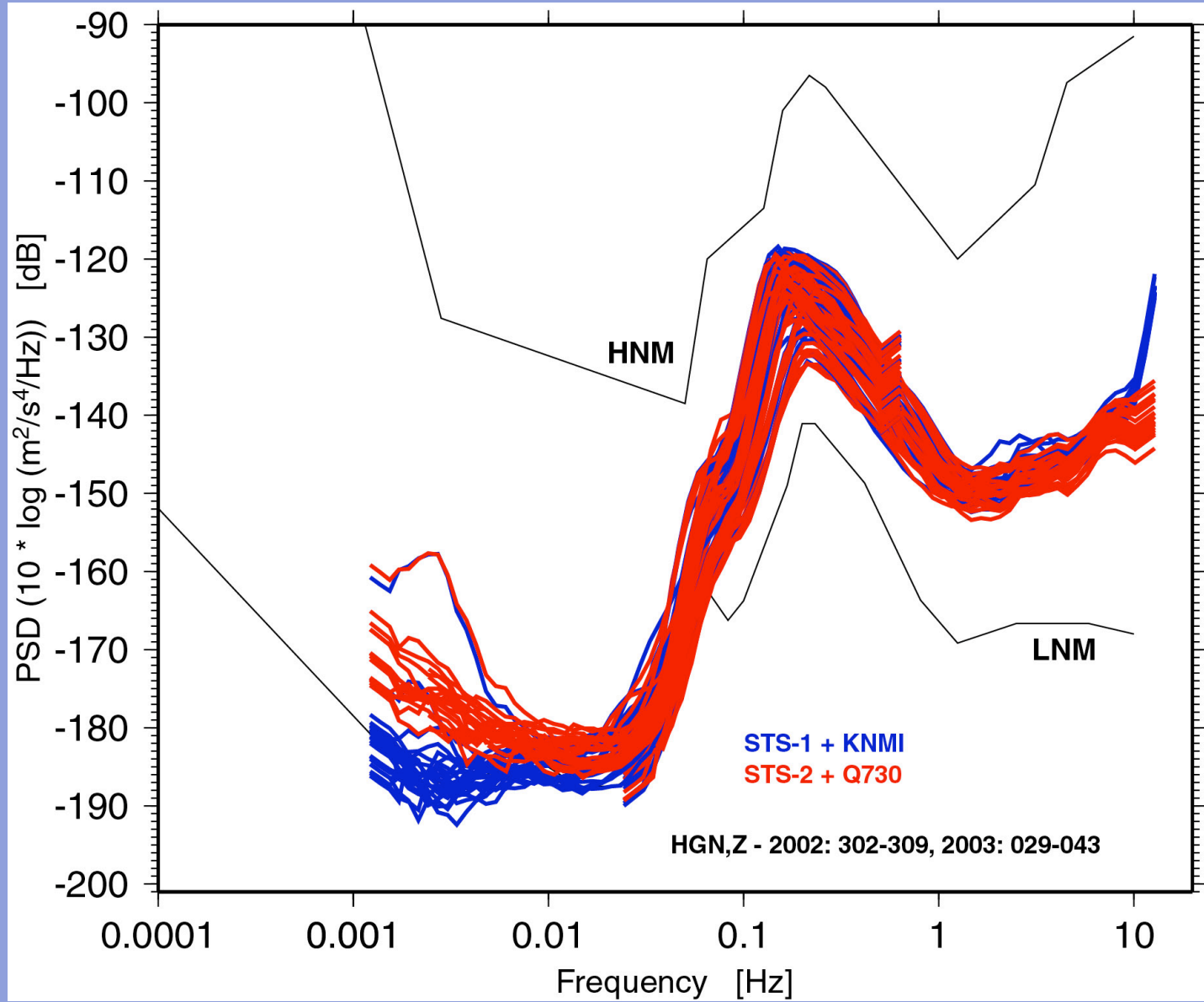
# Digitizers and dynamic range

Reinoud Sleeman  
ORFEUS Data Center  
sleeman @ knmi.nl

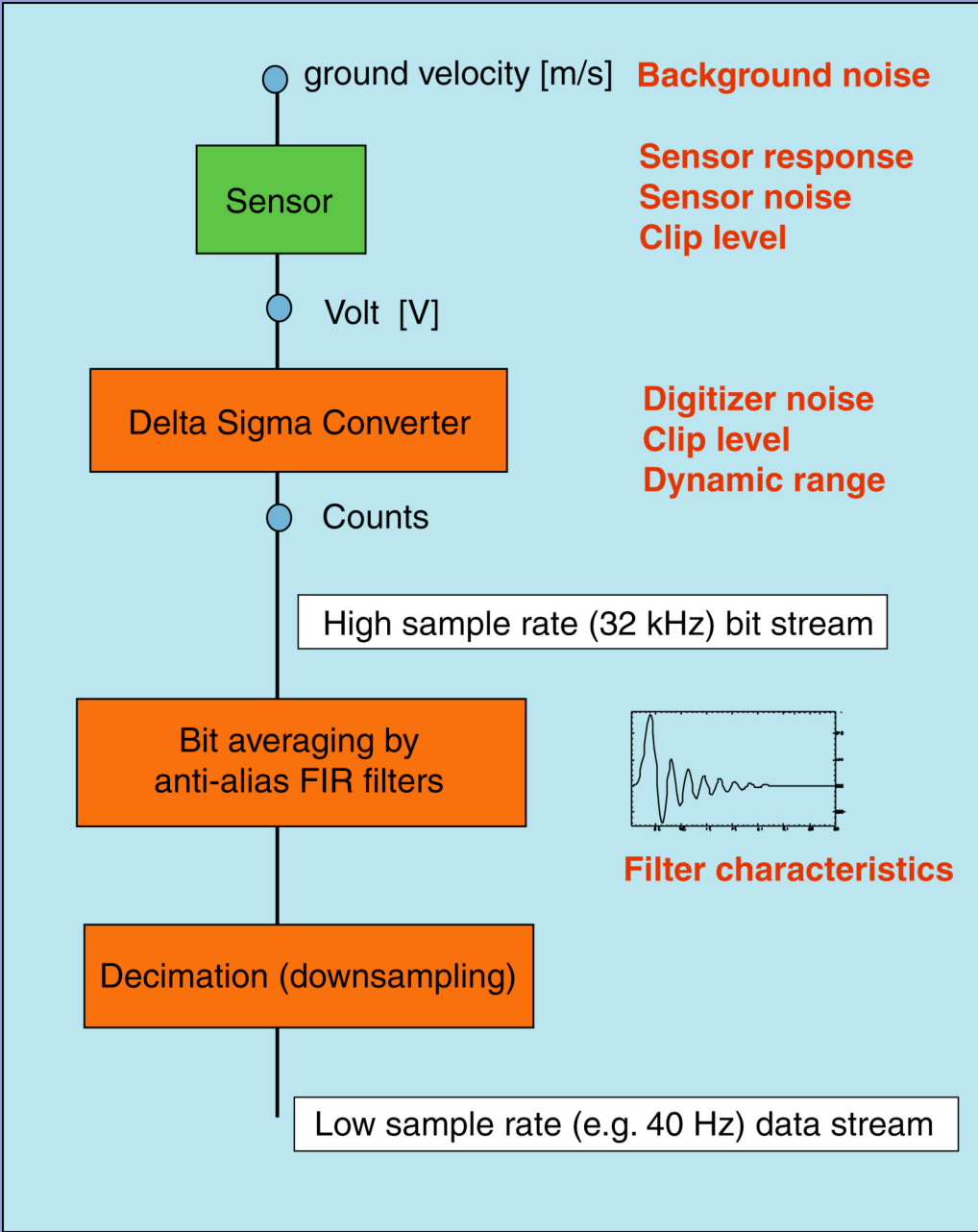


**IRIS / ORFEUS Workshop**  
**Understanding and Managing**  
**Information from Seismological Networks**

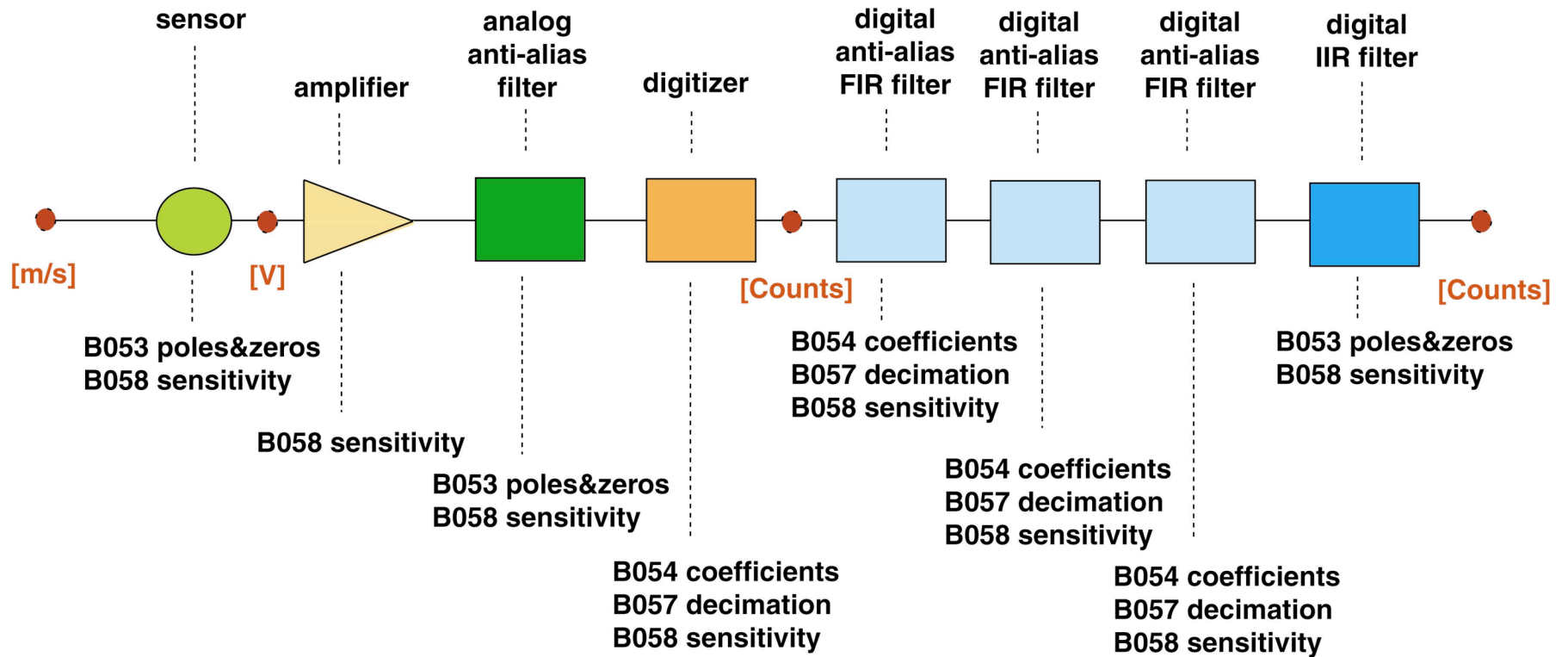
28 Feb – 4 Mar 2005, Palmanova, Italy



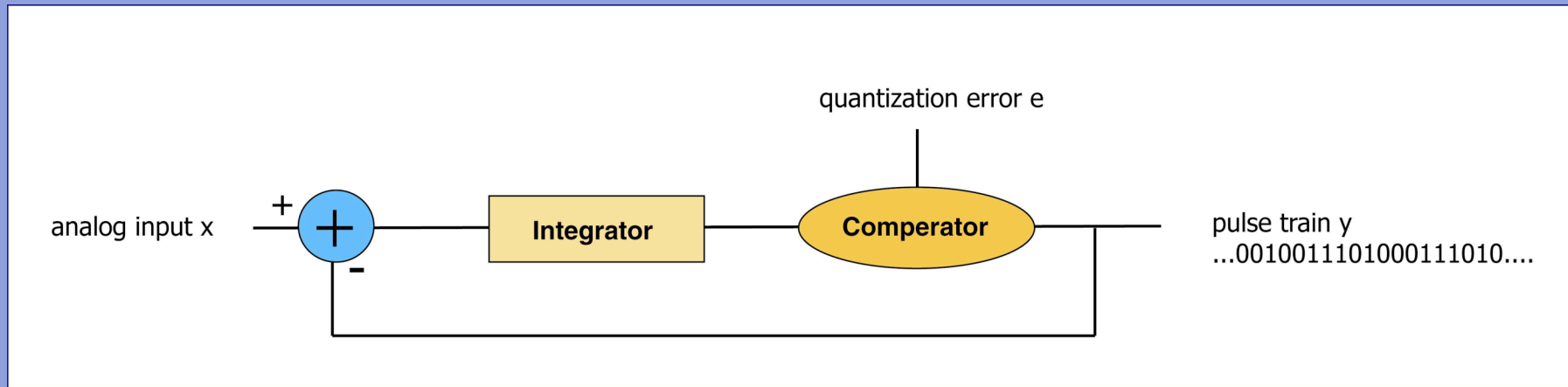
# Seismograph system



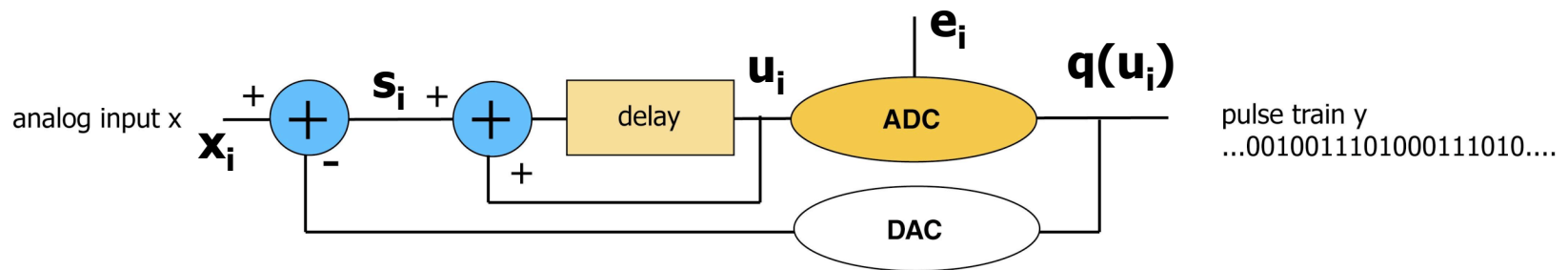
## Acquisition system block diagram and SEED response blockettes



# Oversampled Delta-Sigma A/D Digitizer (one-bit noise shaping converter)



<b>Comperator:</b>	<b>ADC or quantizer</b>
<b>Feedback:</b>	<b>average of y follows the average of x</b>
<b>Integrator:</b>	<b>accumulates the quantization error e over time</b>
<b>Pulse train:</b>	<b>pulse density representation of x</b>
<b>Oversampling:</b>	<b>increase of resolution</b>



$$s_i = x_i - q(u_i)$$

$$u_i = s_{i-1} + u_{i-1}$$

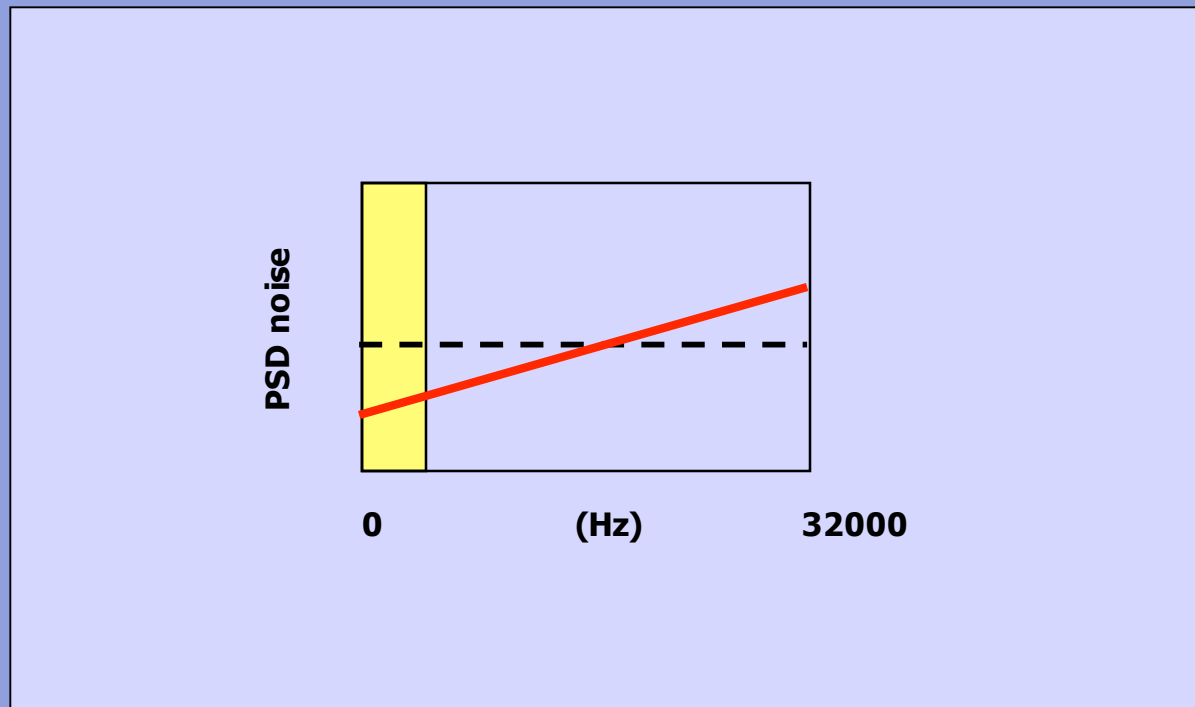
$$q(u_i) = u_i + e_i$$

$$\left. \begin{array}{l} s_i = x_i - q(u_i) \\ u_i = s_{i-1} + u_{i-1} \\ q(u_i) = u_i + e_i \end{array} \right\} q(u_i) = x_{i-1} + \underbrace{(e_i - e_{i-1})}$$

**Noise shaping**

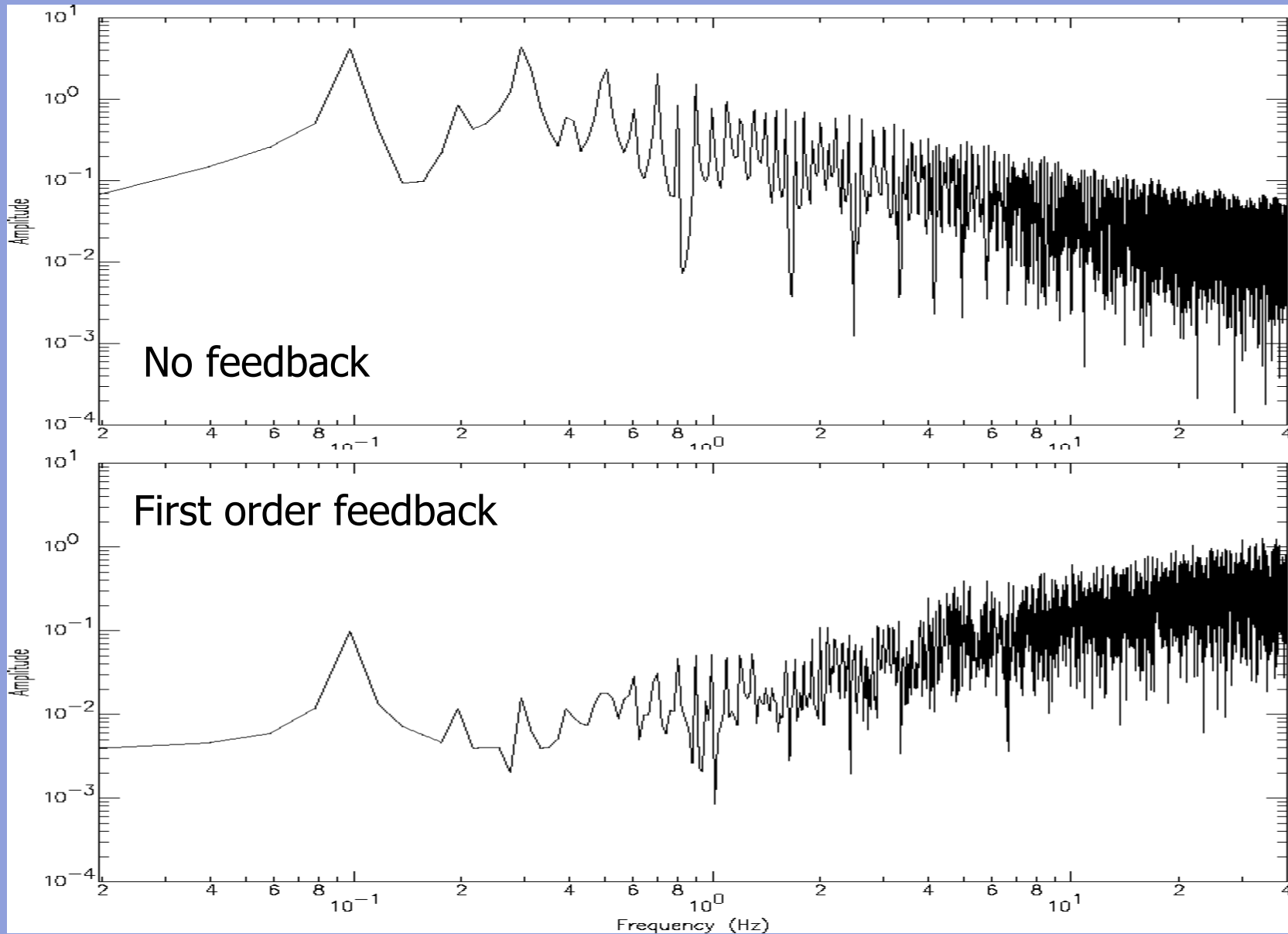
**2-nd order:**

$$q(u_i) = x_{i-1} + \underbrace{(e_i - 2e_{i-1} + e_{i-2})}$$



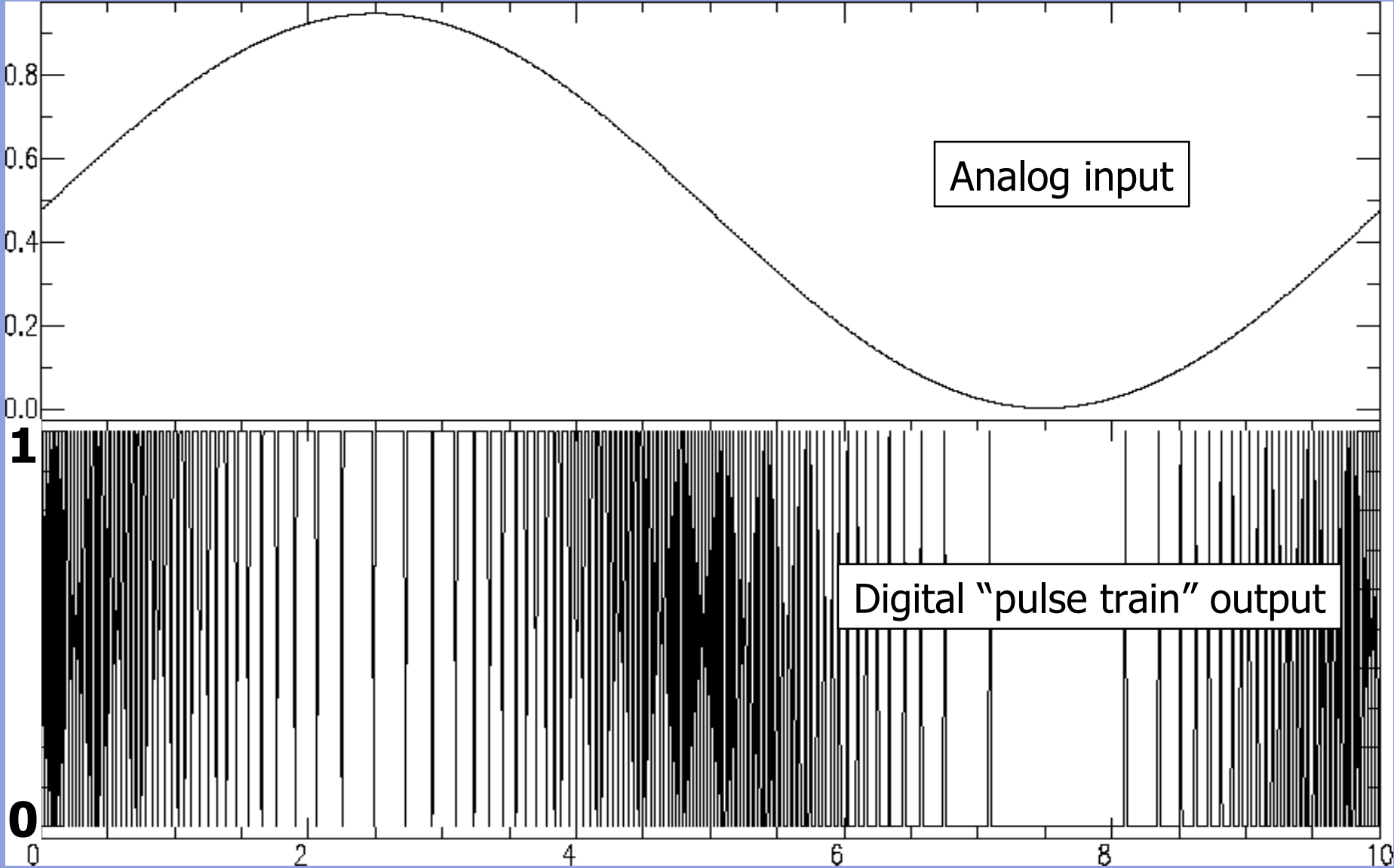
Assumption: quantization noise is white noise

# 1-bit ADC: quantization error spectrum (oversampling factor 2)

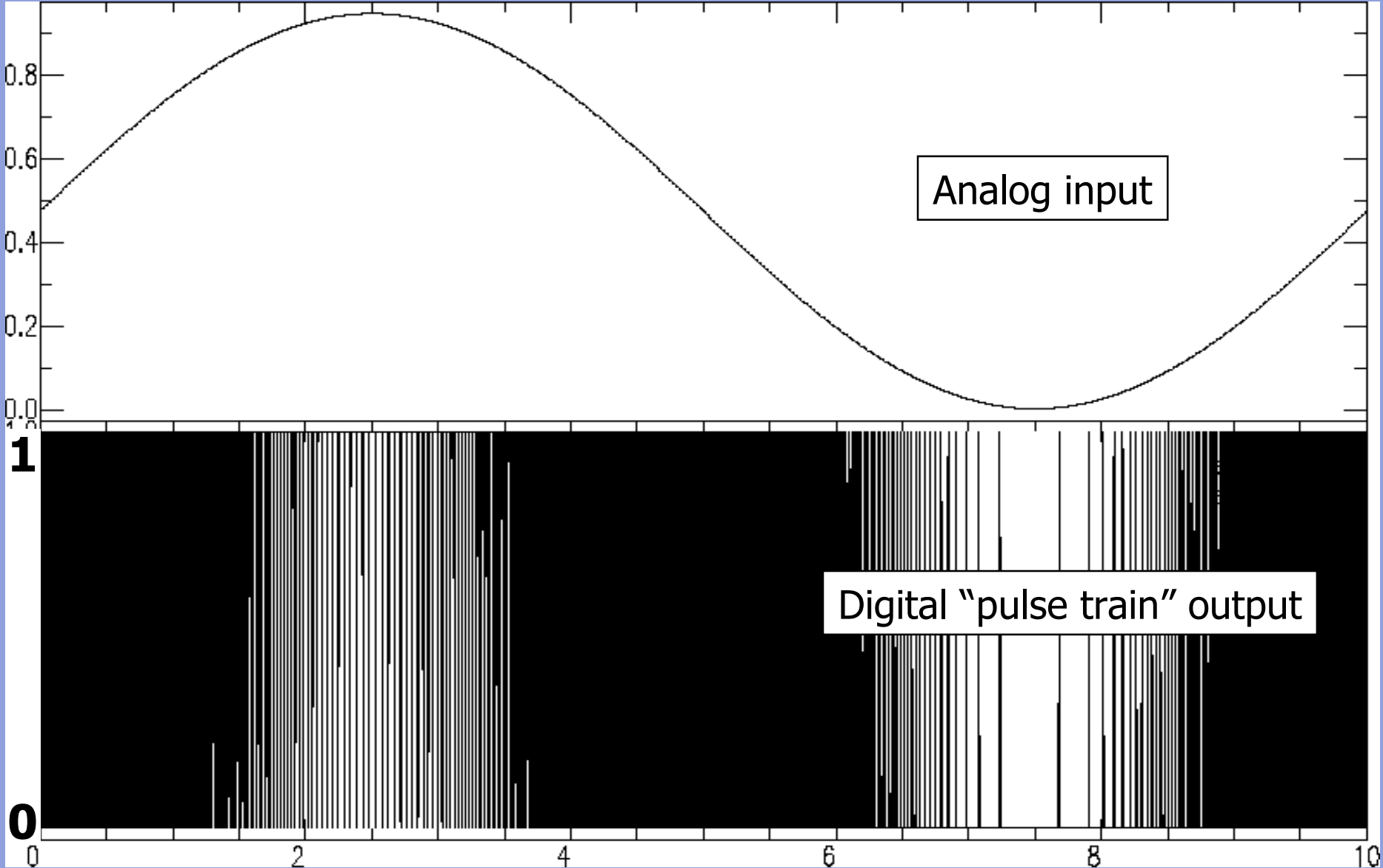




Oversampling factor: 2



Oversampling factor: 5

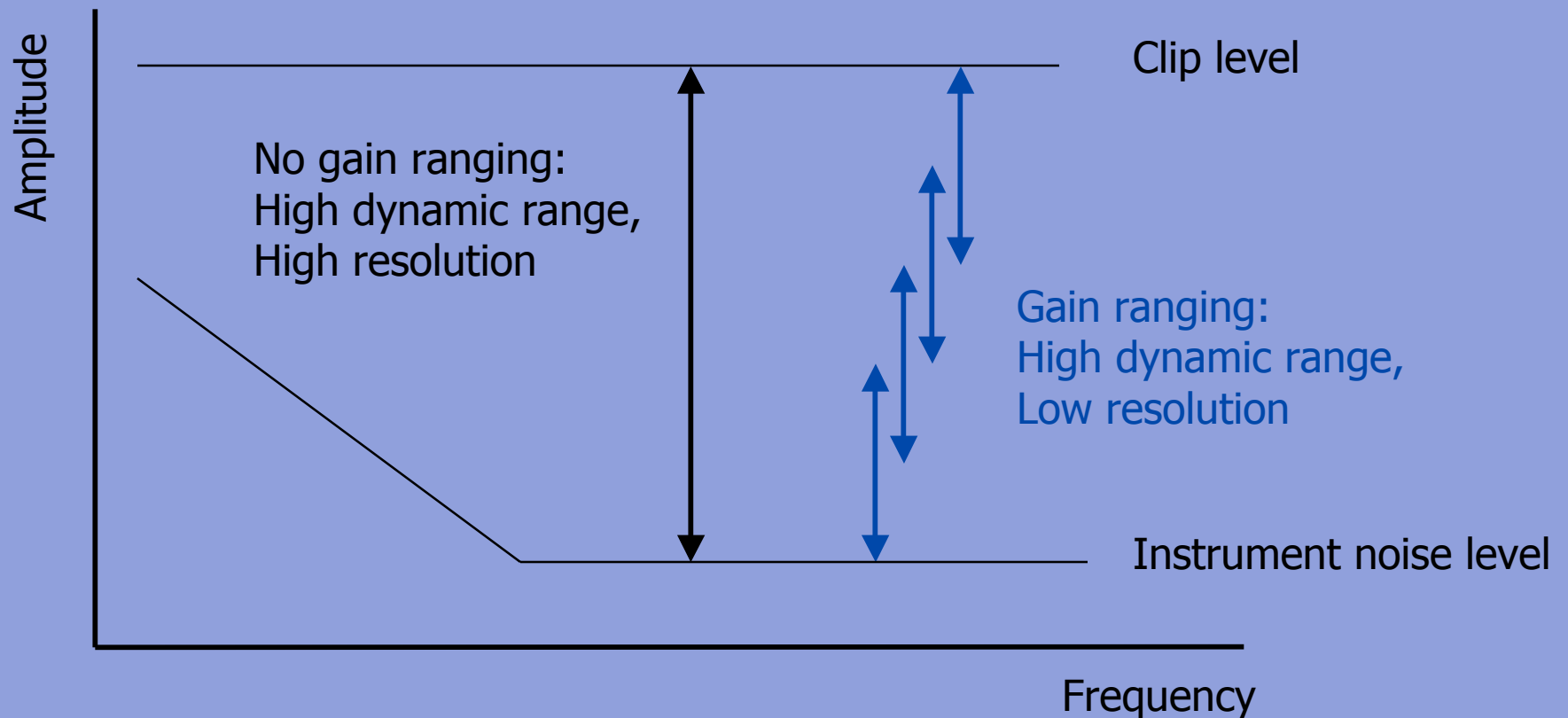


Very high sampling rate ADC with very poor resolution (1 bit)

- Feedback  $\implies$  quantization error reduction

What is the relation with  
“**dynamic range**”  
and  
“**resolution**”  
?

# Dynamic range / resolution



Dynamic range is frequency dependent

## Dynamic range of a digitizer – some theory:

**Quantization:**

$$e = q(x) - x$$

**Variance of error:**

$$e_{rms}^2 = \int_{-\infty}^{\infty} [q(x) - x]^2 p(e) de = \frac{1}{\Delta} \cdot \int_{-\Delta/2}^{\Delta/2} e^2 de = \frac{\Delta^2}{12}$$

**Number of quantization levels:**

$$\frac{2A}{\Delta} = 2^n$$

**Dynamic range:**

(Benett, 1948)

$$SNR = 10 \cdot \log \left( \frac{s_{rms}^2}{e_{rms}^2} \right)$$

$$s_{rms}^2 = A^2 / 2$$

⇒

$$SNR = 10 \cdot \log \left( \frac{A^2 / 2}{\Delta^2 / 12} \right) = 1.76 + n \cdot 6.02$$

↓  
6 dB per bit

$$SNR(f) = 10 \cdot \log \left( \frac{PSD_{\max}(f)}{PSD_{\min}(f)} \right)$$

$$PSD_{\min} = 10 \cdot \log \left( \frac{\Delta^2}{12} \cdot \frac{1}{f_{Nyquist}} \right)$$

$$PSD_{\max} = 10 \cdot \log \left( \frac{A^2}{2} \cdot \frac{1}{f_{Nyquist}} \right)$$

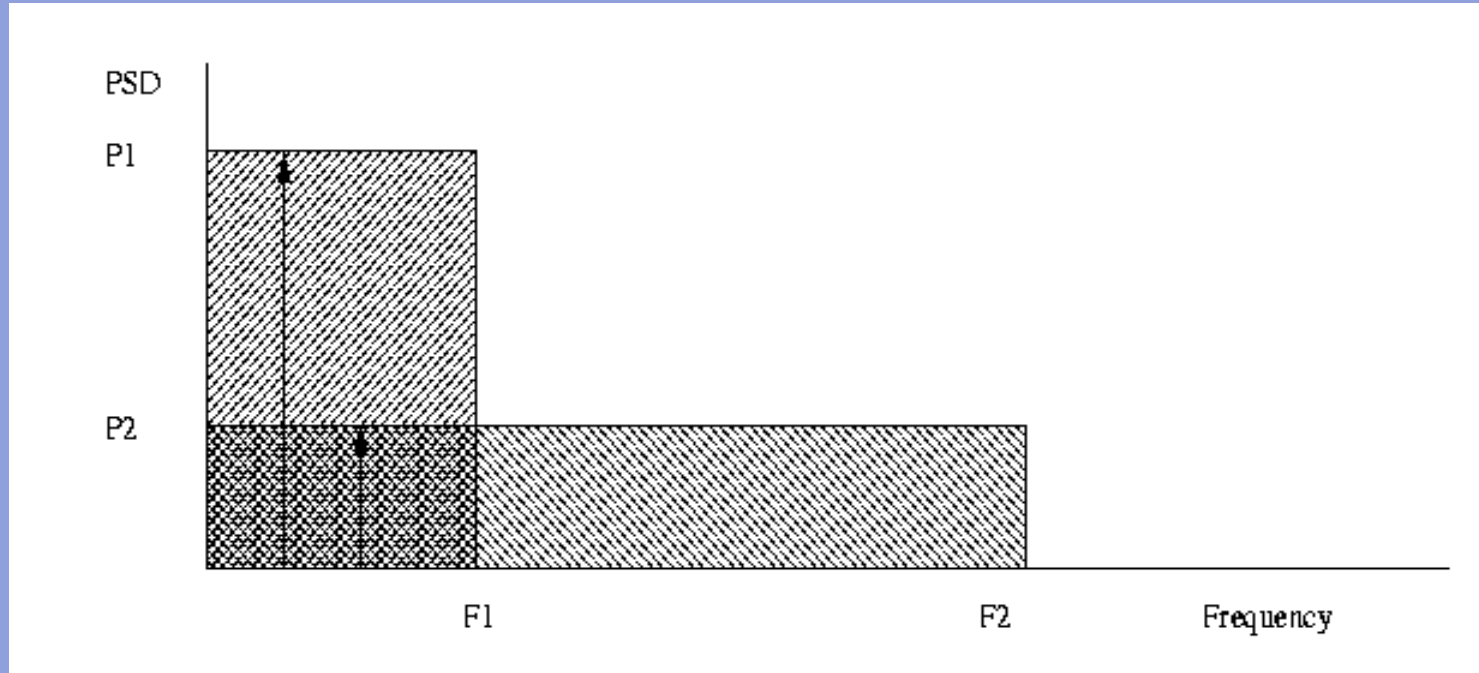
$$SNR = 10 \cdot \log \left[ \frac{V_{pp}^2}{2} \cdot \frac{1}{f_{Nyquist}} \right] - 10 \cdot \log \left[ \left( \frac{V_{pp}}{2^n} \right)^2 \cdot \frac{1}{12 \cdot f_{Nyquist}} \right]$$

rel. to 1 V<sup>2</sup>/Hz

$V_{pp}$  : peak-to-peak input in

$n$ : number of bits

$f_{Nyquist}$  : Nyquist frequency



Very high sampling rate ADC with very poor resolution (1 bit)

- Feedback  $\implies$  quantization error reduction
- Oversampling  $\implies$  higher resolution
- Bitstream averaging + decimation  $\implies$  low sample rate



**Improved dynamic range and high resolution at a lower effective sampling rate**



**Demonstration**  
**Java application**  
**of a**  
**delta-sigma modulator**

## How to measure the dynamic range of a datalogger

### ● **shorten the input and the record self noise**

- ratio of maximum peak amplitude and clip level
- specify frequency band:

$$\text{(RMS noise) / (RMS full scale sine)}$$

- PSD graph, as function of the frequency

### ● **common input signal**

- PSD graph (cross spectral analysis)

$$\text{Dynamic range} = (\text{RMS noise}) / (\text{RMS full scale sine})$$

### Shortened inputs:

Q4120:  $V_{pp} = 40 \text{ V} \Rightarrow V_{rms} = 14.1 \text{ V}$

$dt = 0.05 \text{ s}$

0.01 – 8 Hz: RMS noise 0.8  $\mu\text{V}$   
(measured)

143.8 dB

rel to  $1 \text{ V}^2 / \text{Hz}$

at 20 sps

$$\text{Dynamic range} = (\text{RMS noise}) / (\text{RMS full scale sine})$$

### Shortened inputs:

**Q4120:  $V_{pp} = 40 \text{ V} \Rightarrow V_{rms} = 14.1 \text{ V}$**

**$dt = 0.05 \text{ s}$**

**0.01 – 8 Hz: RMS noise 0.8  $\mu\text{V}$   
(measured)**



**143.8 dB**

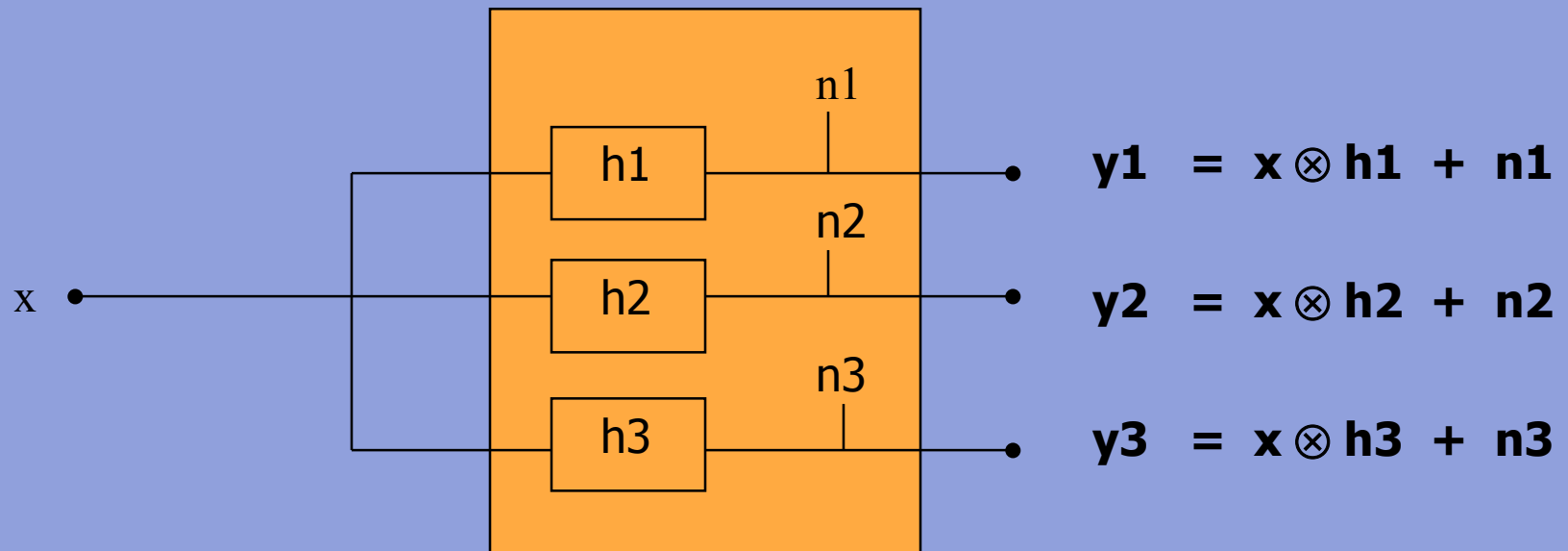
**rel to  $1 \text{ V}^2 / \text{Hz}$**

**at 20 sps**

$$SNR = 10 \cdot \log\left(\frac{A^2 / 2}{\Delta^2 / 12}\right) = 1.76 + n \cdot 6.02 \quad \longrightarrow \quad 23.7 \text{ bits}$$

# Cross spectral analysis

## Linear noise-model



### Assumptions

- (1)  $\int x \cdot n1 = \int x \cdot n2 = \int x \cdot n3 = 0$
- (2)  $n1, n2$  and  $n3$  are uncorrelated

# Auto/Cross correlation

## Time domain

$$y1 = x \otimes h1 + n1$$

$$y2 = x \otimes h2 + n2 \quad \Leftrightarrow$$

$$y3 = x \otimes h3 + n3$$

## Frequency domain

$$Y1 = X \cdot H1 + N1$$

$$Y2 = X \cdot H2 + N2$$

$$Y3 = X \cdot H3 + N3$$

$$\begin{aligned} \text{corr}(y1,y1) \quad \Leftrightarrow \quad Y1 \cdot Y1^* &= X \cdot X^* \cdot H1 \cdot H1^* + N1 \cdot N1^* = \\ &= P_{xx} \cdot H1 \cdot H1^* + P_{n1n1} = \\ &= P_{y1y1} \quad (\text{Auto Power Spectrum}) \end{aligned}$$

$$\begin{aligned} \text{corr}(y1,y2) \quad \Leftrightarrow \quad Y1 \cdot Y2^* &= X \cdot X^* \cdot H1 \cdot H2^* + N1 \cdot N2^* = \\ &= P_{xx} \cdot H1 \cdot H2^* + P_{n1n2} = \\ &= P_{y1y2} \quad (\text{Cross Power Spectrum}) \end{aligned}$$

$$P_{y_1y_1} = Y_1 \cdot Y_1^* = X \cdot X^* \cdot H_1 \cdot H_1^* + N_1 \cdot N_1^*$$

$$P_{y_1y_2} = Y_1 \cdot Y_2^* = X \cdot X^* \cdot H_1 \cdot H_2^* + \cancel{N_1 \cdot N_2^*}$$

$$P_{y_1y_3} = Y_1 \cdot Y_3^* = X \cdot X^* \cdot H_1 \cdot H_3^* + \cancel{N_1 \cdot N_3^*}$$

$$P_{y_2y_2} = Y_2 \cdot Y_2^* = X \cdot X^* \cdot H_2 \cdot H_2^* + N_2 \cdot N_2^*$$

$$P_{y_2y_3} = Y_2 \cdot Y_3^* = X \cdot X^* \cdot H_2 \cdot H_3^* + \cancel{N_2 \cdot N_3^*}$$

$$P_{y_3y_3} = Y_3 \cdot Y_3^* = X \cdot X^* \cdot H_3 \cdot H_3^* + N_3 \cdot N_3^*$$

$$P_{y_2y_3} / P_{y_1y_3} = H_2 / H_1$$

$$P_{y_2y_2} / P_{y_1y_2} = H_2 / H_1 + N_2 \cdot N_2^* / P_{y_1y_2} = P_{y_2y_3} / P_{y_1y_3} + P_{n_2n_2} / P_{y_1y_2}$$

$$P_{n_2n_2} = P_{y_2y_2} - [P_{y_1y_2} / P_{y_1y_3}] \cdot P_{y_2y_3}$$

$$P_{n1n1} = P_{y1y1} - [P_{y1y3}/P_{y2y3}] \cdot P_{y2y1}$$

$$P_{n2n2} = P_{y2y2} - [P_{y1y2}/P_{y1y3}] \cdot P_{y2y3}$$

$$P_{n3n3} = P_{y3y3} - [P_{y3y2}/P_{y1y2}] \cdot P_{y3y1}$$

Cross power spectrum of outputs 1 and 3

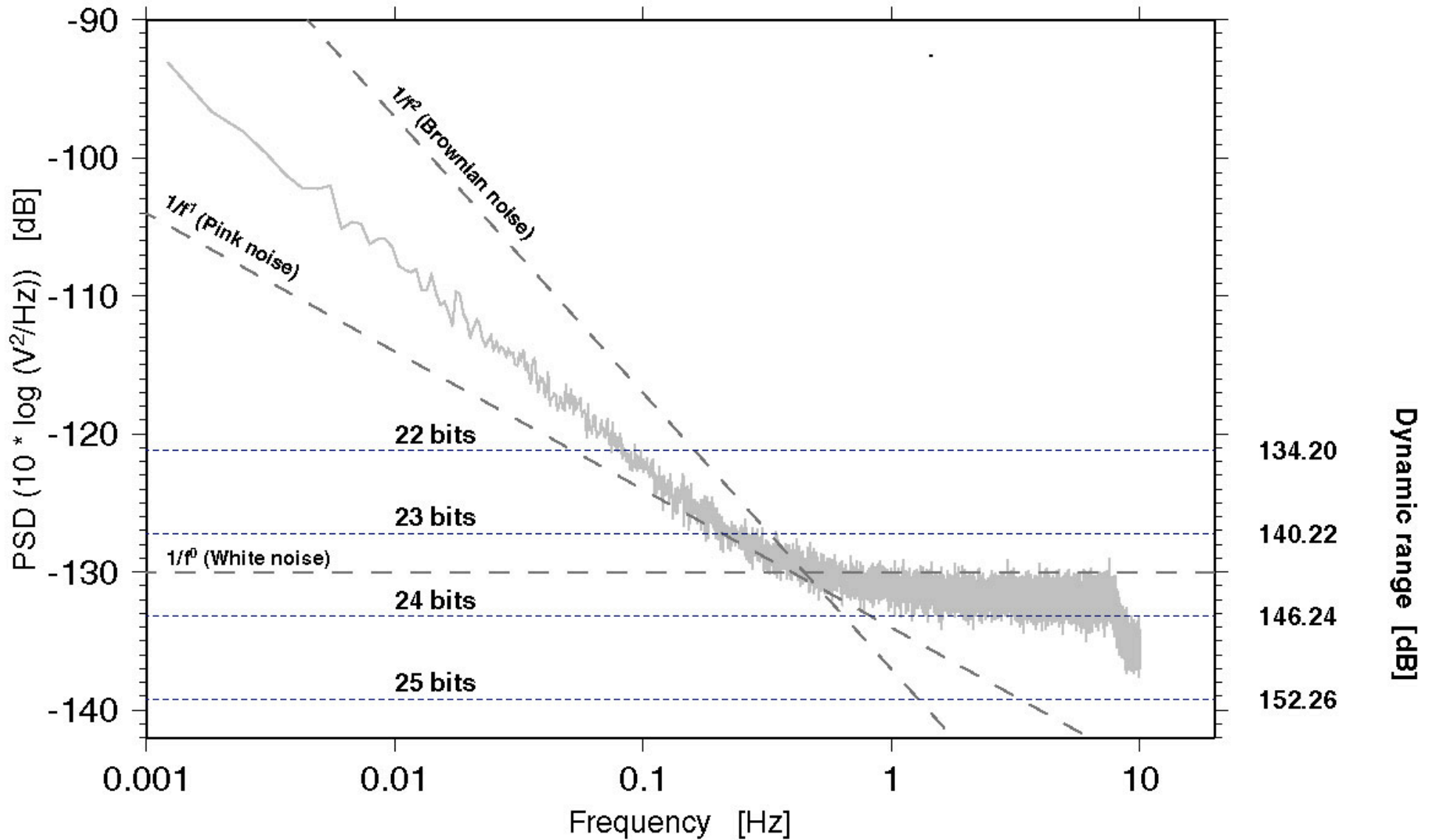
Power spectrum of 3<sup>rd</sup> digitizer-noise



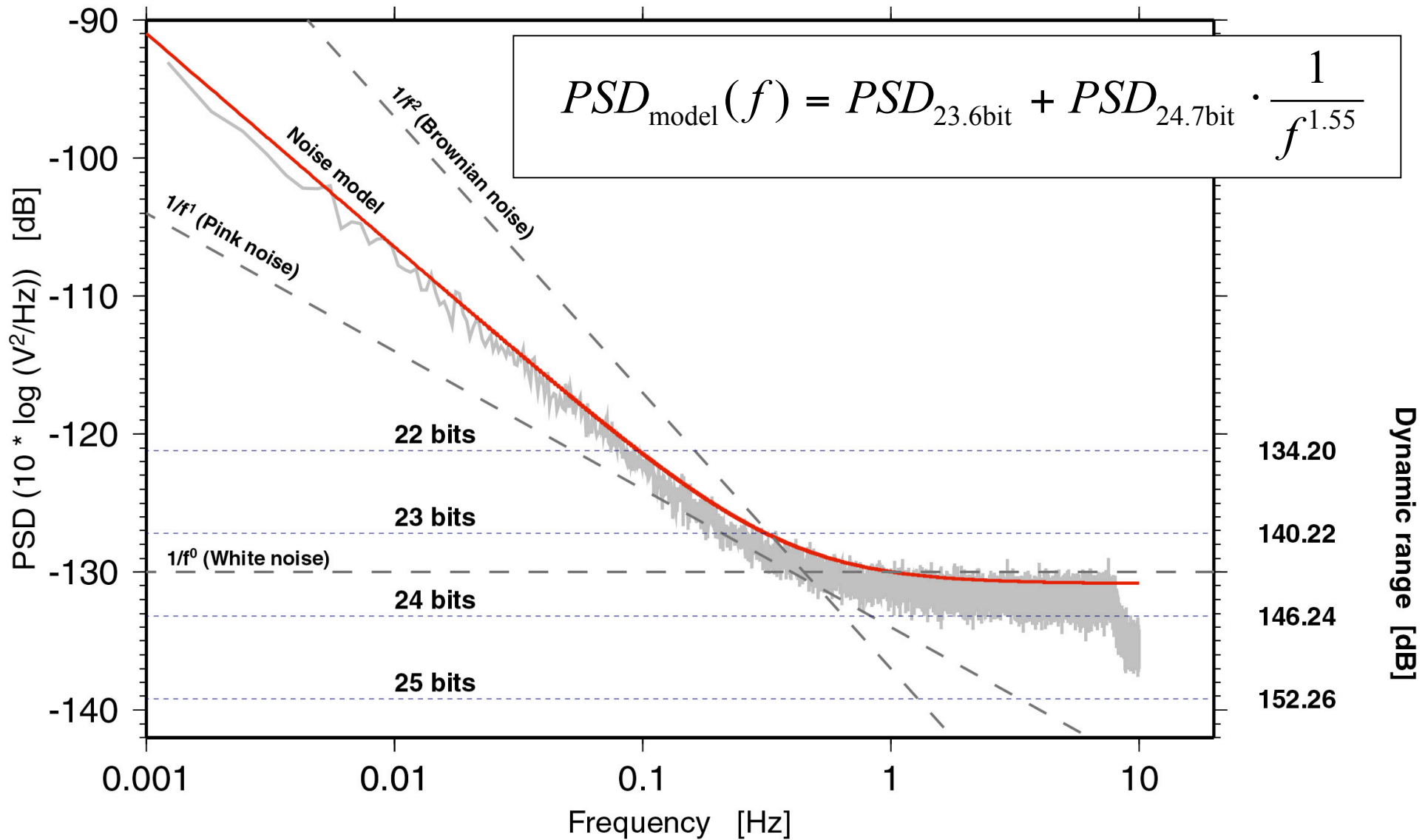
## **PSD estimation (Welsh, 1978):**

- **50 % overlapping time sections (of 2048 samples at 20 Hz)**
- **tapering (Hanning window)**
- **auto/cross correlation**
- **Fourier transform**
- **averaging over the number of time sections**

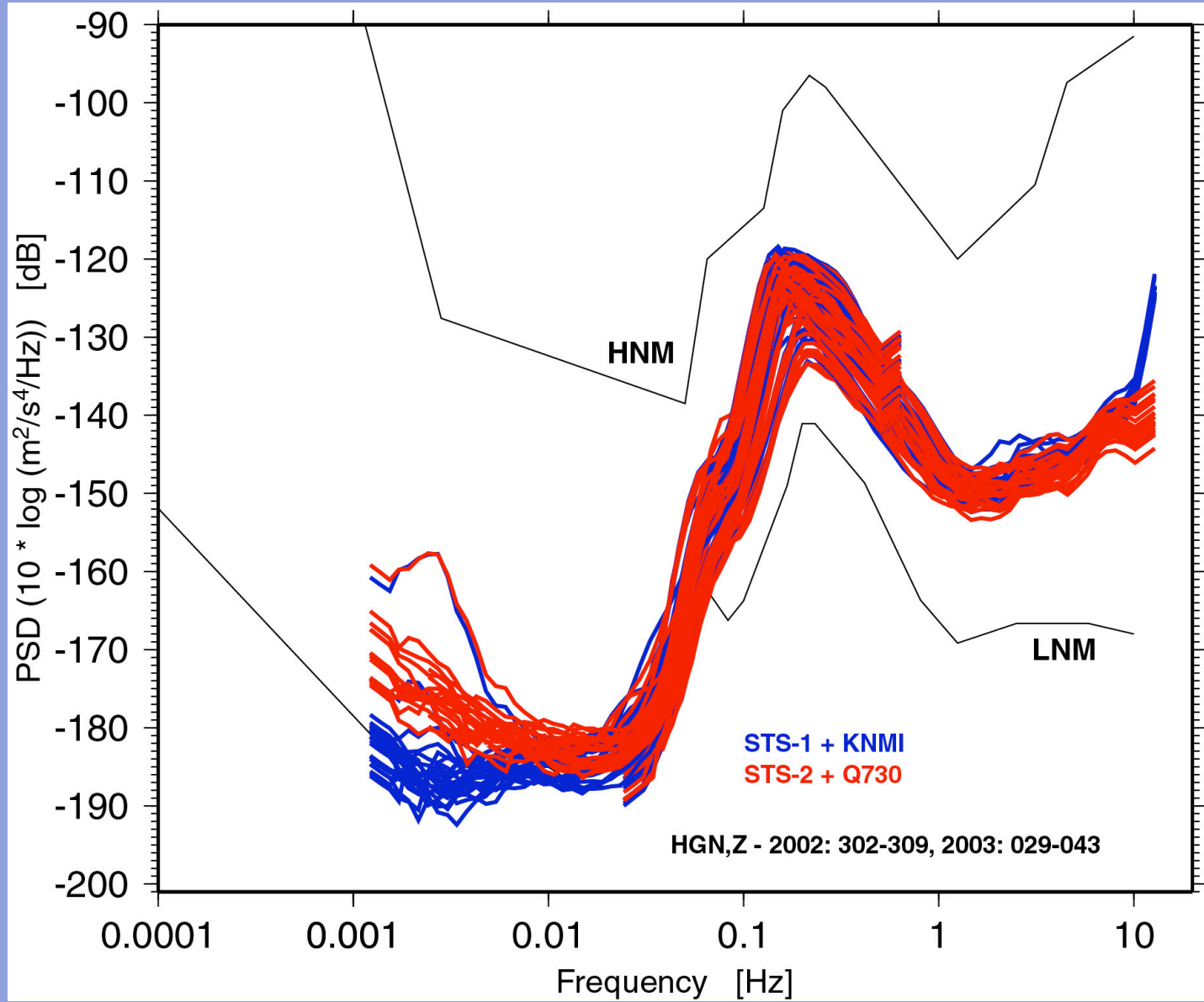
## PSD of self-noise Q4120 measured with common STS-2 vertical signal (@ 20 sps)

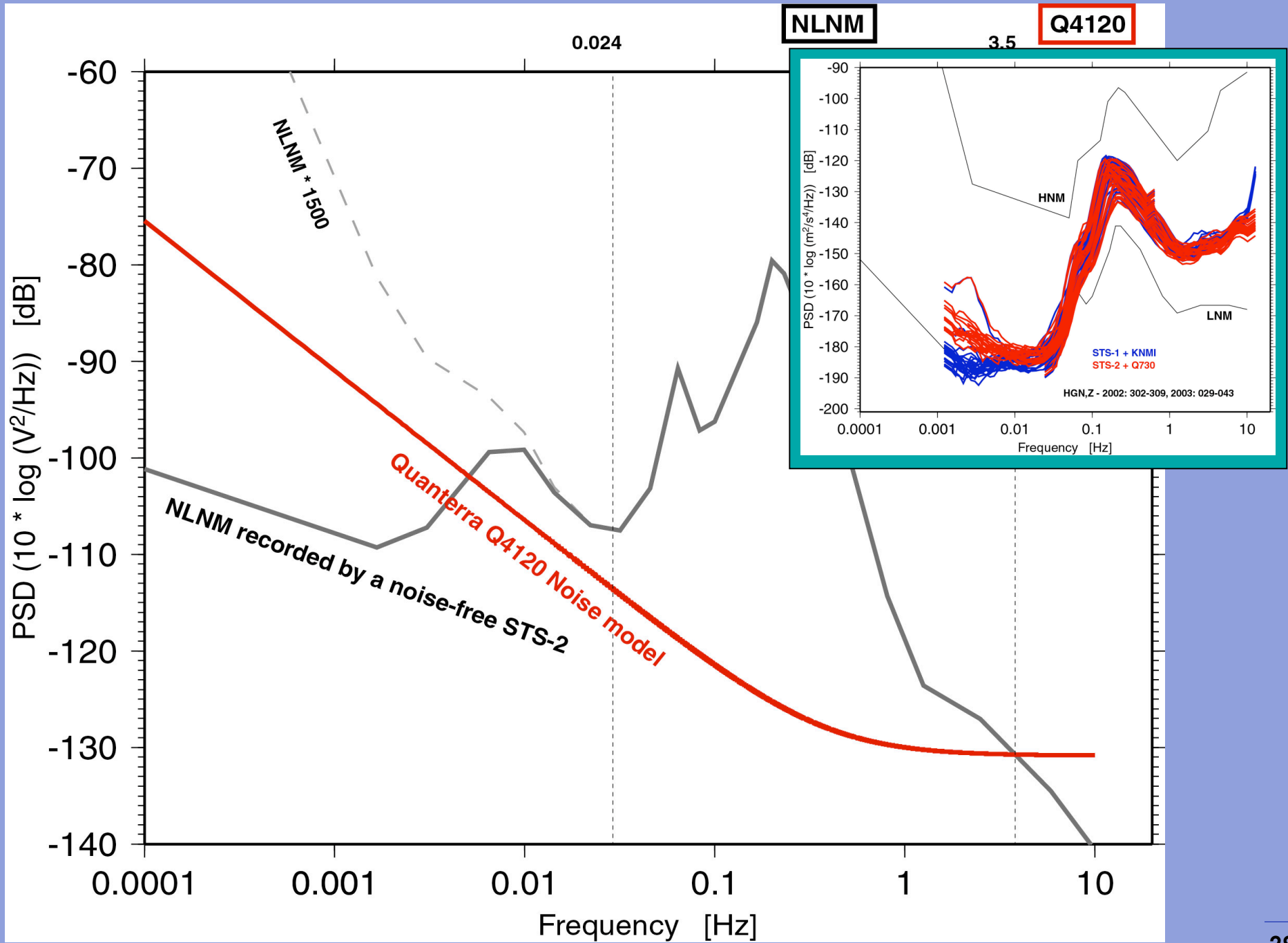


## PSD of self-noise Q4120 measured with common STS-2 vertical signal (@ 20 sps)

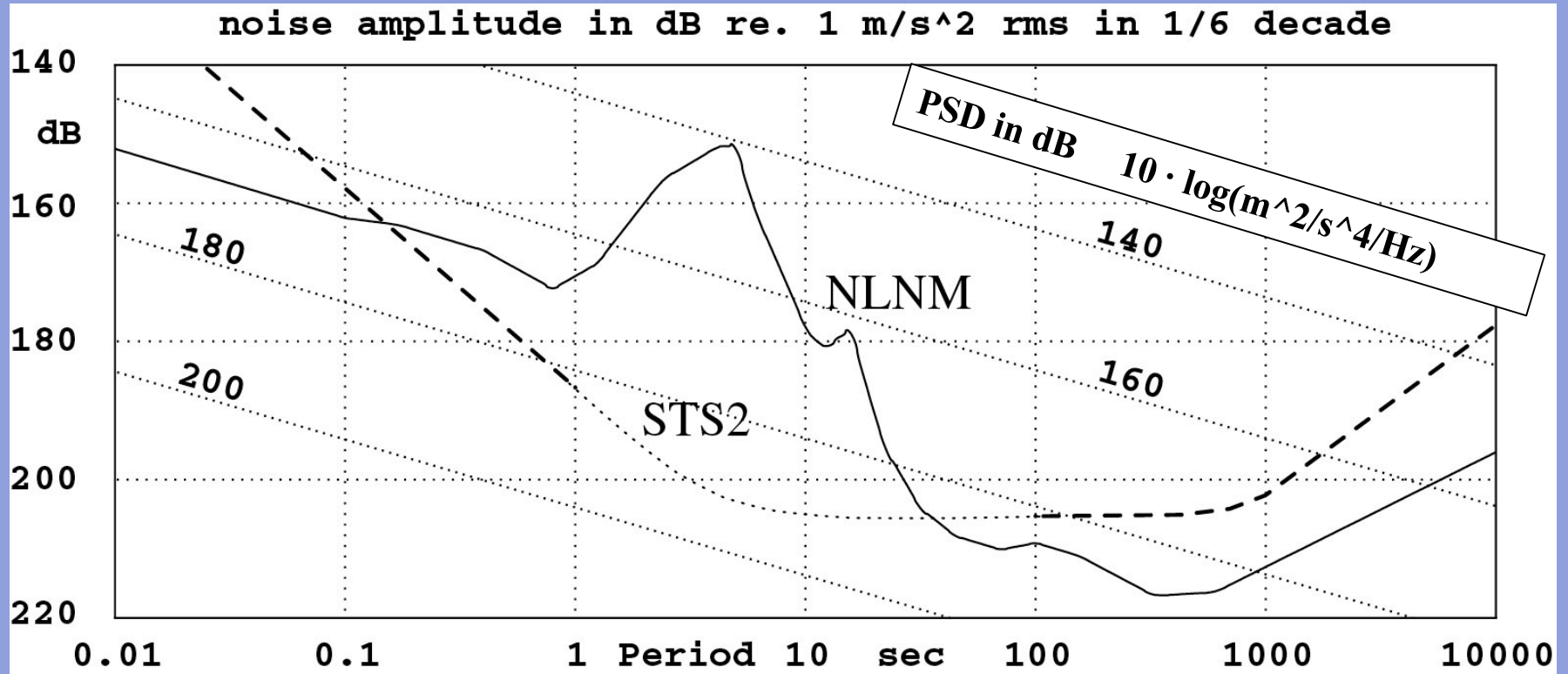


Dynamic range [dB]

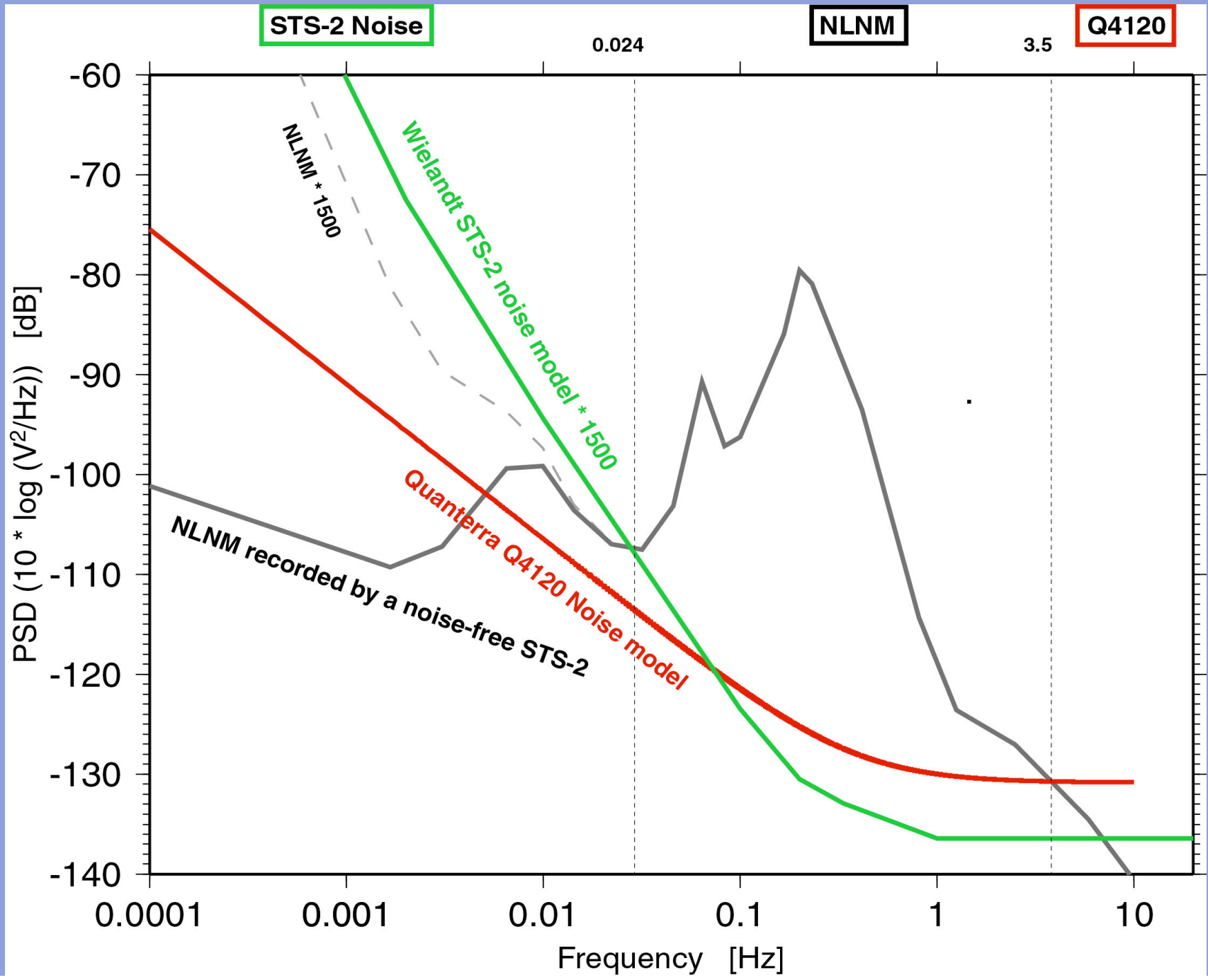


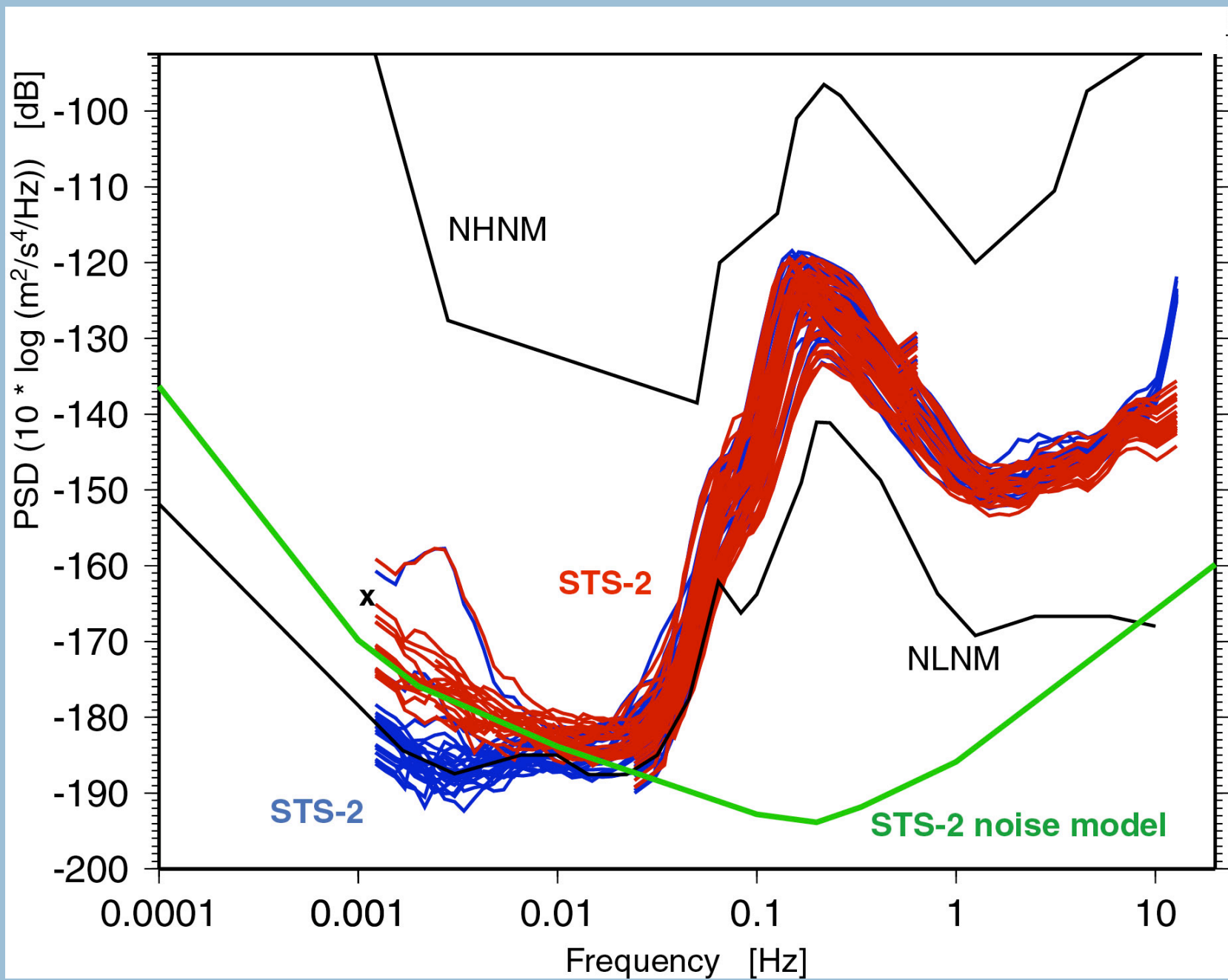


From: Wielandt



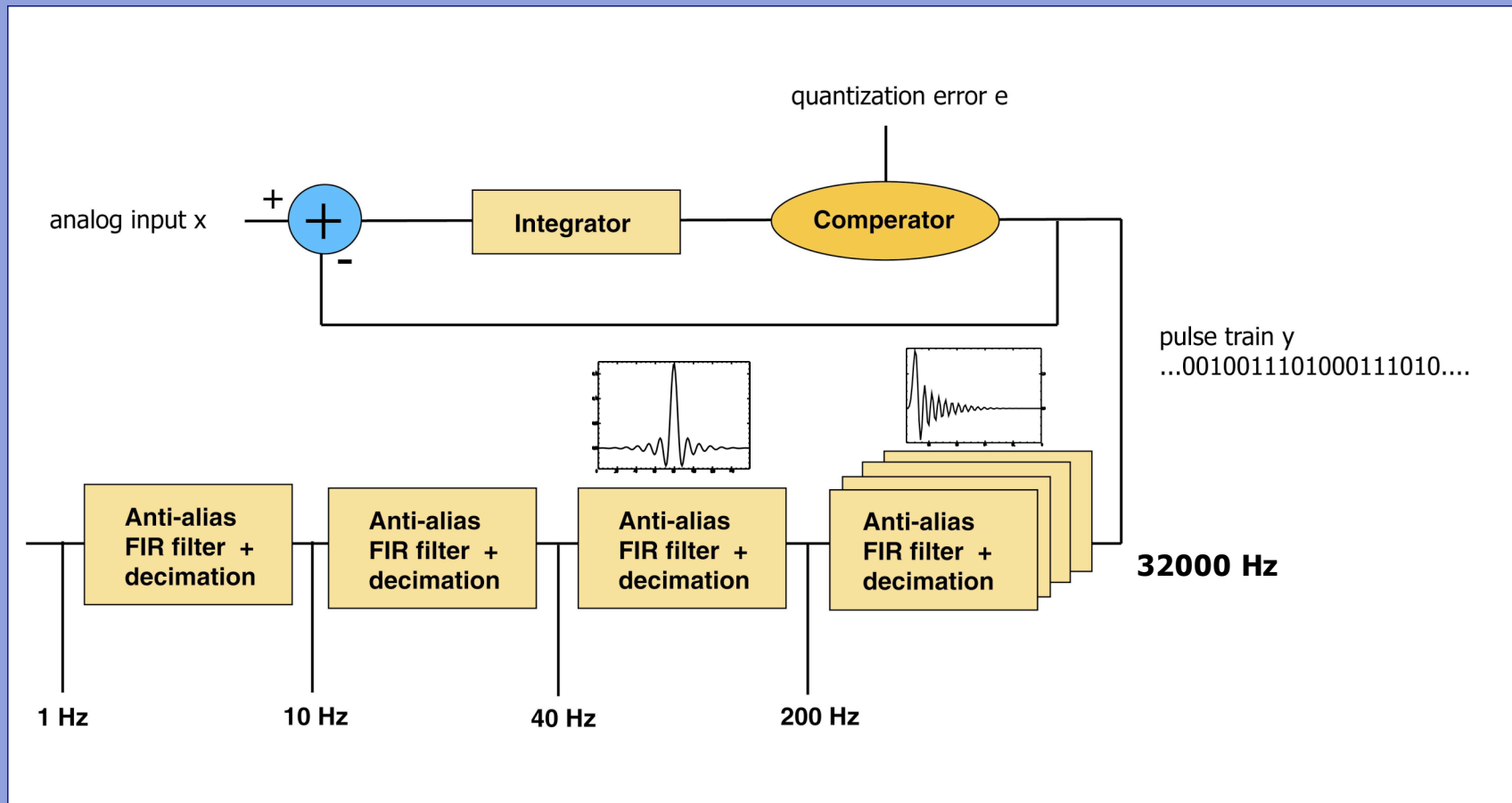
@1000 sec:      RMS = 212 dB      PSD =  $\text{RMS}^2/\text{BW}$   
                  PSD = 180 dB



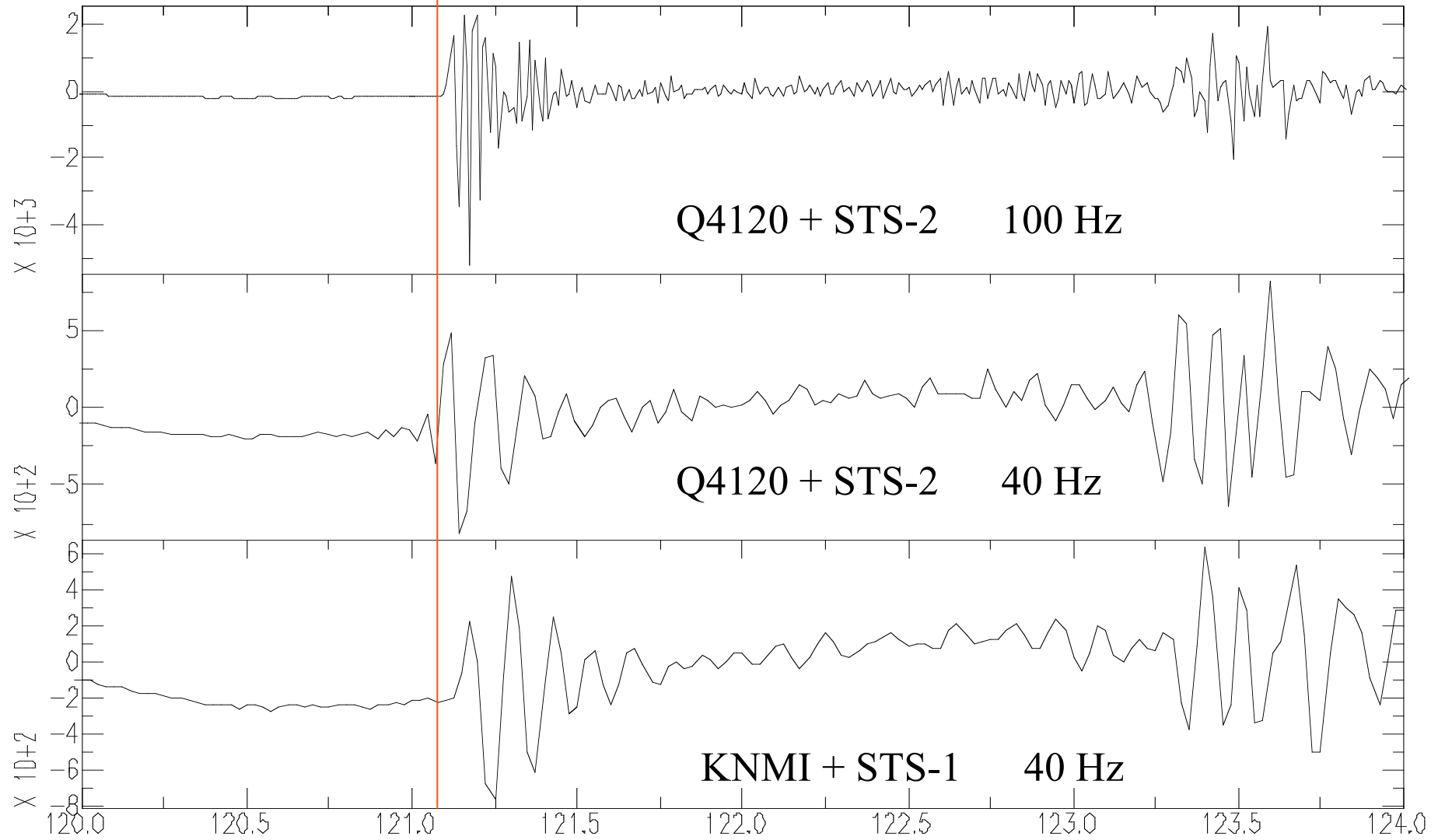


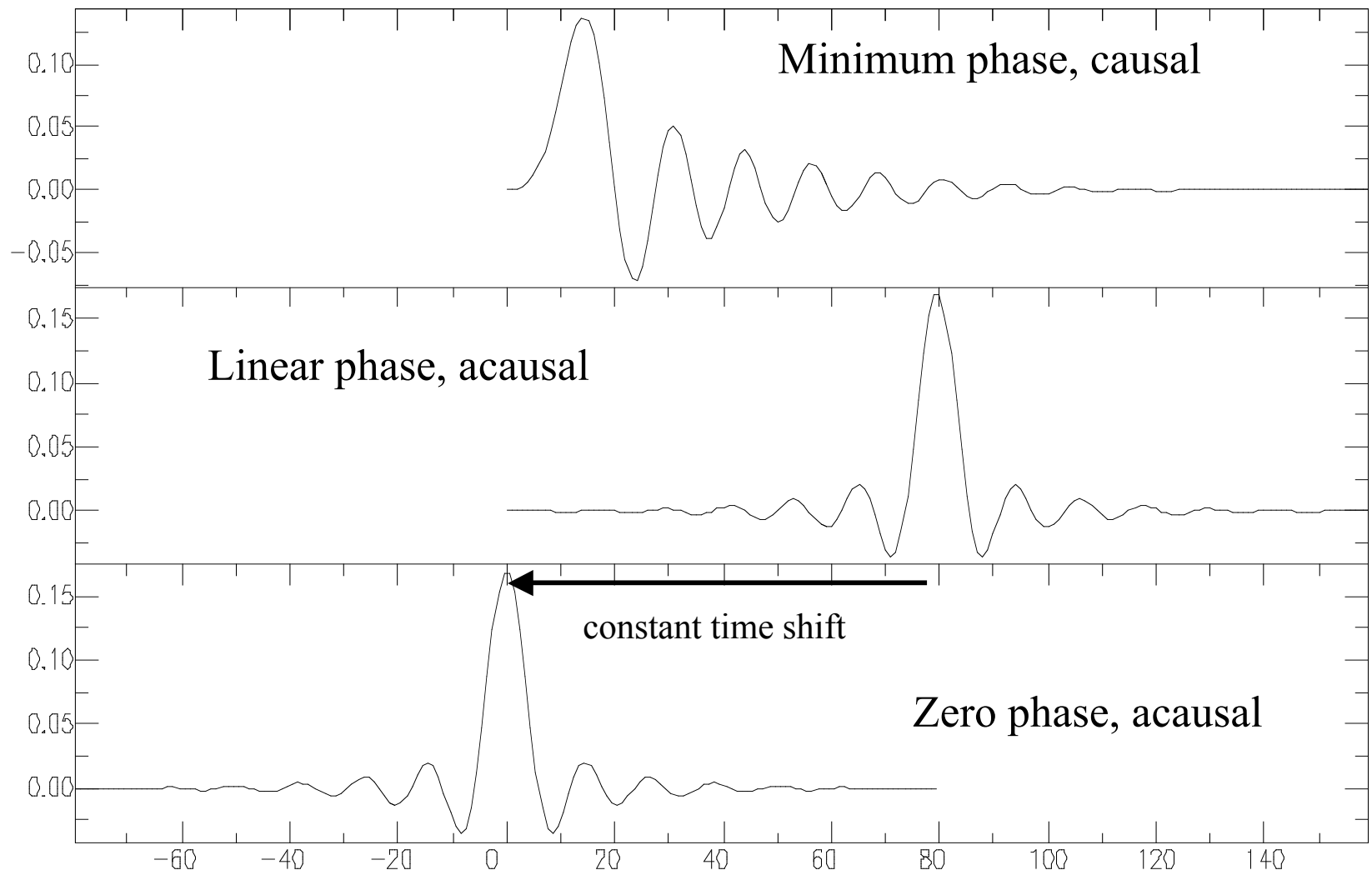


# Datalogger data processing



20 May 2003 22:13:03





## Removing acausal effects of the zero phase FIR filter

- FIR coefficients (polynomial coefficients) - Quanterra
- polynomial root finding (Jenkins/Traub, Muller, Newton) -  
Octave, Markus Lang, Mathematica
- replace maximum phase roots by minimum phase roots (  $c \rightarrow 1/c^*$  )
- get minimum phase equivalent FIR coefficients - Octave
- apply correction filter - Scherbaum

## IIR filter: relation between coefficients and poles/zeros

$$y_k = \sum_{m=0}^M a_m \cdot x_{k-m}$$

time domain

$a_m$  : filter coefficients

$$Y(z) = \sum_{m=0}^M a_m \cdot z^{-m} \cdot X(z)$$

z-domain

complex variable:  $z = e^{s \cdot T}$

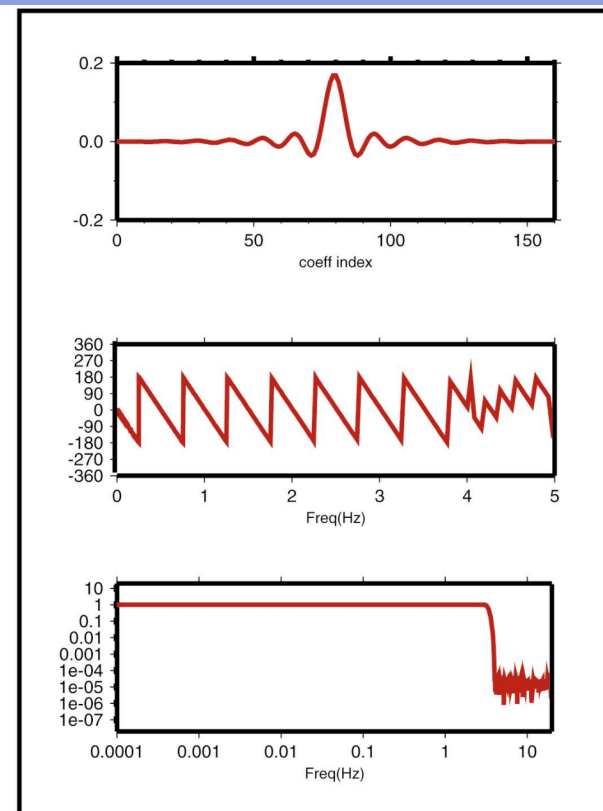
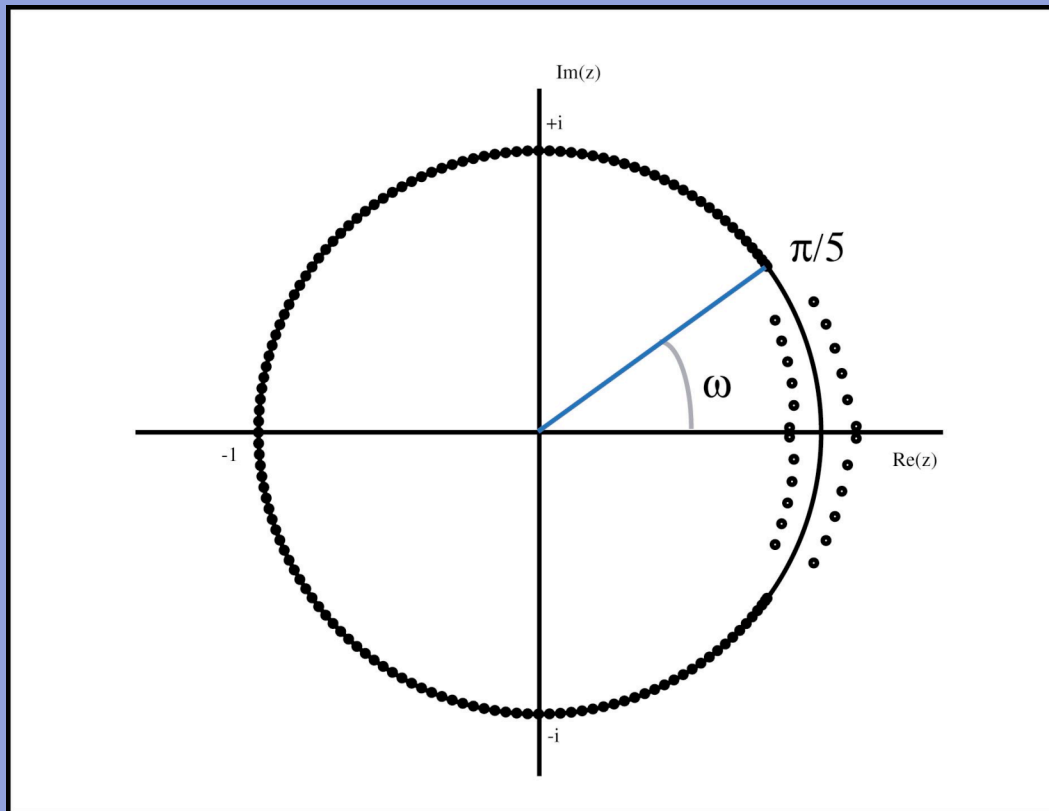
Numerator coefficients

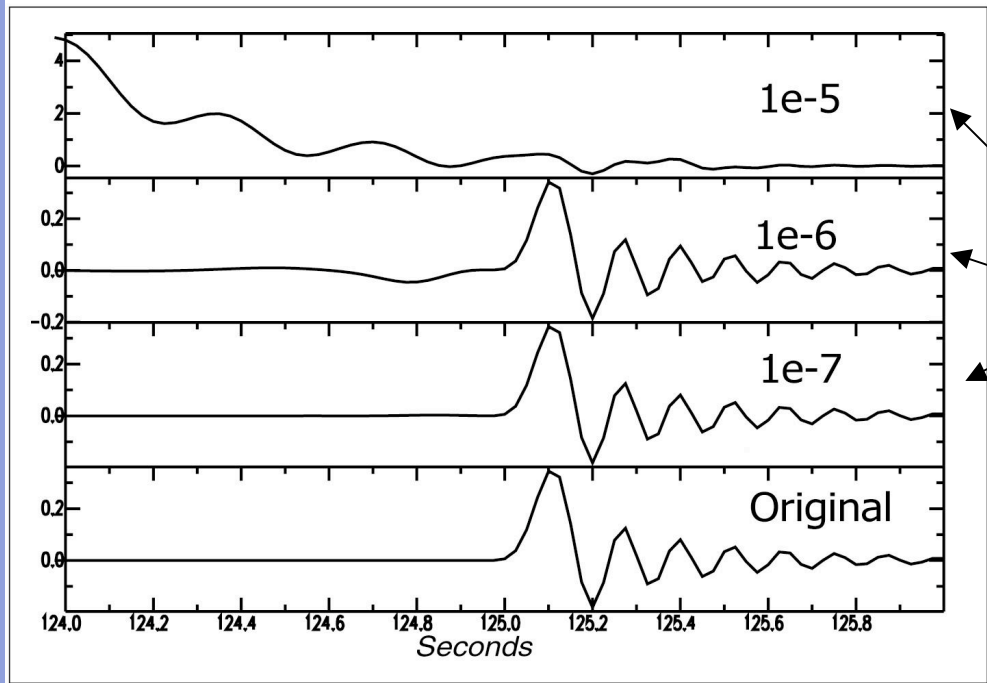
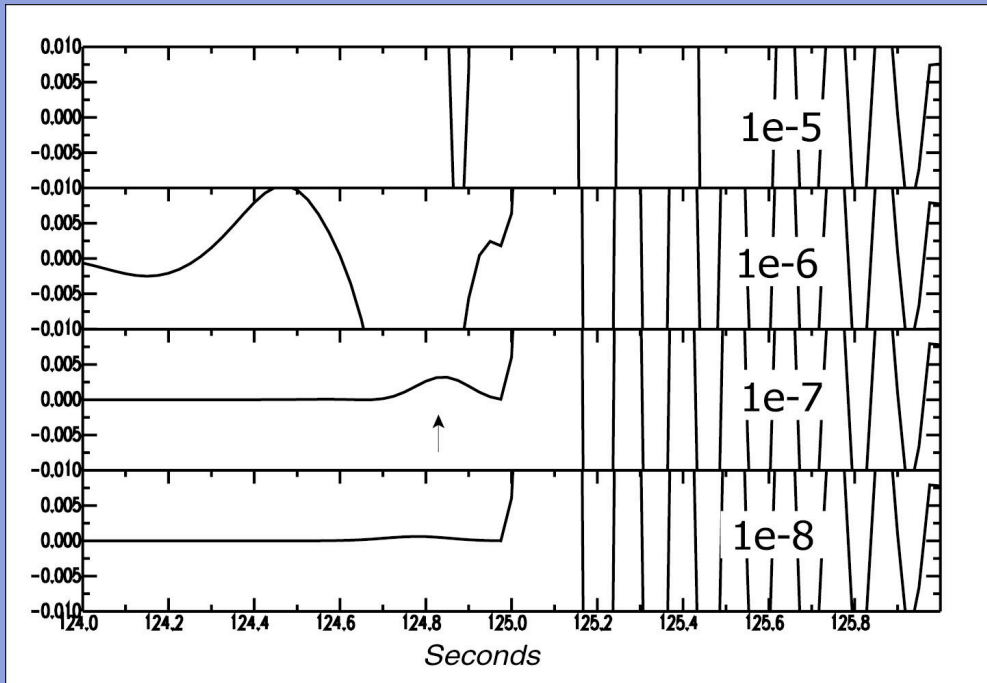
$$H(z) = \frac{Y(z)}{X(z)} = \frac{\sum_{m=0}^M a_m \cdot z^{-m}}{\sum_{k=0}^K b_k \cdot z^{-k}} = \frac{a_0 \cdot \prod_{m=1}^M (1 - c_m \cdot z^{-1})}{b_0 \cdot \prod_{k=1}^K (1 - d_k \cdot z^{-1})}$$

Denominator coefficients

$c_m$  : Roots of polynomial (zeros)  
 $d_m$  : Roots of polynomial (poles)

# Polynomial root finding





random noise addition