

## Introduction to Audio Content Analysis

module 3.2.5: fundamentals — blocking

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## introduction

overview



### corresponding textbook section

section 3.2.5

#### lecture content

- splitting the audio signal into blocks
- block length and hop size

### learning objectives

- describe the reasons for blocking
- summarize the principle using the correct terminology



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# block based processing introduction



- typical audio applications process blocks of audio data
- instead of having a function called per sample, it is called with block of samples

#### reasons:

- block based processing methods such as the Short-Time Fourier Transform
- audio hardware characteristics (real-time systems)
- efficiency (memory allocation, SIMD)
- typical block lengths:
  - 1... thousands of samples
  - often powers of 2 (1024, 2048, ...)

# block based processing introduction



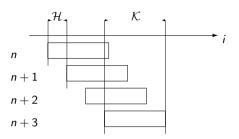
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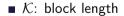
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### block based processing description

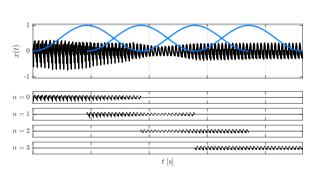




 $\blacksquare$   $\mathcal{H}$ : hop size

n: block index

■ *i*: sample index



definitions

block boundaries:

$$\begin{array}{lcl} \emph{i}_{s}(\emph{n}) & = & \emph{i}_{s}(\emph{n}-1) + \mathcal{H} \\ \emph{i}_{e}(\emph{n}) & = & \emph{i}_{s}(\emph{n}) + \mathcal{K} - 1 \end{array}$$

overlap ratio:

$$o_{
m r} = rac{\mathcal{K} - \mathcal{H}}{\mathcal{K}}$$

time stamp:

$$t_{\rm s}(n) = \frac{i_{\rm e}(n) - i_{\rm s}(n) + 1}{2 \cdot f_{\rm S}} + \frac{i_{\rm s}(n)}{f_{\rm S}} = \frac{\mathcal{K}}{2 \cdot f_{\rm S}} + \frac{i_{\rm s}(n)}{f_{\rm S}}$$

- K: block length
- $\blacksquare$   $\mathcal{H}$ : hop size
- *n*: block index
- *i*: sample index
- $f_S$ : sample rate

## summary lecture content



- audio signal is typically split into blocks
- each block processed individually
- terms:
  - block length:
    - minimum: 1
    - ► typical: 256 . . . 16384
  - hop size:
    - minimum: 1
    - maximum: block length
    - typical: half of block length
  - block *time stamp*:
    - typically refers to either start or middle of block

