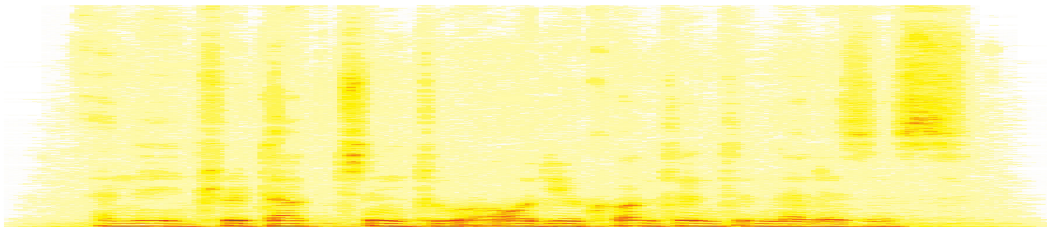


Introduction to Audio Content Analysis

Module 2.2: Fundamentals — Quantization

alexander lerch



introduction

overview

corresponding textbook section

Chapter 2 — Fundamentals: pp. 11–13

- **lecture content**

- discretization of signals in amplitude
- properties of the quantization error
- number representation in Matlab

- **learning objectives**

- understanding the principle and impact of quantization
- representing audio signals in Matlab



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digital signals

introduction

digital signals are represented with a limited number of values

⇒

- 1 **sampling**: time discretization
continuous time \mapsto discrete equidistant points in time
- 2 **quantization**: amplitude discretization
continuous amplitude \mapsto discrete, pre-defined, set of values

digital signals

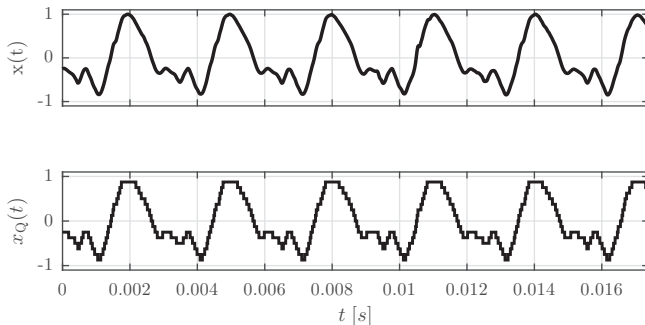
quantization

- continuous amplitude values are mapped to pre-defined, equidistant set of values
- signal stored in binary \Rightarrow # quantization steps equals **power of 2**
- **example: 4-bit quantization**
 - *word length: $w = \log_2(\mathcal{M}) = 4$ bit*
 - *number of quantization steps: $\mathcal{M} = 2^w = 16$*

digital signals

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digital signals

quantization wordlength

What are typical wordlengths?



digital signals

quantization wordlength

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- 8 bit: speech
- 12–14 bit: low quality audio/music
- 16 bit: (consumer) audio/music
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digital signals

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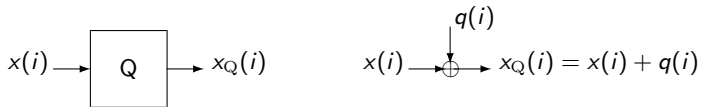
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w	16 bit	12 bit	8 bit	4 bit	2 bit
					



digital signals

quantization error



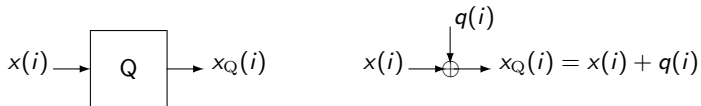
model for quantization:

quantization noise q is added to input signal x

$$\begin{aligned}x_Q(i) &= x(i) + q(i) \\ q(i) &= x(i) - x_Q(i)\end{aligned}$$

digital signals

quantization error



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digital signals

quantization error magnitude

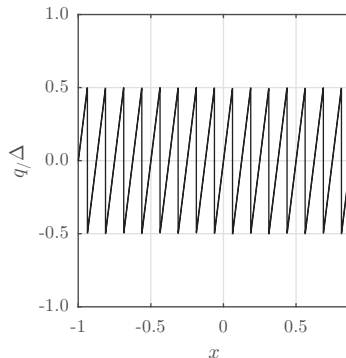
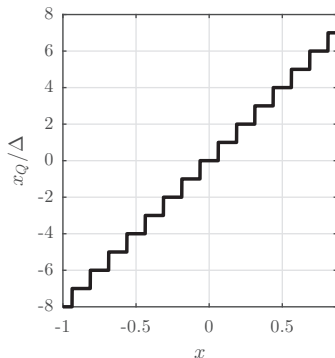
What is the maximum amplitude of the quantization error?



digital signals

quantization error magnitude

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digital signals

quantization error properties

Under the assumption that the signal has a variance much higher than the quantization step size (no derivation), we find that the quantization error

- is white noise and uncorrelated to signal,
- is uniformly distributed, and
- its power W_Q is directly related to the wordlength.

The quantizer quality is usually given by its *Signal-to-Noise Ratio (SNR)*

$$SNR = 10 \cdot \log_{10} \left(\frac{W_S}{W_Q} \right) [dB]$$

digital signals

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digital signals

quantization: SNR

signal-to-noise ratio (quantizer)

$$SNR = 6.02 \cdot w + c_S \quad [dB]$$

- every additional bit adds app. 6 dB SNR
 - constant c_S depends on *signal* (scaling and PDF)
-
- square wave (full scale): $c_S = 10.80$ dB
 - sinusoidal wave (full scale): $c_S = 1.76$ dB
 - rectangular PDF (full scale): $c_S = 0$ dB
 - Gaussian PDF (full scale = $4\sigma_g$): $c_S = -7.27$ dB



digital signals

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digital signals

amplitude in DSP

- when represented as integer, different wordlengths lead to different maximum amplitude ranges
- most common: normalize to the absolute maximum integer value and represent the signal in **floating point format**

⇒ signal amplitude:

$$-1 \leq x_Q < 1$$

⇒ level:

max. amplitude $\mapsto 0dBFS$

- floating point representation

$$x_Q = M_G \cdot 2^{E_G}$$

- internal float point representation usually treated as signal being **not quantized**

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summary

lecture content

- **quantization process**
 - modeled as addition of noise
- **quantization error properties**
 - maximum amplitude is half the step size
 - number of steps depends on wordlength
- **SNR**
 - SNR depends on input signal characteristic and wordlength
 - SNR increases linearly (6 dB/bit) with wordlength
- **signals in Matlab**
 - maximum amplitude independent of wordlength (-1...1)

