Exploring Universal Attribute Characterization of Spoken Languages for Spoken Language Recognition

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Outline

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Introduction

 Spoken language is described with a common set of fundamental units defined "universally"

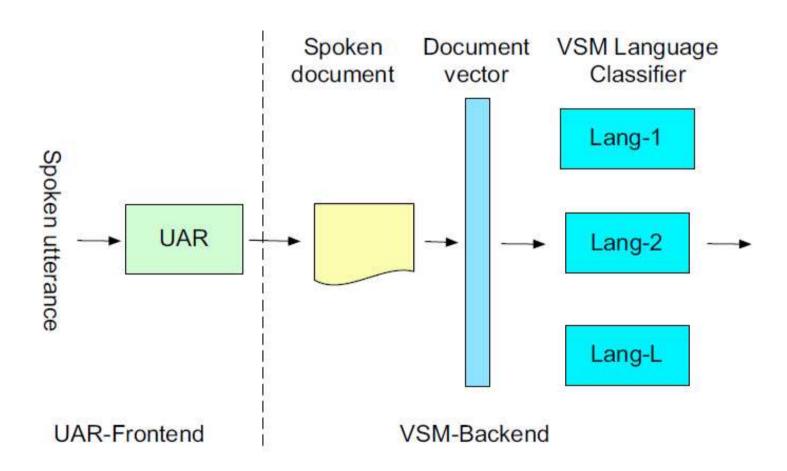
 The advantage of using attribute-based unit is they can be define universally across all language.

Introduction

 Using the <u>vector space modeling</u> (VSM) approaches to language identification (LID) a spoken utterance is first decoded into a sequence of attributes.

A feature vector consisting of <u>co-occurrence statistics of attribute units</u> is created, and the final LID decision is implemented with a set of vector space language classifiers.

System Overview



UAR-Frondend

 Tokenize all spoken utterances into sequences of speech unit using a universal attribute recognizer (UAR).

 Two phoneme-to-attribute table are created that are <u>phoneme-to-manner</u> and phoneme-to-place.

 Manner-based and place-based transcriptions representing <u>speech</u> <u>documents</u> are produced for each speech utterance.

 Each transcription is converted into a vector-based representation by applying LSA.

LSA (latent semantic analysis)
is a three step procedure:

1. term-count vector is created by counting the number of times each term appears in the speech document.

2. term-document matrix, $W = \{\omega_{i,j}\}$, consists of weighted count values given by

$$w_{i,j} = \left[1 + \frac{1}{\log N} \sum_{j=1}^{N} \frac{n_{ij}}{n_{i.}} \log \frac{n_{ij}}{n_{i.}}\right] \frac{n_{ij}}{n_{.j}}$$

where n_{ij} is the number of times term i occurs in document j, and n_i is the number of times that term i appears in the N training documents, and n_{ij} is the number of terms in document j.

 term-document matrix is quite <u>sparse</u> since many higher order *n*-grams do not appear in training documents.

3. Use <u>singular value decomposition</u> (S.V.D) to reduce the dimensionality and improve the sparsity problem.

 The <u>OGI-TS</u> corpus is used to train the articulatory recognizer. This corpus has phonetic transcriptions for six language.

Table 1: Amount of recorded speech of the OGI-TS corpus in terms of hours per each language.

	I		1	I .	MAN		1
Train.	1.71	0.97	0.71	0.65	0.43	1.10	5.57
Valid.	0.16	0.10	0.07	0.06	0.03	0.10	0.52
Test	0.42	0.24	0.17	0.15	0.03 0.11	0.26	1.35

 CallFriend corpus is used for training the back-end language models.

• Test are carried out on the NIST 2003 spoken language evaluation material.

- Language recognition results are reported in terms of <u>equal error rate (EER)</u>, which is the point where the <u>rate of false alarms</u> equals the <u>rate of false rejections</u>.
- Manner-based UAR-VSM (UMR-VSM) system
- place-based
 UAR-VSM (UPR-VSM) system

