

# Using MPEG-7 Audio Low Level Descriptor Scalability: A Guided Tour

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# Overview

- The need for scalable metadata
- MPEG-7 Audio scalability
  - Scalability over frequency
  - Scalability over time -> scalable series
- The ScalableSeries
  - Concept
  - Application examples
  - Limitations
- Conclusions



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# The Need for Metadata Scalability

Rich / precise  
descriptions



Compact/coarse  
descriptions

- Offer highest retrieval performance
- Occupy large amount of data
- High computational matching effort

- Reduced retrieval performance
- Smaller description database
- Manageable matching effort

- There is no fixed 'one size fits all applications' trade-off!
- MPEG-7 Audio format needs to serve all conceivable applications  
=> Concept of "Scalability"

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# Low Level Descriptor Overview

- |                                |   |
|--------------------------------|---|
| Basic                          | <ul style="list-style-type: none"><li>• Instantaneous waveform, power, silence</li></ul>  |
| Basic Spectral                 | <ul style="list-style-type: none"><li>• Power spectrum, spectral centroid, spectral spread, spectral flatness ...</li></ul>   |
| Signal Parameters              | <ul style="list-style-type: none"><li>• Fundamental frequency, harmonicity</li></ul>  |
| Spectral Basis Representations | <ul style="list-style-type: none"><li>• Used primarily for sound recognition, projections into low-dimensional space</li></ul>  |
| Timbral Temporal               | <ul style="list-style-type: none"><li>• Log attack time, temporal centroid of a monophonic sound</li></ul>  |
| Timbral Spectral               | <ul style="list-style-type: none"><li>• Features specific to the harmonic portions of signals (harmonic spectral centroid, spectral deviation, spectral spread, ...) etc.</li></ul> |



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# Metadata Scalability

## General Concept

- Many descriptive elements are *scalable* in one or more dimensions
  - Derive reduced-rate (compact) precision description from original rich version [may happen on-the-fly, e.g. streaming apps]
  - Result of “downscaling” is still compatible to original version (i.e. can be compared in a meaningful way)
  - *Hierarchical* scalability concept, i.e. richer representations are supersets of smaller ones

(see also “MPEG-4 Scalability”, i.e. parts of AV bitstream can be dropped => lower quality reconstruction of signal)



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# MPEG-7 Audio Scalability (2)

## Spectral Scalability

- Some LLDs (e.g. `AudioSpectrumEnvelope`) have *spectral resolution / coverage* as scalability dimension
    - Resolutions are powers of 2 (1/16 ... 8 oct.)
    - Frequency bands anchored around 1kHz
- => Commensurate frequency band definitions
- => All conceivable descriptions can be compared to each other
- Easy downscaling via simple additive combination of adjacent band energies



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# MPEG-7 Audio Scalability Concepts (3)

## Temporal Scalability

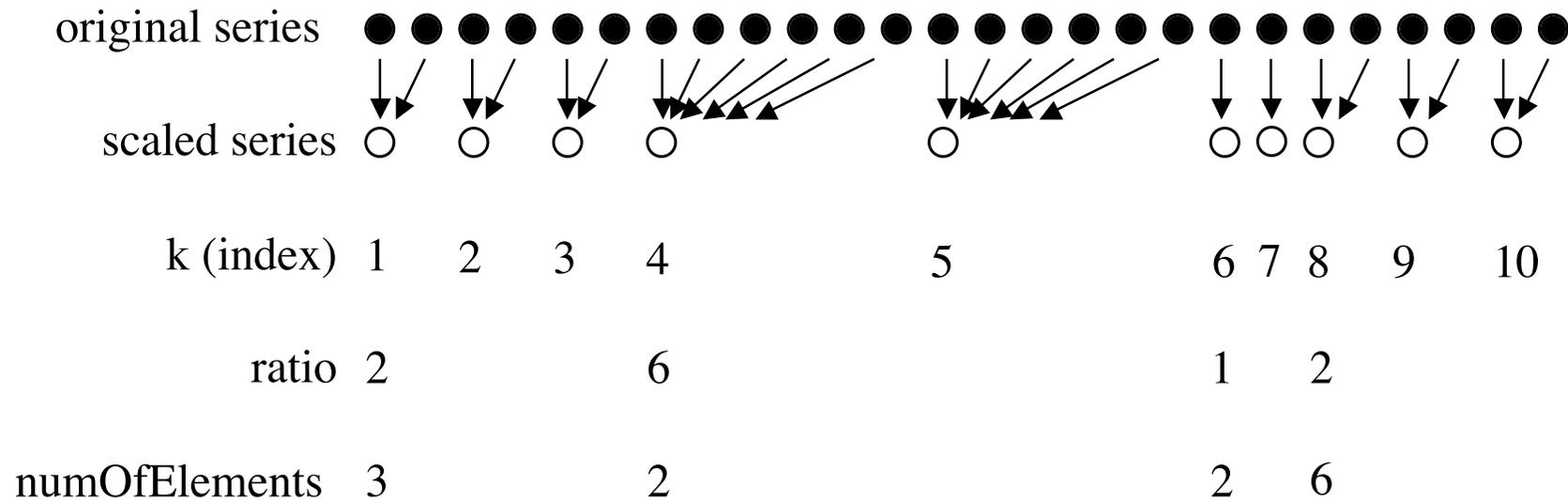
- Many LLDs describe time-varying signal properties □ sampled time series
- How fine sampling is needed ? (depends on individual application □ cannot be fixed at time of definition)
- Sampling rate determines data rate

## Approach

- Define one generic construct for sampled time series which enables a scalable representation [Alain de Cheveigné]:  
□ `Concept of ScalableSeries`



# Scalable Series



General Idea: Variable decimation of sampled data (factor `ratio`)



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## Scalable Series (2)

### Concept of ScalableSeries

- Allow downsampling of sampled descriptor data in time
- Decimation may happen in several steps, yet results are independent of intermediate steps
- Provides summarization of raw data
  - Mean, Variance
  - Min, Max, First, Last, Random
- Provides additional hints
  - Weigh



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# Scalable Series (3)

## Data types

- Available for
  - scalar data types (e.g. AudioSpectrumCentroid)  
`SeriesofScalar`
  - vector data types (e.g. subband-based data such as AudioSpectrumEnvelope)  
`SeriesofVector`
- Subtypes: Decimation by powers of 2
  - `SeriesofScalarBinary`,  
`SeriesofVectorBinary`
  - Strictly hierarchical
  - Allows to retain some statistics of what has been lost by progressive downsampling (“scalewise variance”)



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# Application Example: Scalable Audio Fingerprint

LLD/Feature

- `AudioSignature/AudioSpectrumFlatness,`  
**data type** `SeriesofVectorBinary`

Offers different "operating points":

Time

- Duration / Time Coverage

Granularity

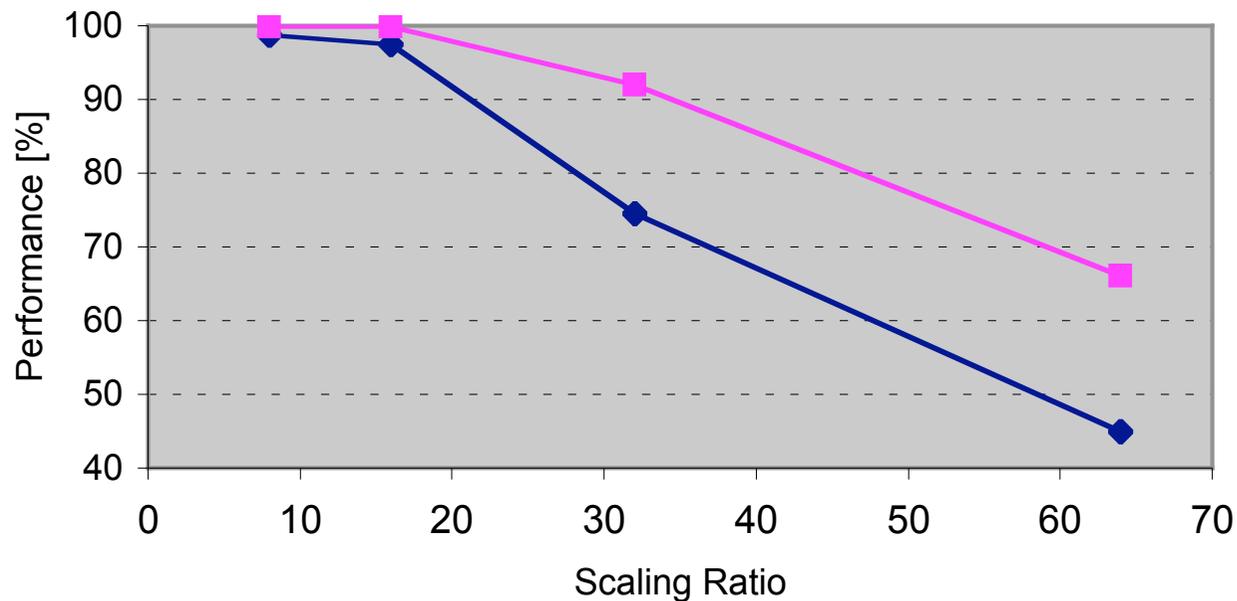
- Temporal Granularity / Resolution:  
60ms ... 1s ... 4s

Richness

- Spectral Information Amount  
(number of frequency bands):  
4 ... 16 ... 24
- Datarate of 2Byte/s ... 32 ... 800Byte/s,  
Fingerprints can be "transcoded"



## Application Example (2)



- Performance of an audio fingerprinting application ( $ASF$  LLD) as a function of temporal resolution of description ( $ratio$ )
- Varying scaling ratio between 8 and 64

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# Application Example: Search in Large Databases

## Fast Preselection Search

- Varying the spectral coverage
- Database: 15,000;
- Test: 1,000 Items; Length 20s ; Top2%

No. of bands	4	8	16(def.)
Speed up factor	4	2	1
MP3@96kbps	100.0%	100.0%	100.0%
Microphone	100.0%	99.9%	99.9%
Resampling	99.4%	100.0%	100.0%
Equalizer	97.3%	99.9%	100.0%
DynComp	100.0%	100.0%	100.0%



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# Application Example: Search in Large Databases

## Fast Preselection Search

- Decreasing the temporal resolution
- Database: 15,000;
- Test: 1,000 Items; Length 20s ; Top2%

Grouping	32(def. 0.96s)	128 (3.84s)	256 (7.68s)
Speed up factor	1	16	64
MP3@96kbps	100.0%	100.0%	99.1%
Microphone	99.9%	99.6%	98.7%
Resampling	100.0%	100.0%	98.4%
Equalizer	100.0%	100.0%	96.9%
DynComp	100.0%	99.9%	99.5%



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# Scalable Series: Limitations

- Most important data field: `mean`
- Can be modeled as a Moving Average (MA) filter with subsequent decimation stage
- Decimation ratio  $N$  is equal to the number of grouped data points

## Advantages

- Simplicity; computation in several (intermediate) steps leads to same result

## Limitations

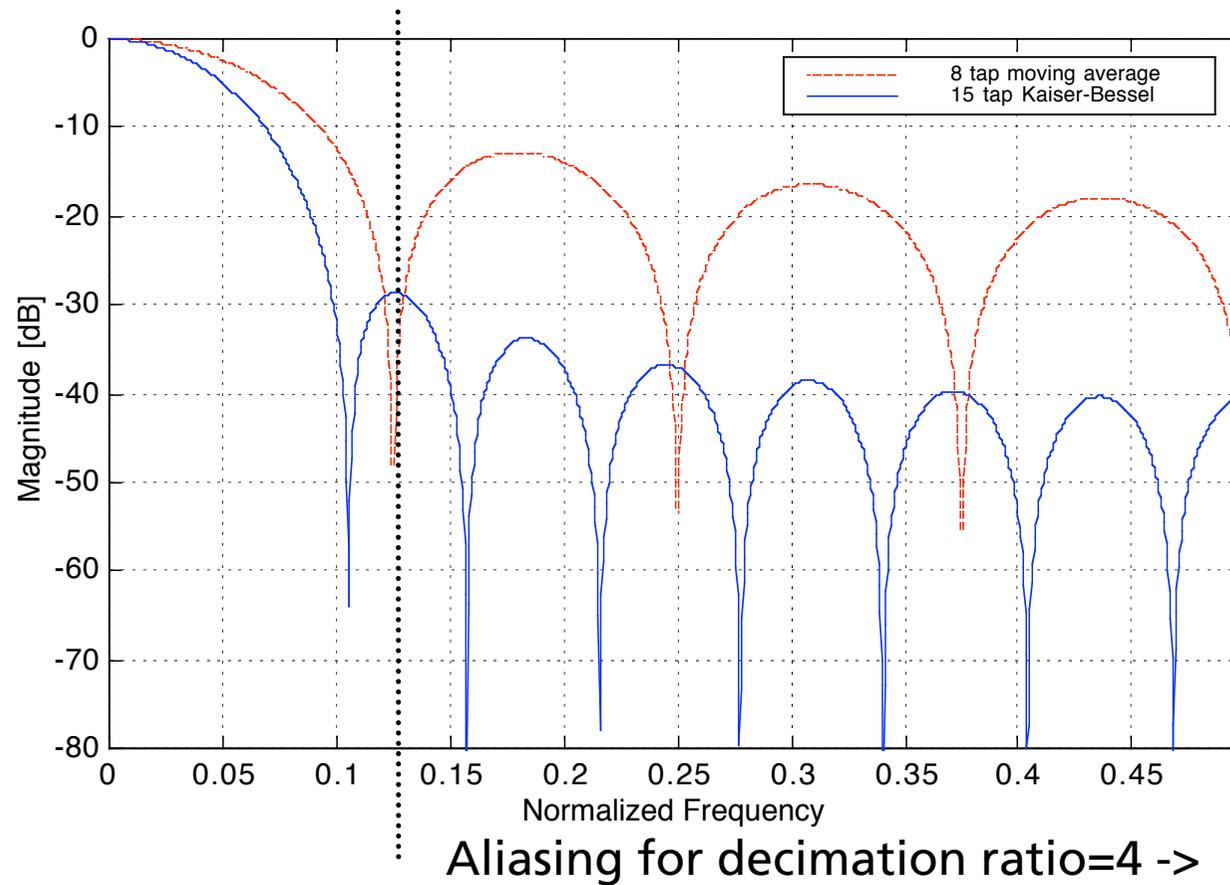
- Poor filter attenuation above cut-off frequency => aliasing after decimation!

## Question

- Performance penalty for applications due to aliasing components?



# Scalable Series: Filter Frequency Response



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# A Test Case: Audio Fingerprinting For Cell Phone

## Test Database

- 440 audio excerpts recorded over GSM cell-phones (10-20s duration, highly distorted)

## Reference items

- 15,000 fingerprints extracted from high quality audio material (CD)

## Feature Setup

- ASE, 250Hz - 4000Hz; 2, 1, 1/2, 1/4 octaves
  - total of 2,4,8,16 bands
- Decimation by factor of 4 for recognition
- Comparing application performance:
  - Standard ScalableSeries Decimation (MA)
  - Decimation after 15 taps Kaiser-Bessel LP



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# Results

Resolution [octaves]	2	1	1/2	1/4
Moving Average	95.07%	97.98%	98.20%	97.98%
Kaiser-Bessel	96.63%	98.65%	98.43%	98.43%

- Some performance penalty, may not be significant for many applications

Remedy: Use lower `ScalableSeries` decimation factor ...



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# Conclusions

- Scalability is essential for a universal metadata format - allows flexible trade-off between computational efficiency/ compactness and precision
- MPEG-7 Audio includes several types of scalability
  - Frequency scalability
  - Time resolution (via ScalableSeries)
- ScalableSeries is simple & efficient way to decimate data over time (small penalty)
- Both scalability types have clearly proven their usefulness in applications ...

