Speech Indexing

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Introduction

- Every Organization chooses how much metadata attaches to its multimedia collections.
- "Information is in the audio, video is for entertainment" Richard Schwartz.
 - Using Automatic Speech Recognition Technologies:
 - Speech -> Text + Linguistic annotations.







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Other Clues;

- Information about the speaker:
 - Gender, Background, Emotional State
- Then, if the objective of the recognition of the words spoken is to support retrieval it is called Spoken Document Retrieval (SDR)



Feature Extraction

- Digital acoustic signals transformed into a compact representation that captures the characteristics of the speech signal.
- Vector of features for every 10ms in 16ms overlapping windows.





- The coefficients can be the MFCC (Mel Frequency Cepstral Coefficients)
 - Take the Fourier Transform of selected window.
 - Map the amplitudes into the Mel scale.
 - Take the Discrete Cosine Transform.



- Acoustic Modeling
 - Observations represented as O
 - Task:
 - Find the sequence of words $W = \{w_{1}, w_{2}, \dots, w_{N}\}$ most likely to match O
 - Choosing the highest probability:

 $\hat{W} = \underset{w}{\operatorname{arg\,max}} P(W|O)$

Then:

$$P(W|O) = \frac{P(O|W) \cdot P(W)}{P(O)}$$



- Popular approach in speech recognition is the use of hidden Markov Models (HMMs)
- HMMs represent connected states each one having a transition probability.
- The problem goes P(O|W) to P(O|M)
 - M represents the sequence of a word associated to W. Each model M can be a phone (smallest unit in speech)



- Language Modeling
 - P(w) can be expressed as:

$$P(W) = \prod_{i=1}^{n} P(w_i | w_{1, \dots, w_{i-1}})$$

- The probability of w_i being spoken preceding the sequence (w₁ ... w_{i-1})
- For long history it is not feasible.
- Use Markov assumption:
 - "The probability of a future event can be predicted by looking at its immediate past"



- N-grams, number of previous words.
 Usually 1 or 2.
 - Two-word history, trigram models can be generated as:

$$P(w) \approx P(w_0) \cdot P(w_1 | w_0) \cdot \prod_{i=2}^{n} P(w_i | w_{i-1}, w_{i-2})$$



Dictionary

- Vocabulary: List of all the words in the language model.
- Pronunciation Dictionary: Link between the acoustic model and the language model.
- Usually represented as lexical trees.



- How to obtain word pronunciations:
 - Manually.
 - Time Consuming
 - G2P tools (Grapheme to Phoneme)
 - Produces the *average* pronunciation.
 - Pronunciation varies according to age, gender and dialect. 40% of the words are not correctly pronounced.
 - To override this issue include pronunciation variations in the lexicon.





- To support document retrieval is necessary to create a textual representation.
- Depending on companies and/or libraries management, it may exist:
 - Human generated metadata.
 - Bibliographic (tombstone) information.
- However, it is still difficult to find specific

passages.

55 kilos ...





- Manual VS Automatic
 - Instead of speech recognition technology do it manually.
 - Minutes for meetings are done this way and, if there is video, the annotations can be very useful.
 - Manually means people, people means money. Therefore, manually is expensive.



ASR (Automatic Speech Recognition)
 has largely improved lately.

Techniques

- Synchronization of available textual resources.
 - Collateral textual resources that are closely related can be exploited.
 - e.g., Subtitle information for the hearing impaired. (some even provide boundaries)
 - The time labels from the sources are crucial for the indexing.
 - If there are no time labels some synchronizations shall be done.



- Synchronization technique:
 - A SRS is used to generate a "inaccurate" transcript of the speech -> Hypothesis
 - The Hypothesis is aligned to the Minute using dynamic programming.
 - Where the Hypothesis and Minute match generate an "anchor"
 - Using the timing from the SRS and the "anchors" generate Segments.
 - Finally, individual segments of audio and text are accurately synchronized.





- Large vocabulary speech recognition.
 - When a system produces errors successful retrieval will be doubtful.
 - A speech recognition performance of 50% is the minimum for a useful performance.
 - Todays systems require:
 - Speaker-independent, trained using large amount of example audio from the domain.
 - Large Vocabularies: 65,000 words.
 - Out of the Vocabulary
 - When a word is not in the vocabulary it can not be recognized and will not appear in the minute.



 e.g., BBN technologies has 1.8M of words for training and an America English GigaWord News corpus of 1 Billion words of text.



- Keyword spotting.
 - Feasible approach when computer power is limited.
 - Keywords are usually fixed in advance.
 - Method works for a restricted domain.
 - e.g., weather reports
 - Useful when heavy-weigh speech recognition is not feasible.



- Sub-word unit representative.
 - Previous presented approaches focus on words as units of the decoded speech.
 - An alternative is to use sub-word units, such as *phones*.
 - The document is represented in terms of these sub-words units.
 - Phone recognizer requires only an acoustic model and a small phone grammar. It is much faster than large vocabularies approach.



- Is less sensitive to out-of-vocabulary words.
- However, tends to produce higher error rates. Because it is based solely on acoustic information.





- For retrieval purpose it is important to have the *content words* right.
 - e.g., nouns, names, adjectives
- The indexing process will discard the rest of the words.
 - e.g., articles
- Analyzing the global word error may not be adequate.



- Reference is the original transcript and the Hypothesis is the generated transcript.
- The word error rate WER is calculated as follow:

 $WER = \frac{Insertions + Deletions + Substitutions}{Total words in Reference}$



• The term error rate TER is calculated as follow: $\sum_{k=1}^{\infty} |R(t)-H(t)|$

$$TER = \frac{\sum_{t \in T} IR(t)}{\sum_{t \in T} R(t)}$$

Where t represents a given query.

 The TER gives a more accurate measure of speech recognition performance.



- Query and Document expansion
 - This technique simply adds words to the query in order to improve retrieval.
 - Process:
 - After initial run from the top N most relevant documents,
 - Select the top T terms and add them to the query to enrich it.





- Vocabulary Optimization
 - For success is necessary to minimize the out-of-vocabulary words.
 - Selection of appropriate set of vocabulary words reassembling the domain.
 - The maximum number of words that can be included in the dictionary is restricted.
 - Typically to 65,000 words.



- With topics changing is necessary to revise the vocabulary and update accordingly.
 - Include new words (new topics)
 - Remove obsolete words (old topics)
- There are words that occur in certain times of the year.
- Some words appear suddenly and can not be "foreseen"





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- Topic-based language models
 - Words are selected with a focus on a specific segment of an audio.
 - Include the *n*-grams specific to the topic.
 - <u>bank</u>
 - the interest rate in the <u>bank</u> more
 - the <u>bank</u> of the river less



- Topic-based requires 5 steps:
 - 1) Segmentation of the audio file.
 - Segments can be interpreted to be on the same topic.
 - In practice, segmentation is not known. Then change of speaker, silence intervals, etc. are used.
 - 2) Speech recognition on the segments.
 - 3) Definition of the *topic*
 - Using the transcripts and collateral text.
 - Collateral text can come from different sources:
 - e.g., newspaper, topics database



- 4) Generation of the topic specific language model.
 - Using a ranked list of similar documents.
- 5) Speech recognition using the topic-based language model.



- Acoustic Adaptation
 - To be robust a speech recognition system shall have good performance even when the quality of the input is low.
 - Background Noise, Cross-Talk, Low Audio Quality
 - Speaker to Speaker characteristics vary due to:
 - Vocal tract, Age, Speaking style, etc.



- To override the problem more Training can be done, or
- apply normalization and dynamic adaptation procedures.
 - Start with a stable baseline and tune it to the specific conditions in the task domain.
- A) Vocal Tract Length Normalization
 - The average length of the tract is 17cm, but it varies with dimension of the person.



- Vocal tract length normalization technique aim is to compensate acoustic differences.
- Normalization from the cluster of speakers to the "generic" speaker.
- The normalization is done by warping the frequency axes.







B) MAP and MLLR Adaptation

- Maximum A Posteriori
- Maximum Likelihood Linear Regression
- These methods aim at adjusting the model parameter not the spectral information.
- Model adaptation can be done off-line or at recognition time (online)
- MLLR aims to capture the general relationship between the speaker independent modal set and fit it to the adaptation data.



- MAP combines the information from the adaptation data and some prior knowledge about the model.
- A disadvantage is that large number of adapted models may be generated.



Cross-media Mining

- Go one step further:
 - Exploit collateral or text resources that are accessible.
 - Mining these resources can support access to different perspectives on the available data.
 - Linking newspaper articles with broadcast items.
 - Available semantic annotations for documents with similar profiles can be exploited to improve the searching.

Cross-media Mining

Example:

- Video conference in a company:
 - Speech Recognition of the Video (transcript).
 - Minutes.
 - Reports.
 - Information of people attending the meeting.



Conclusions

- For Speech Indexing and Retrieval it is necessary to have a transcript of the contents of the multimedia data.
- How:
 - Speech Recognition.
 - Synchronization of Data Sources.
 - Cross-media Mining.
- Do not forget:
 - That every approach has certain advantages and disadvantages. Which to use depends on the application.





Henk Blanken, et al. Multimedia Retrieval. Springer. 2007. [Speech Indexing Ch. 7]

