COMPUTER-BASED STUTTERED SPEECH DETECTION SYSTEM USING HIDDEN MARKOV MODEL

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To my beloved family and friends

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ABSTRACT

Stuttering has attracted extensive research interests over the past decades. Most of the available stuttering diagnostics and assessment technique uses human perceptual judgment to overt stuttered speech characteristics. Conventionally, the stuttering severity is diagnosed by manual counting the number of occurrences of disfluencies of pre-recorded therapist-patient conversation. It is a time-consuming task, subjective, inconsistent and easily prone to error across clinics. Therefore, this thesis proposes a computerized system by deploying HMM-based speech recognition technique to detect the stuttered speech disfluency. The continuous Malay digit string has been used as the training and testing set for fluency detection. Hidden Markov Model (HMM) is a robust and powerful statistical-based acoustic modeling technique. With their efficient training algorithm (Forward-backward, Baum-Welch algorithms) and recognition algorithm, as well as its modeling flexibility in model topology and other knowledge sources, HMM has been successfully applied in solving various tasks. In this thesis, a set of normal voice for digit string as database is used for training HMM. Then, the pseudo stuttering voice was collected as testing set for proposed system. The generated experimental results were compared with the results made by Speech Language Pathologist (SLP) from Clinic of Audiology and Speech Sciences of Universiti Kebangsaan Malaysia (UKM). As a result, the proposed system is proven to be capable to achieve 100% average syllable repetition detection accuracy with 86.605% average sound prolongation detection accuracy. The SLP agreed with the result generated by the software. This system can be further enhanced for detecting stuttering disorder for daily speaking words where Microsoft Visual C++ 6.0 and Goldwave have been used for developing the software which can be executed under the window-based environment.

ABSTRAK

Sejak beberapa dekad yang lalu, kegagapan telah menarik minat bagi para-para pengkaji dan ahli terapi. Kebanyakan ujian diagnostik bagi kes-kes gagap masih menggunakan pemerhatian dan persepsi dari pakar atau ahli patologi untuk mengesan tahap keterukan kegagapan. Biasanya, ujian tersebut dilaksanakan secara konvensional dengan mengira jumlah ketidaklancaran percakapan yang dilakukan oleh penyakit. Teknik ini sangat memakan masa, terlalu subjektif, tidak konsisten dan mudah terjadinya ralat yang disebabkan oleh faktor kepelbagaian klinik. Oleh itu, tesis ini mencadangkan suatu sistem komputer yang menggunakan teknik pengecaman suara berasaskan *Hidden Markov Model (HMM)* untuk mengesan ketidaklancaran pertuturan seseorang pesakit. HMM merupakan sebuah teknik permodelan akustik berasaskan statistik yang popular dan canggih. Dengan mengaplikasikan algoritma-algoritma permodelan seperti *Baum-Welch, Forward-backward* dan serta teknik permodelan topologi serta yang lain-lain., ia telah menyelesaikan pelbagai masalah yang dihadapi dalam projek ini. Sebuah pengkalan data yang mengandungi set suara pertuturan rentetan digit melayu yang berterusan dan normal telah dikumpul bagi latihan HMM. Seterusnya, satu set suara *pseudostuttering* telah dikumpul sebagai pangkalan data suara untuk menguji keberkesanan fungsi sistem yang dicadangkan ini. Keputusan eksperimen dan penyelidikan telah dibandingkan dengan keputusan yang dihasilkan oleh ahli patologi dari Klinik Audiologi dan Sains Pertuturan, Universiti Kebangsaan Malaysia. Sistem cadangan ini telah terbukti mampu mengesan sikap kegagapan secara automatik iaitu mencapai purata ketepatan sebanyak 100% dalam mengesan bunyi pengulangan suku kata dan 86.605% dalam mengesan bunyi pemanjangan. Walaubagaimanapun, sistem cadangan ini masih dapat dipertingkatkan lagi dengan menggunakan vokabulari pertuturan harian di mana Microsoft Visual C++ 6.0 dan Goldwave telah digunakan untuk pembangunan projek ini.

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LIST OF ABBREVIATIONS

GMMs	-	Gaussian Mixture Models
HMM	-	Hidden Markov Model
SLP	-	Speech Language Pathologist
MFCC	-	Me-Frequency Cepstral Coefficient
GUI	-	Graphical User interface
DCT	-	Discrete Cosine Transform
SP	-	Sound Prolongation
SR	-	Syllable Repetition

LIST OF SYMBOLS

π_i	-	Initial state distribution
λ	-	HMM model
μ_{jm}	-	Mean vector for the mth mixture component at state j
$\delta_t(i)$	-	HMM best score
$\alpha_t(i)$	-	Forward variable
$\beta_t(i)$	-	Backward variable
a_{ij}	-	State transition probability
a_p	-	LPC coefficients
A	-	State transition probability distribution
API	-	Application Programming Interface
$b_j(k)$	-	Emission probability
B	-	Emission probability distribution
C_{jk}	-	Weight coefficient for the mth mixture component at state j
F_0	-	Fundamental frequency
M	-	Number of HMM observations symbols per state
$mel(f)$	-	Mel-scale frequency
O	-	Observation probability
q_t	-	HMM state at time t
S	-	HMM state
$W(n)$	-	Hamming window function
E	-	Energy
$b_j(x)$	-	Continuous probability density function (pdf)
N	-	Number of HMM States
Σ_{jm}	-	Covariance Matrices

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CHAPTER 1

INTRODUCTION

1.1 Background of Research

Stuttering is a fluency disorders associating with the disruption or disfluency of speech flow in a person (Yaruss and Reardon, 2004). The patients suffer from this disorder confront with difficulties in communication. Thus, they invariably deal with embarrassment attributable to the anxiety and fear in communicating. There is no explicit cause of stuttering in a person. A variety of causes have been suspected but scientists believe that many forms of stuttering are genetically induced (Guitar, 2006).

There are various kinds of speech disfluency types which include interjections, revisions, incomplete phrases repetitions, word repetitions, part-word-repetitions, prolonged sounds, broken words and so on, however, several types of the disfluency speech are related to stuttering. The sound or syllable repetition, monosyllabic whole word repetition, sound prolongation and blocks are categorized as stuttering primary behaviors (Ambrose, 1999).

In stuttering therapy field, the pathologists usually implement an assessment session by using manual counting of the occurrence frequency of speech disfluencies exhibited by patients; the frequency of speech disfluency is one of the criteria that reflect the recovery progression during speech therapy treatment. Conventionally,

the assessment is performed by reviewing the recorded tape while counting the number of occurrences of disfluencies speech during the therapist-patient conversation. However, it is time-consuming, subjective, inconsistent and error-prone across clinics (Cordes et al., 1992; Kully and Boberg, 1988). The influence of a clinician's subjective opinion as each clinic using slightly different definitions of stuttering events, and mistakes in the counts (Yaruss, 1997) lead to the inconsistency in assessment. Nevertheless, stuttering has become an interest subject of researchers and pathologist from various domains such as signal analysis, speech pathologist and so on. More and more therapy programs such as Overall Assessment of the Speaker Experience of Stuttering (OASES), Stuttering Severity Inventory (SSI), Lidcombe Program and other tools have been proposed and studied. Less effort has been done on the research of computer-based stuttering speech diagnostics and assessment system (Ooi, 2007). Thus, in this work project, the studies of the stuttering etiology such as nature and characteristics of stuttering, existing stuttering diagnostics and severity tools were studied in order to deal with the underlying problems by developing an automated Malay stuttered speech detection system which was dedicated to provide the more objective supplementary tools to improve the weakness of the perceptual-based conventional stuttering diagnostics methods.

1.2 Problem Statements

Stuttering has become the interesting research issues over past decades. Professionals from different domains exerted utmost effort in developing stuttering assessment methods for the purpose to optimize the performance of the assessment the stuttering behavior such as Stuttering Severity Instrument (SSI-3), Lidcombe Program and others. However, each of the existing conventional assessment technique has its own availability, validity, and inadequateness; the existing problems are discussed as below:

a) Reliability

Researchers and clinicians have been debating on how to best diagnose the stuttering behaviors based on the aspect of the validity and reliability concerns (Carol, 1998). It includes the attempts to identify and quantify the number of occurrence of observable stuttering behaviors accurately such as involuntary, audible or silent, repetitions or prolongations in the utterance of short speech elements (Wingate, 1964). However, such perceptual-based conventional counting measurement may be reliable within clinics but may not consistent across clinics (Kully and Boberg, 1988). Indeed, Cordes and Ingham (1995) demonstrated relative agreement among professionals who work at the same facility but relative disagreement among professionals who work at the different facilities. In addition, the research examination of Packman *et al.* (1993) discovered that seven clinicians who work in the same clinic found that the number of stutters counted from set of given speech samples differ when the clinicians re-counted the same samples at a later time(Thomas and Howell, 2001), the inconsistency of stuttering severity rating was being. Cordes *et al.* (1992) also reported that the degree of experience of a judge in detecting stuttering speech disfluency forms becomes an important factor to ascertain the accuracy of stuttering severity rate of person who is stutter(PWS).

b) Time-Consuming, Subjective and Error-prone

The conventional method of the assessment is to count and classify the observed stuttering behaviors in pre-recorded sample manually by speech therapist. It includes the analysis of sound of dysfluency types such as fast repetition, short pause, unusual lengthening and so on. However, the manual analytical method is a time-consuming task; it is because therapist needs to vacate excessive time that involves the process of manually counting the number of disfluency occurrences in pre-recorded sample. However, such counting processes are subjective and error-prone. Therefore, an automated robust system is required to provide more objective and accurate results instead of using perceptual-based counting methods; the computer-based is also capable to generate a large quantity of results within a short period of time to enable therapist vacate more time on interaction with patients in such a way to monitor their recovery progression more effectively.

c) Cost

The computer-based stuttering assessment system is capable to provide the cost benefit expectation. It can optimize the use of therapists by reducing the numbers of staff involve in analyzing the speech stuttering behaviors of patient. Generally, the cost of the therapy session is high and becomes a burden for the patients; the longer assessment is taken for a patient, the higher of the therapy cost (Ooi, 2007). There are group of individuals that are not affordable for the long-term therapy process, because some of the insurance companies does not cover the rehabilitation fees especially for the fluency disorders in their insurance plan or policy. Thus, proposed system can help patient to reduce the duration of diagnostic session and as a result to minimize the cost of a clinical session.

1.3 Objectives

The main objective of the thesis is to develop a computer-based Malay stuttered speech detection system with high accuracy that is able to mend the inconsistency problems of the conventional perceptual judgments on occurrences of the stuttering behaviors. It includes speech recognition, identification and classification of various kinds of stuttering speech behaviors such as syllable repetition, sound prolongation and so on with high accuracy.

The second objective of the thesis is to design a phonetically balance Malay pronunciation dictionary for any acoustic recognition units in consideration of the robustness of acoustic modeling.

Third objective of the thesis is to design a language model decoding network (speech decoder) that is feasible to recognize and detect various stuttering speech behaviors.

Finally, an automated malay stuttered speech detection report has been designed carefully for the speech therapist to follow closely the recovery progression of people who is stutter(PWS).

1.4 Scopes

In recent years, the research development of speech recognition has achieved a matured level. However, In Malaysia, the domain of applying the speech recognition technology into Malay stuttering therapy field is still in the preliminary stage. Far less effort has been done on the computer-based stuttering assessment system. If the project is to be started from scratch, it would take a very long time before the malay stuttering recognition and detection system can be finalized, used for detail and in-depth study. Alternatively, utilizing the available development tool give a good jump-start towards the creation of the automated stuttering recognition and detection system, and shorten the development time. Therefore, a hidden markov toolkit (HTK) engine has been selected to be utilized during the development of this master thesis. Due to time and contextual constraints, the scope of the thesis has been dedicated as below:

1. Existing of multiple types of the stuttering speech behaviors can be exhibited by stutterer. However, the study focuses on creating a stuttered speech recognizer which is capable to detect two core stuttering behaviors associated with syllable repetition and sound prolongation with high recognition accuracy.
2. Mel Frequency Cepstrum Coefficient (MFCC) is utilized to extract speech features by converting the waveforms to the parameterize forms for the use of HMM.

3. The stochastic-based continuous density hidden Markov model (CDHMM) with Gaussian mixture is used for acoustic modeling.

4. Embedded Baum-Welch training integrated with forward-backward algorithm is applied for re-estimate and compute the optimal parameters of each HMM.

5. The vocabulary consisting of Bahasa Malaysia continuous utterances which are digit zero (“kosong”) to nine (“sembilan”) (see Appendix E) is collected in quite office environment with 16 bits format and sampled at 16 kHz using fixed high-quality microphone Shure model SM48.

6. Pseudostuttering database which consists of Malay continuous digit stuttered utterance associated with repetition of sound and sound prolongation is collected and verified by Dr. Etain Vong who is the stuttering specialist from Clinic of Audiology and Speech Sciences in UKM for the purpose to ensure the collected pseudostuttering samples are analogous to the real stuttered phenomenon for research purpose in this project work.

7. All the formulated experiments are based on Malay continuous digit speech domain.

8. Spastic Children Association of Johor and Clinic of Audiology and Speech Sciences , UKM assisted in this work project, as well as giving professional feedback.

1.5 Outlines of the Thesis

This thesis is divided into 6 chapters. The first chapter is the introduction of the thesis which explains the problem background, defines problem statement and outlines the project objectives and scopes. In chapter II, the studies is given on the overview of stuttering disorders, followed by the characteristics of the core behaviors and secondary behaviors of stuttering, assessment methods used by pathologists, descriptions of the available stuttering severity tools and so on are discussed in this chapter.

Chapter III reviews the concepts of the speech recognition technique. Acoustic modeling and classification technique like CDHMM, N-gram language model decoding technique, speech signal processing technique and software tools used in this project will be describes in this chapter. Furthermore, the methodology and the design of the proposed system throughout the thesis are explained in Chapter IV. Chapter V discussed all the results obtained in the experiments. The last chapter, Chapter VI gives conclusion and future works to be done.

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