

VK Multimedia Information Systems

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Dienstags, 16.00 Uhr c.t., E.1.42

Audio & Music Retrieval



- What is Digital Audio?
- Features & Descriptors
- Speech
 - Speech Recognition
 - Speaker Detection
- Event Detection
- Music Retrieval
 - Motivation & Problems
 - Algorithms & Methods



What is Digital Audio



- Analogue signal goes digital
- Digitization: PCM
- Formats:
 - Compression
 - Containers

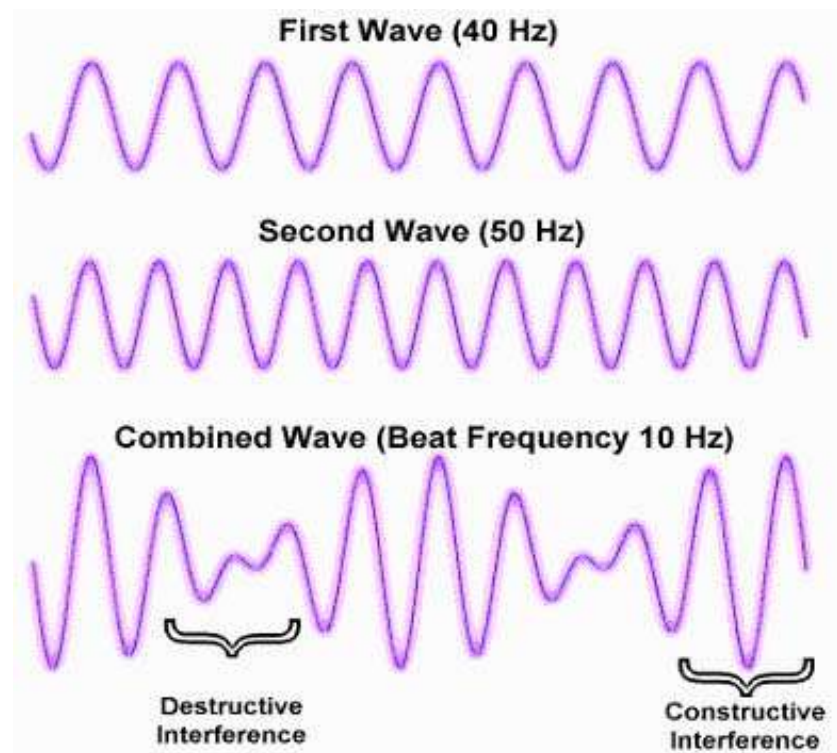
What is sound?



What is sound?



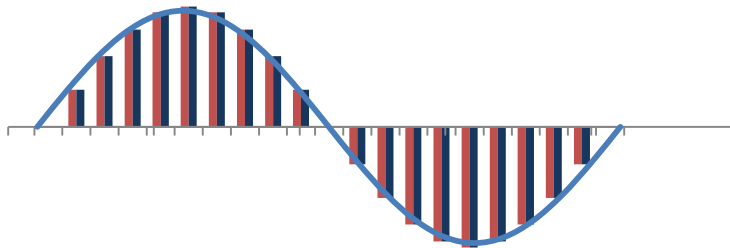
- Multiple sounds at the same time?



What is digital sound?



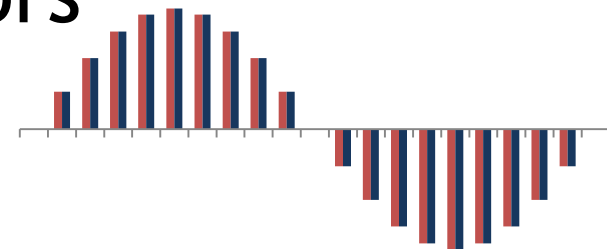
- A digitization of the wave.
 - Either a recipe for reconstruction
 - Or a discrete approximation



Sampled sound



- Wave gets sampled x times a second
 - E.g. 48.000 times -> 48 kHz sampling rate
- Obtained values are stored
 - E.g. 256, 240, 13, -7, -12, -44,
 - Quantization to e.g. 2^8 levels -> 8 Bit
- Possibly from different sensors
 - Stereo -> 2 channels



Sampled sound

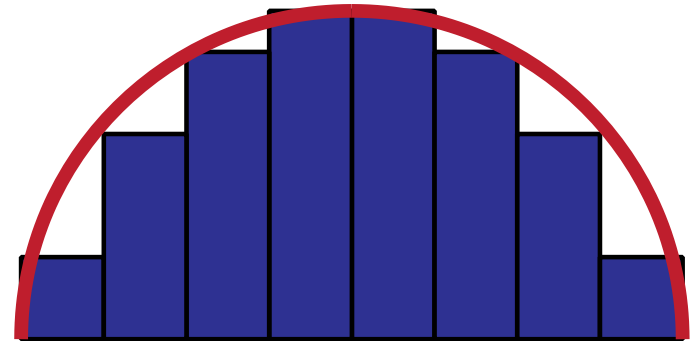
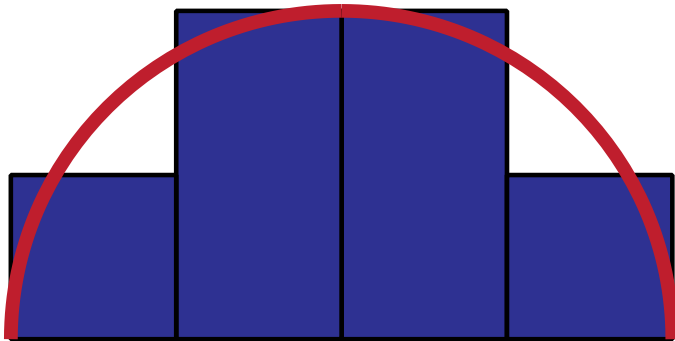


- Example: 8 kHz, 16 bit Stereo
 - Sound wave is sampled 8.000 times a second
 - Samples are stored in 16 bit numbers
- That's *Pulse Code Modulation* (PCM)
 - Often used in WAV files ...
 - Also as input from microphone or line in

Sampling Rates



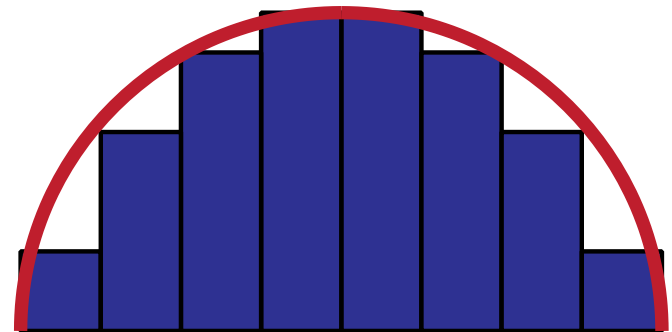
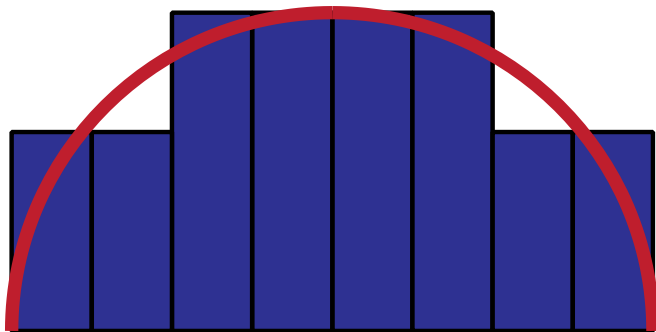
- With sampling rate x we can approximate frequencies up to $x/2$
- Assume frequency 1
 - sampling rate of 1 \rightarrow “0”
 - sampling rate of 2 \rightarrow “1,-1”



Quantization



- Reduces the possible values of the samples to a certain value
 - 8 Bit -> 256 levels, etc.



What do we want to capture?



- Humans can hear
 - From around 16 – 21 Hz
 - To around 16 kHz – 19kHz
 - 16 bit is enough (CD), 32 bit even better

Sound Formats



- **Waveform Audio Format**
 - Container for several compression formats
 - Includes PCM, MP3, GSM, μ -Law
- **Musical Instrument Digital Interface**
 - Control codes for instruments
 - Instruments can be “emulated”
- **Compressed Audio Formats**
 - MP3, OGG, AAC, ...

Audio & Music Retrieval



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- **Event Detection**
- **Music Retrieval**
 - Motivation & Problems
 - Algorithms & Methods

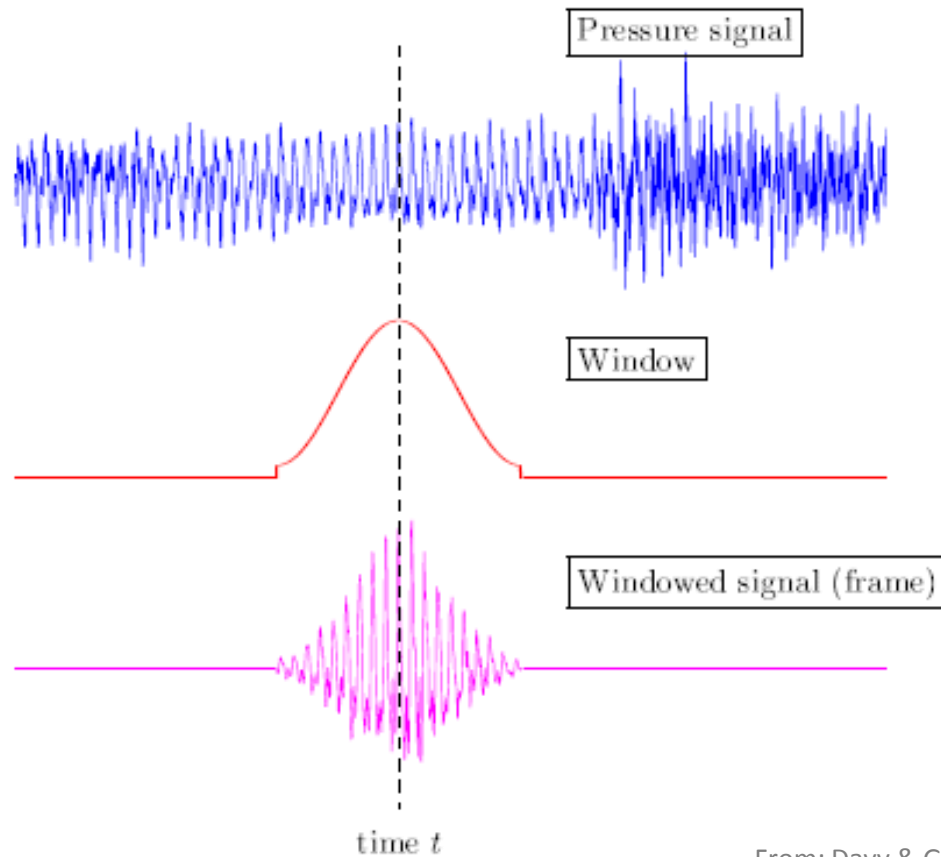


Audio Low-Level Features



- Lots of different applications
 - Speech Recognition
 - One of the first applications
 - Music Information Retrieval
 - Environmental Sound Recognition
- Different specific Features / Descriptors
- Standardization Efforts
 - MPEG-7, *low-level audio descriptors*

Audio Frames



From: Davy & Goodsill, "Audio Information Retrieval: A Bibliographical Study", TR, 2002

Audio Low-Level Features



- Features describe audio frames
- Frame definition critical to outcome
 - Shape (rectangular, Hamming, etc.)
 - Size (e.g. 150 ms)
- Features capture aspects
 - Energy (loudness)
 - Frequencies
 - Change over time (attack time, etc.)

Audio Low-Level Features



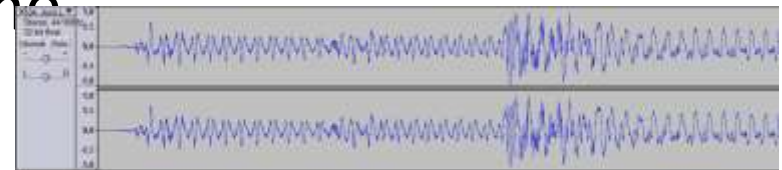
- MPEG-7 collects assortment of usable features
 - Basic
 - Basic Spectral
 - Spectral Basis
 - Signal Parameters
 - Timbral Temporal
 - Timbral Spectral

Basic Features



Short description of audio waveform

- **AudioWaveform**
 - Mainly for display
 - Minimum and maximum of the envelope
- **AudioPower**
 - Average square of the waveform samples
 - Power of the signal over time

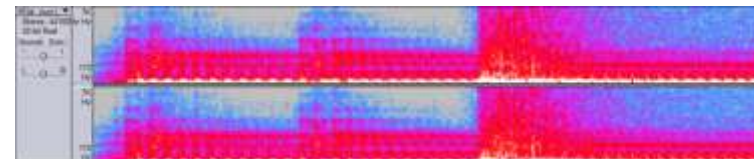


Basic Spectral



Basic properties of the spectrum of a signal

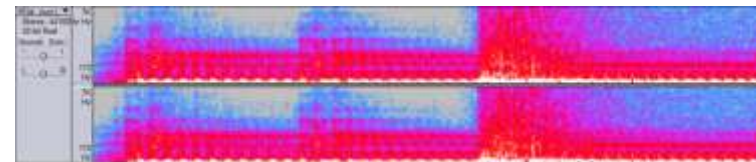
- **AudioSpectrumEnvelope**
 - short-term power spectrum of a signal
 - logarithmic frequency scale
 - imitating the human ear
- **AudioSpectrumCentroid**
 - Center of gravity in above descriptor
 - Indicates whether high/low freq. dominate



Basic Spectral



- AudioSpectrumSpread
 - deviation of the power spectrum from centroid
 - separation of tonal from noise-like sounds
- AudioSpectrumFlatness
 - deviation of the spectrum from a flat shape
 - designed to perform *fingerprinting*

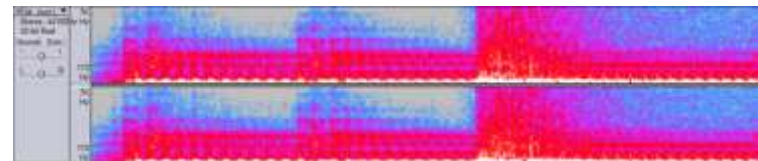


Spectral Basis



... general-purpose sound recognition

- **AudioSpectrumBasis**
 - Transforms spectrum to a lower-dimensional representation
 - Based on power spectrum
- **AudioSpectrumProjection**
 - Also transformation / reduction of data
 - Based on orig. signal & above descriptor



Signal Parameters



- **AudioFundamentalFrequency**
 - Fundamental frequency of a sound
 - Applicable to sound segmentation of speech and music
- **AudioHarmonicity**
 - Measure for the degree of harmonicity
 - Allows distinction between
 - sounds with a harmonic spectrum (e.g., musical tones or voiced speech [e.g., vowels]),
 - sounds with an inharmonic spectrum (e.g., metallic or bell-like sounds) and
 - sounds with a non-harmonic spectrum (e.g., noise, unvoiced speech, or dense mixtures of instruments)

Timbral Temporal



- Usually employed in music retrieval, independent of pitch and loudness
- LogAttackTime
 - logarithm of attack time of a sound
 - attack time is the time from the beginning of a sound signal to a point in time where its amplitude reaches a maximum
- TemporalCentroid
 - Time point of highest signal energy

Timbral Spectral



Descriptors rely on harmonic peak estimation

- Harmonic peaks
 - Correspond to frequencies that are a multiple of the fundamental frequency
 - Are used to describe the timbre of a signal

Timbral Spectral



- HarmonicSpectralCentroid
 - Amplitude-weighted average of harmonic peaks in spectrum
- HarmonicSpectralSpread
 - Amplitude-weighted deviation of harmonic peaks from above feature

Timbral Spectral



- HarmonicSpectralDeviation
 - Deviation of harmonic peaks from spectral envelope
- HarmonicSpectralVariation
 - Correlation of harmonic peaks in adjacent frames
- SpectralCentroid
 - Power-weighted average of frequencies in the power spectrum

Audio & Music Retrieval



- Features & Descriptors
- **Speech**
 - Speech Recognition
 - Speaker Detection
- Event Detection
- Music Retrieval
 - Motivation & Problems
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Speech Recognition: The Problem



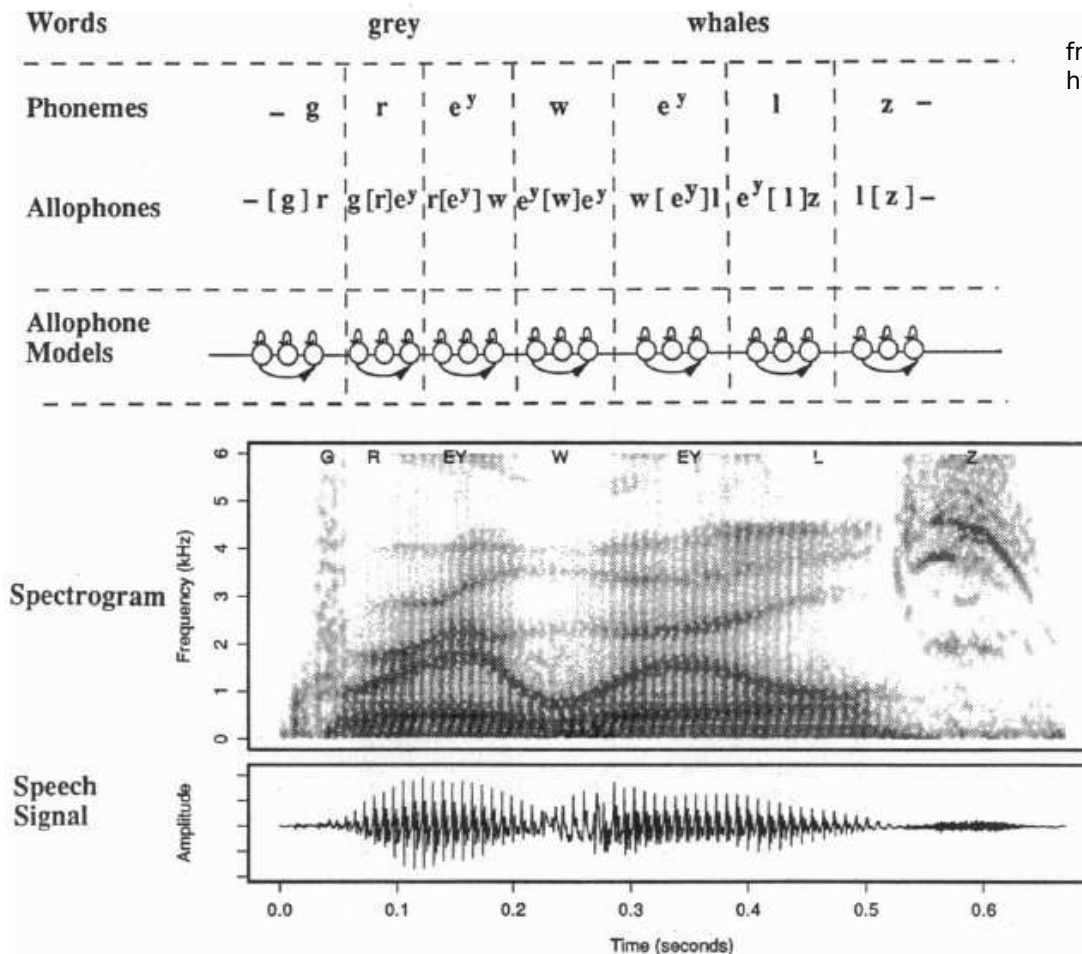
- Conversion of **acoustic signal** into words
- Different possible approaches
 - Isolated-word speech recognition
 - Continuous speech recognition
- Dependence on speaker
 - Training samples or independent

Speech Recognition: The Problem



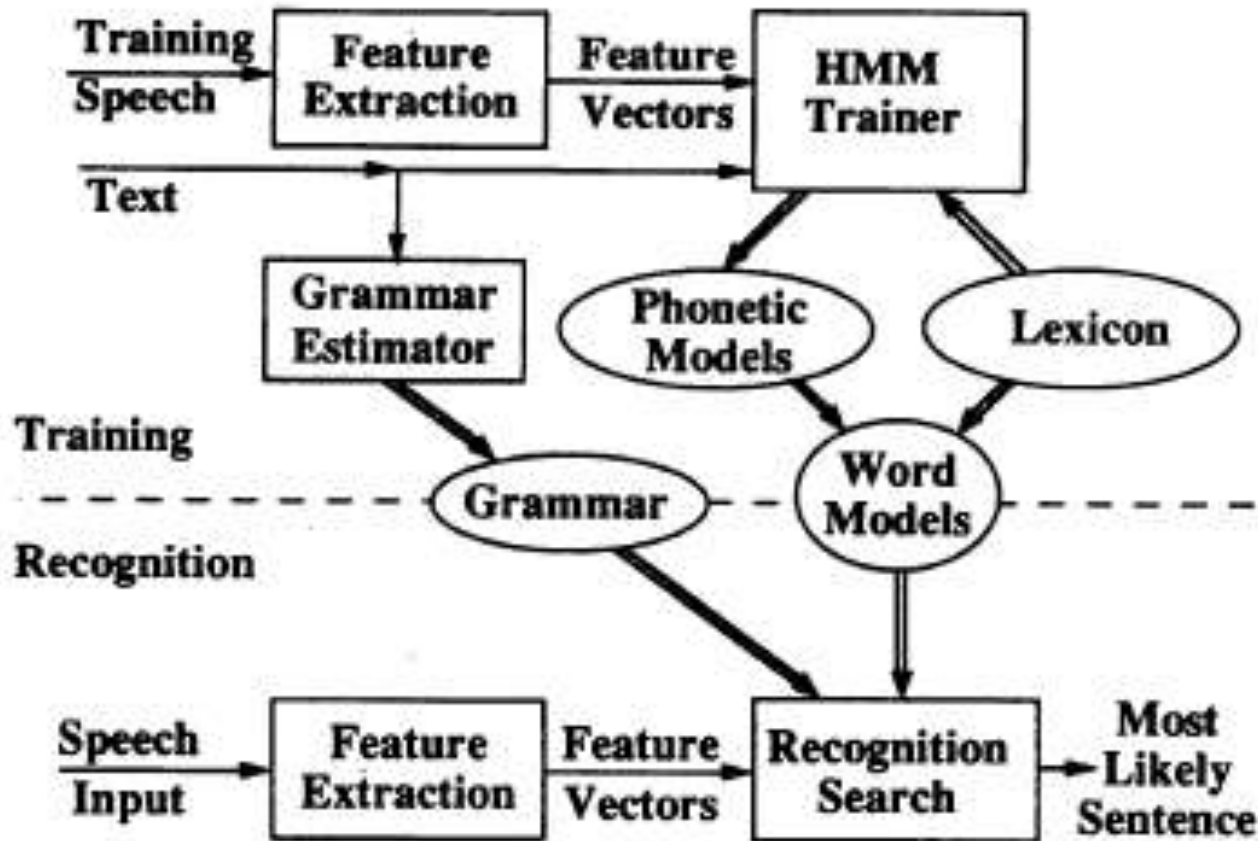
- Spontaneous vs. speech read from script
 - Spontaneous has disfluencies, more challenging task
- Language model is used for word sequences
 - Restriction to combination of words
- Measure for difficulty of the task: *Perplexity*
 - geometric mean of the number of words that can follow a word after the language model has been applied

Words & Phonemes



from: State of the Art in Continuous Speech Recognition:
<http://www.pnas.org/cgi/reprint/92/22/9956>

Speech Recognition: Process



Speech Recognition: Process



- Feature Extraction
 - Recognition from signal (near real time)
 - Amount of data (matching & indexing)
- Training
 - Modeling characteristics of speakers
 - Pronunciation -> Phonetic HMM
 - Grammar -> Markov Model (bi- or tri-word)
- Recognition
 - Search among possible word sequences
 - Highest possibility as match

Speaker Recognition



Detect speaker (change) in continuous speech

- Proper features to describe speaker
 - based on group of possible speakers
- Proper classification algorithm
- Robust against natural influence
 - Noise, cold, emotions, etc.

Speaker Recognition: Applications



- Video Analysis
 - Segmentation of interviews, etc.
- Media Analysis
 - How long did the J. Doe speak in TV this month?
- Security
 - Access restrictions: “Computer, shut down the warp drive!”

Audio & Music Retrieval



- Features & Descriptors
- Speech
 - Speech Recognition
 - Speaker Detection
- **Event-Detection**
- **Music Retrieval**
 - Motivation & Problems
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Event Detection



- Several domains have simple characteristics:
 - **Sports** events follow rules, participants behave similar
 - **News** broadcasts have a simple scheme, news anchormen introduce and summarize stories
 - **Surveillance** applied in 'dull' scenarios to detect 'extreme situations' like fire, panic, etc.
 - **Ad blocks** in TV have rough and fast scene cuts and raised volume

Event Detection



- Events are ‘peaks’ in one or several dimensions
 - Appropriate dimensions have to be found
 - Possibility of event has to be calculated
- Several domain rules might apply
 - Scoring in soccer after final whistle not possible
 - Foul and applause unlikely to occur immediately after another
 - Certain temporal distance between ad blocks

Event Detection: Example Soccer



- Event of Scoring:
 - Applause following a goal
 - Raised volume in commentators voice
 - Whistle of the referee
- Event of Foul:
 - Cheers of 'boo'
 - Whistle of the referee
- Event of Start / End of game
 - First and last whistle of the referee
 - Certain minimum amount of time in between

Audio & Music Retrieval



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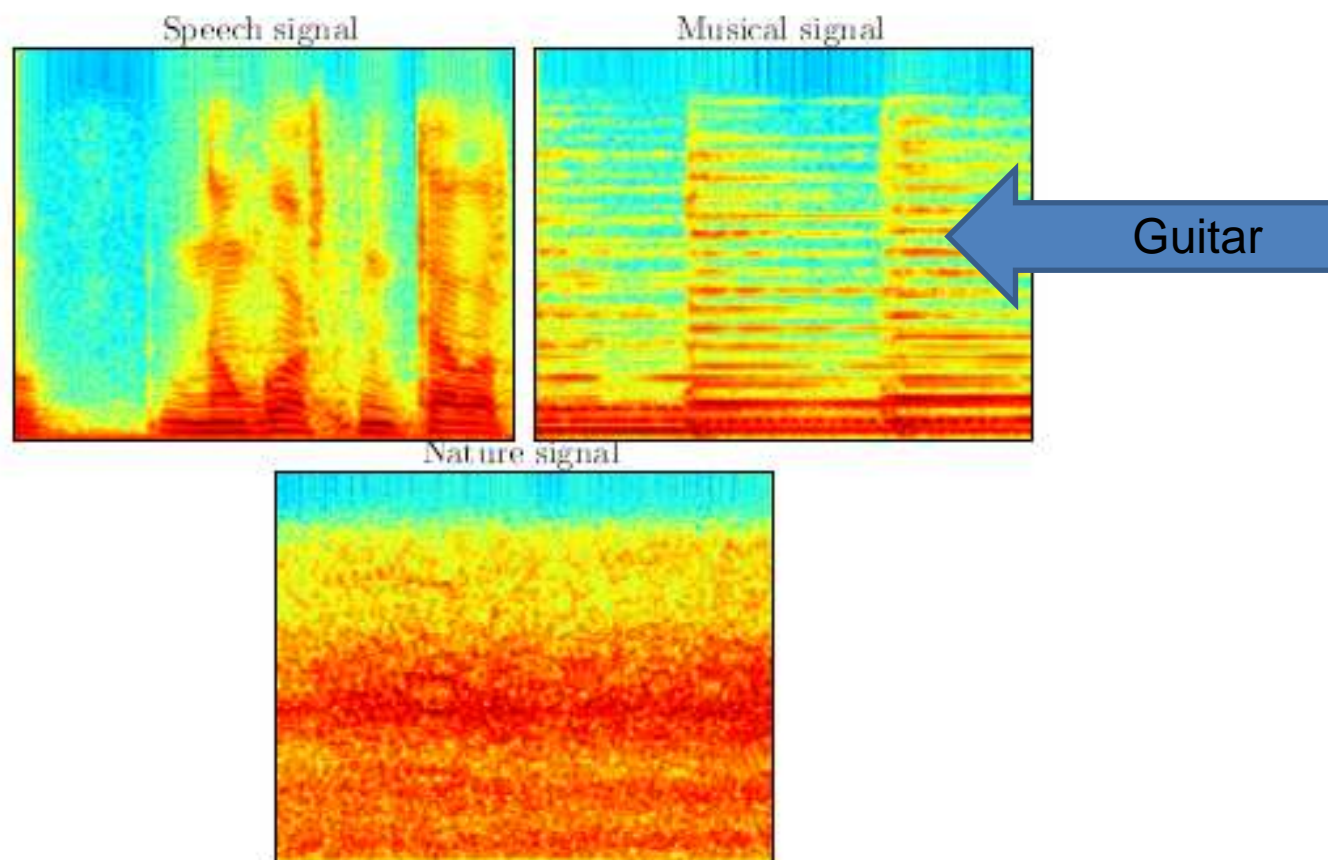


MIR: Background



Sounds produced by musical instruments are ...

- almost **periodic vibrations**
- a combination of **different frequencies**
- all multiple integers of a **fundamental frequency** (called *F0*)



MIR: Three basic features



- **Pitch**
 - based on the fundamental frequency
 - low/deep to high/acute
- **Intensity**
 - the intensity of the amplitude
 - the energy of the vibration
- **Timbre**
 - sound characteristics that allow listeners to perceive as different two sounds with same pitch and same intensity

The User



- Three different intentions
 - Listen to particular performance / musical work
 - Building a collection of music
 - Verifying or identifying works
- Information need of users
 - Formalization of need often not easy
 - Unexperienced vs. professional user
 - Query-by-example easier
 - Possible task: Automatic playlist generation

Music Processing: Melody



- Melody is key feature
 - Many genres have single relevant melody line
 - Discrimination even without rhythm
 - Eventually easy to extract (e.g. Midi)
 - Main feature for query-by-humming
- Retrieved & indexed using n-grams
 - Short sequences of same length
 - Segmentation remains issue



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Music Processing: Harmony



- Chord sequences are considered as relevant descriptors
- Extraction is challenging task
 - Transcription even for user hard

Music Processing: Timbre



- Most difficult feature to characterize
 - defined as acoustic feature that is neither pitch nor intensity
 - Mainly related to the spectrum
- Timbre parameters are left to choice of performers
 - At least in Western classical music
 - Not formalized through transcription
- Listeners are very sensitive to change in timbre

Music Processing: Orchestration



- How the particular work is orchestrated
- Described through musical instruments
 - Style of play is more a matter of timbre
- Recognition of musical instruments
 - Main way of extraction
 - Used rarely in MIR
 - Recognition rates rather high
 - 100% for easily recognizable instruments (like flute)
 - 75% for harder tasks (like chello)
 - 80% on average on large instrument databases

Music Processing: Rhythm



- Intuitively easily recognizable
 - Assumption comes from Western music
 - Africa & Eastern Europe -> highly complex task
- Pop and Rock music simple examples
 - Rhythm is based on variations
 - Four equally spaced beats
 - 1st and 3rd are stronger
- Tempo Tracking
 - relevant for dance / mix / radio

Music Information Retrieval: Examples



Pandora ... www.pandora.com

- Pandora is a personalized internet radio
 - Selection of 'songs one likes'
 - Stream composed on music retrieval
- Audio Genes (fingerprinting)
 - Retrieve music with similar content
- Recommenders and Classifications
 - Based on several characteristics like heavy guitars, impressive voice, etc.

Music Information Retrieval: Musipedia.org



- Portal for searching music
- Several different search options
 - Keyboard search
 - Contour search
 - Query by humming
 - Rhythm search
- Example: Contour search
 - U ... Up, D ... Down, R ... Repeat
 - DDUUUDRDR - Austrian National Anthem
 - UUDUDDDUUDDDDUDU – Haydn, Emperor's Hymn

Thank you ...



... for your attention