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Self-Organizing Maps for Sound Corpus Organization

MASTER'S THESIS

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Eidesstattliche Erklärung

Hiermit erkläre ich, dass ich die vorliegende Arbeit selbstständig und eigen-
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Berlin, den February 10, 2019

Jonas	Μ	ar	g	ra	ıf									



Zusammenfassung	Die Zusammen	fassung auch au	f Deutsch.	

Acknowledgements

This is where the thank yous go.

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1 Introduction 1

1 Introduction

This is the Introduction. Here's a citation about Self-Organizing Maps (SOMs)(Kohonen, 1990).

- 1.1 Motivation and Problem Description
- 1.2 Aims and Objectives
- 1.3 Previous Work

2 Background

This is the Background section.

2.1 Audio Feature Extraction

Make sure to quote Lerch (2012), Rawlinson et al. (2015), Rawlinson et al. (2019a), Mathieu et al. (2010) Mathieu et al. (2019).

2.1.1 Fundamentals

2.1.2 Audio Pre-Processing

2.1.3 Time-Domain Features

Define x[n], n

2.1.3.1 Root Mean Square (RMS) measures the power of a signal (Lerch, 2012, p.73f). It describes sound intensity and is sometimes used as a simple measure for loudness (Rawlinson et al., 2019b) that does not take the nonlinearity of human hearing into account (Fletcher and Munson, 1933). It is calculated for an audio frame x[n] consisting of n samples such that

$$v_{RMS} = \sqrt{\frac{\sum_{i=1}^{n} x(i)^2}{n}}.$$
(1)

2.1.3.2 Zero-Crossing Rate (ZCR) represents the rate of the number of sign changes in a signal. It can be used as a measure of the tonalness of a sound (Lykartsis, 2014) and as a simple pitch detection method for monophonic signals (de la Cuadra, 2019). It is defined as

$$v_{ZCR} = \frac{1}{2 \cdot n} \sum_{i=1}^{n} |sgn[x(i)] - sgn[x(i-1)]|.$$
 (2)

2.1.4 Frequency-Domain Features

Define N_{FFT} , X(k)

2.1.4.1 Spectral Centroid is a measure of the center of gravity of a spectrum. A higher value indicates a brighter, sharper sound (Lerch, 2012). The spectral centroid is defined as

$$v_{SC} = \frac{\sum_{k=0}^{N_{FFT}/2-1} k \cdot |X(k)|^2}{\sum_{k=0}^{N_{FFT}/2-1} |X(k)|^2}.$$
 (3)

2.1.4.2 Spectral Flatness is a measure for the tonality or noisiness of a signal, defined as the ratio of the geometric and arithmetic means of its magnitude spectrum. Higher values indicate a flatter (and therefore noisier) spectrum, whereas lower values point towards more tonal spectral content. It is defined as

$$v_{SFL} = \frac{\sqrt[N_{FFT}/2]{\prod_{k=0}^{N_{FFT}/2-1} |X(k)|}}{(2/N_{FFT}) \cdot \sum_{k=0}^{N_{FFT}/2-1} |X(k)|}.$$
 (4)

2.1.4.3 Spectral Kurtosis indicates whether a given magnitude spectrum's distribution is similar to a Gaussian distribution. Negative values result from a flatter distribution, whereas positive values indicate a peakier distribution. A Gaussian distribution would result in a value of 0. Spectral Kurtosis is defined as

$$v_{SKU} = \frac{2\sum_{k=0}^{N_{FFT}/2-1} (|X(k)| - \mu_{|X|})^4}{N_{FFT} \cdot \sigma_{|X|}^4} - 3,$$
 (5)

where $\mu_{|X|}$ represents the mean and $\sigma_{|X|}$ the standard deviation of the magnitude spectrum |X|.

2.1.4.4 Spectral Skewness assesses the symmetry of a magnitude spectrum distribution. It is defined as

$$v_{SSK} = \frac{2\sum_{k=0}^{N_{FFT}/2-1} (|X(k)| - \mu_{|X|})^3}{N_{FFT} \cdot \sigma_{|X|}^3}.$$
 (6)

2.1.4.5 Spectral Slope represents a measure of how sloped or inclined a given spectral distribution is. The spectral slope is calculated using a linear regression of the magnitude spectrum such that

$$v_{SSL} = \frac{\sum_{k=0}^{N_{FFT}/2-1} (k - \mu_k)(|X(k)| - \mu_{|X|})}{\sum_{k=0}^{N_{FFT}/2-1} (k - \mu_k)^2}.$$
 (7)

2.1.4.6 Spectral Spread is a descriptor of the concentration of a magnitude spectrum around the Spectral Centroid and assesses the corresponding signal's bandwidth. It is defined as

$$v_{SSP} = \frac{\sum_{k=0}^{N_{FFT}/2-1} (k - v_{SC})^2 \cdot |X(k)|^2}{\sum_{k=0}^{N_{FFT}/2-1} |X(k)|^2}.$$
 (8)

2.1.4.7 Spectral Rolloff measures the bandwidth of a given signal by calculating that frequency bin below which lie κ percent of the sum of magnitudes of X(k). Common values for κ are 0.85, 0.95 (Lerch, 2012) or 0.99 (Rawlinson et al., 2019b). It is defined as

$$v_{SR} = i \left| \sum_{\substack{k=0 \ k=0}}^{i} |X(k)| = \kappa \cdot \sum_{k=0}^{N_{FFT}/2-1} |X(k)| \right|$$
 (9)

2.1.5 Perceptual Features

2.1.5.1 Total Loudness represents an algorithmic approximation of the human perception of a signal's loudness based on Moore et al. (1997), which uses the Bark scale as introduced by Zwicker (1961). The Total Loudness is the sum of all 24 bands' specific loudness coefficients, defined by Peeters (2004) as

$$v_{TL} = \sum_{i=1}^{24} v_{SL}(i), \tag{10}$$

where

$$v_{SL}(i) = E(i)^{0.23} (11)$$

is the specific loudness of each Bark band (see Moore et al. (1997) for further details).

2.2 Self-Organzing Map

The *self-organizing map* (SOM) is a machine learning algorithm for dimensionality reduction, visualization and analysis of higher-dimensional data. Sometimes also referred to as *Kohonen map* or *network*, it was introduced in 1981 by Teuvo Kohonen (Kohonen, 1990).

The SOM is a variant of an artificial neural network that uses an unsupervised, competitive learning process to map a set of higher-dimensional observations (the input vectors) onto a regular, often two-dimensional grid or map of neurons or nodes that is easy to visualize. The SOM can be regarded as a nonlinear generalization of a principal component analysis (PCA) (Yin, 2007) or as a quantization of the input data, with the nodes along the map functioning as pointers into that higher-dimensional space. Each node has a position on the lower-dimensional grid as well as an associated position in the input space, which takes the form of a d-dimensional weight vector $m = [m_1, ..., m_d]$, where d is the number of dimensions of the input vectors. Nodes that are in close proximity to each other on the SOM will also have similar weight vectors (Vesanto et al., 2000).

For an in-depth look at the algorithm, its variants and applications, as well as an extensive survey of research on SOMs, the avid reader is referred to Kohonen (2001).

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3 Implementation

This is the Implementation.

3.1 Groundwork: CataRT Extension

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4 Evaluation

This is the Evaluation.

4.1 Measuring SOM-Induced Quantization

- 4.2 Online Sound Similarity Survey
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5 Results

This is the Results section.

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6 Discussion

This is the Discussion.

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Appendices

A LaTeX Sources

The \LaTeX sources for this work can be found in XXX.

B Thesis Bibliography

The references used in this work can be found in XXX.

Acronyms

SOM Self-Organizing Map.

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Digital Resource

This page holds a data disk.