

Multilingual Speaker Identification using analysis of Pitch and Formant frequencies

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Abstract— In the modern digital automated world, speaker identification system plays a very important role in the field of fast growing internet based communications. In India there are many people who are bi-lingual or multilingual, so the requirements to design such system which is used to identify the multilingual speakers. Present paper explores the idea to identify multi-lingual person by basic features. For this the speech signals of three indian languages i.e Hindi, Marathi and Rajasthani are recorded and basic features pitch, first three formant frequency calculated from PRAAT software. The observation has been presented that the pitch and first three formant frequencies F1,F2 and F3 of speaker are increases when speaker change the language from rajasthani to hindi to marathi. The percentage deviation in pitch as well as formant frequencies for Rajasthani and Marathi from hindi are positive and negative respectively for utterance “p”. Similar analysis has been perform for 'k' and '>'. This observation will help to make such system which is used to identify the speaker in multilingual environments.

Keywords- Pitch, formant frequencies, Multilingual Speaker, PRAAT, Percentage deviation.

I. INTRODUCTION

Today's era the speakers knows more than one languages and it is required that the speaker identification system should give the better performance for this type of speakers who are speaking multiple languages. When the speaker recognition is being transferred to real applications, the need for greater adaptation in recognition is required. The performance of the monolingual speaker identification systems tends to decreases when speaker is speaking in another language. Therefore we need to make such systems which can work for multiple languages.

Languages are usually influenced by other languages that are present in the environment and by the speaker's mother tongue[2]. Multilingual speech processing (MLSP) is a distinct field of research in speech and language technology that combines many of the techniques developed for monolingual systems with new approaches that address specific challenges of the multilingual domain [8].

In order to find some statistically relevant information from speech signal, it is important to have mechanisms for reducing the information of each segment in the audio signal into a relatively small number of parameters, or features. Feature extraction is the first step for the multilingual speaker identification system. Many algorithms were suggested/developed by the researchers for feature extraction. The basic features i.e. Pitch and first three formant frequencies will help for identification of person on multilingual base

II. DATABASE GENERATION

For multilingual speaker identification system, the database of different speakers has been recorded. The sampling rate of recorded sentences is 44KHz. The segmentation is done by Goldwave software. The sentences consist consonants i.e च

ष , झ has been considered for the recording. The total number of speakers involved are 20 including males and females. The recorded sentences are:

मुझे चाय पीना पसंद है। चाय मे षक्कर कम है। तिरंगा हमारा झंडा है।
मला चाय पसंद आहे चाय मधे षक्कर कमी आहे तिरंगा अमच्छये झंडा आहे
मन्ने चाय पीनी पसंद है। चाय मो षक्कर कम है। तिरंगा मारा झंडा है।

III. FEATURE EXTRACTION

The original speech signal contains redundant information. For speech recognition and application eliminating such redundancies helps in reducing the computational overhead and also improve system accuracy. Therefore all most speech application involves the transformation of signal to set of compact speech parameter.

In this work mainly pitch and first three formant frequencies F1,F2 and F3 are calculated from the speech signal of different languages. The pitch is fundamental frequency F0 and it is determined by the vibratory frequency of the vocal folds. The standard range of pitch is from 75 Hz to 500 Hz for human voice. Formants are frequency peaks which have, in the spectrum, a high degree of energy. They are especially prominent in vowels. Each formant corresponds to a resonance in the vocal tract and the spectrum has a formant at approximate every 1000 Hz.

IV. RESULT AND DISCUSSION

In this paper, investigation has been made for two basic features pitch and first three formant frequencies F1,F2 and F3 for male and female speakers in three languages Hindi and Marathi and Rajasthani. The analysis is done for the three utterance च , ष and झ in three languages Hindi and Marathi and Rajasthani. Base of the analysis is to observe the variation in pitch and first three formant frequencies F1,F2 and F3 if the

speaker change the spoken language. The following observations were recorded in table-1 and table-2.

Table-1: Pitch and percentage deviation for utterance च.

Speakers	Pitch(Hz)			% Deviation of Hindi to Marathi	% Deviation of Hindi to Rajasthani
	Hindi Language	Marathi Language	Rajasthani Language		
1	160	168	142	-5.00	11.25
2	155	189	149	-21.94	3.87
3	139	148	125	-6.47	10.07
4	169	187	151	-10.65	10.65
5	142	147	140	-3.52	1.41
6	122	165	120	-35.25	1.64
7	141	163	125	-15.60	11.35
8	124	155	121	-25.00	2.42
9	133	173	125	-30.08	6.02
10	144	165	132	-14.58	8.33
11	126	155	125	-23.02	0.79
12	137	169	125	-23.36	8.76
13	160	170	135	-6.25	15.63
14	168	173	145	-2.98	13.69
15	144	150	129	-4.17	10.42
16	143	186	122	-30.07	14.69
17	146	157	136	-7.53	6.85
18	125	172	120	-37.60	4.00
19	145	163	139	-12.41	4.14
20	132	155	122	-17.42	7.58

From Table-, it has been observed that when speaker change the language, the pitch value has changed for च. The pitch values of Marathi language are more as compared to hindi and Rajasthani languages. The percentage deviation Marathi language and rajasthani languages from hindi language is presented in table no-1.

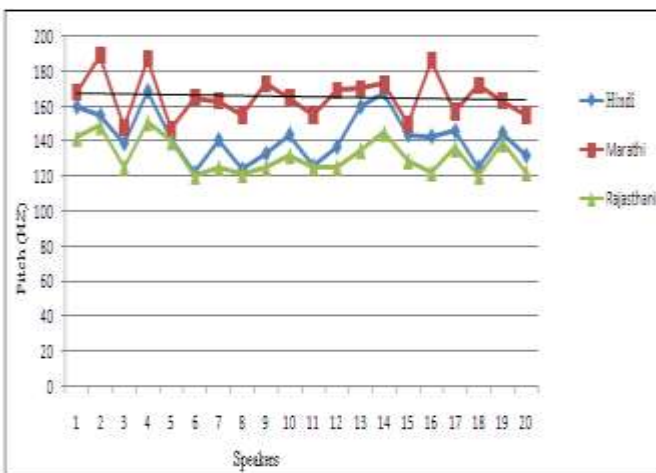


Figure-1: Variation of pitch for three languages of च

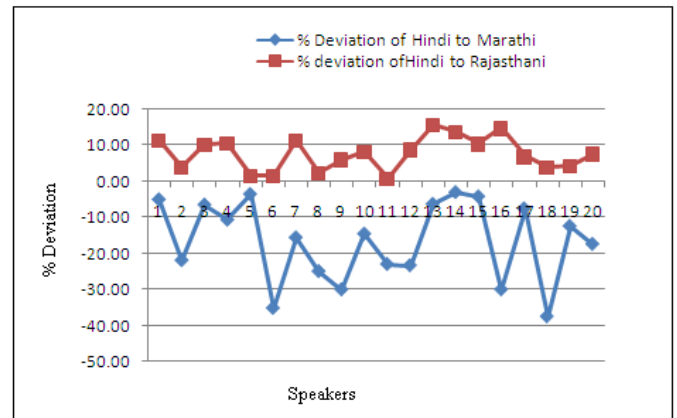


Figure-2: Percentage deviation of Marathi and Rajasthani Languages from Hindi of च

The variation of pitch of च in three languages as shown in figure-1 The observation has been presented that the pitch of speaker are increases when speaker change the language from rajasthani to hindi to marathi. Therefore the percentage deviation in pitch for Rajasthani and Marathi from hindi are positive and negative respectively for utterance “च” which is shown in figure-2. Similar analysis has been perform for ष and झ .

Table-2: % deviation of formant frequencies (F1,F2 & F3) for utterance च

Speaker	F1		F2		F3	
	% Deviation (Hindi to Marathi)	% Deviation (Hindi to Rajasthani)	% Deviation (Hindi to Marathi)	% Deviation (Hindi to Rajasthani)	% Deviation (Hindi to Marathi)	% Deviation (Hindi to Rajasthani)
1	-8.30	18.49	-5.45	17.16	-2.45	6.13
2	-10.19	14.79	-8.69	3.47	-5.12	5.00
3	-0.64	37.09	-4.76	4.85	-2.48	5.51
4	-11.61	28.08	-5.98	1.93	-3.51	4.10
5	-13.70	5.26	-3.27	5.93	-2.77	3.76
6	-7.46	22.71	-0.50	16.89	-7.81	6.00
7	-8.35	17.85	-2.88	11.57	-2.68	5.68
8	-19.35	10.88	-1.92	6.71	-4.56	16.96
9	-19.11	2.78	-4.85	7.25	-8.78	2.71
10	-11.63	14.51	-10.08	16.48	-7.10	4.92
11	-13.07	14.47	-3.36	10.76	-2.01	6.70
12	-11.40	27.43	-6.66	7.91	-5.03	17.07
13	-55.56	11.93	-9.31	3.41	-4.75	4.75
14	-36.29	4.21	-8.95	8.53	-7.16	4.13
15	-10.93	21.34	-2.77	2.66	-4.24	5.08
16	-24.68	12.03	-2.06	0.36	-2.40	18.29
17	-10.96	16.67	-0.56	3.06	-3.45	8.41
18	-10.55	1.23	-6.26	11.76	-2.13	15.99
19	-12.38	17.65	-9.73	1.33	-4.37	5.97
20	-22.42	5.95	-9.94	11.01	-9.41	6.88

In Table-2 the percentage deviation of first three formant frequencies F1,F2 and F3 has been presented. The percentage deviation shows that when the speaker change the language, the first three formant frequencies F1,F2 and F3 has changed. It has been observed that the percentage deviation in formant frequencies F1,F2 and F3 for Rajasthani and Marathi from hindi are positive and negative respectively for utterance “च”. Similar analysis has been perform for च and झ.

V. CONCLUSIONS

Feature extraction has the ability to improve the performance of multilingual speaker identification system. Presented observation, the pitch and first three formant frequencies F1,F2 and F3 of the speakers have changed when the speaker change the spoken language. The pitch and first three formant frequencies F1,F2 & F3 of Marathi lingual speakers has more as compared to hindi and rajasthani lingual speakers. This observation will help to make such system which is used to identify the speaker in multilingual environments.

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