# Applications of Fuzzy set theory in speech recognition

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Abstract

In this paper , a new method is proposed based on fuzzy specch recognition for a speaker-independent . First the paper states the significance of applying the fuzzy set theory to the speech recognition , then briefly introduces the fundamental principle of fuzzy speech recognition , and states the method of setting up a reference dictionary by combining the fuzzy classification with linear matching and DP matching in speech recognition theory . Then it gives the fuzzy set expression of the speech recognized feature , and defines the resemblance degree between the input words and the standard pattern of the reference dictionary according to that expression . Finally , the paper gives the principle block and the implement block of this system . The average recognized rate 98.3% is taken by means of the statistical results from the experiments in this system .

Keywords: Fuzzy expression of the feature, A reference dictionary feature

## 1.Introduction

With the advance of computer technologies , the artificial intelligence technology is continuously permeating every field . People have fully felt the importance of artificial intelligence . The purpose of studying artificial intelligence is to have computer do some intelligent tasks which humans can do . The ultimate task of speech recognition is to transform computer into machines with ears .Therefore , speech recognition is an important component of the artificial intelligence technology .

In recent years , speech recognition technology has developed rapidly , new recognition methods have been brought forward one after another and various recognition systems have appeared[2] , yet , to apply them to practical systems without any problems , we can say we still have a long way to go . One of the main reasons is that the present speech recognition technology is based on accurate mathematical results and binary logic , and this method does not fit in with the human thought and judgement patterns which are based on fuzzy logic or the characteristics of the natural language with fuzziness .

Speech features are usually related to health , character of occupation , age , sex , physiological state , psychological state and mood . Therefore , We can say that fuzziness is one of the essential attributes of the natural language . Correctly describing the characteristics of a speech will directly affect the recognized rate , and so , it is one of the main subjects of study for which many people work . Because the fuzzy theory is an effective means of dealing with fuzziness problems , this paper applies the fuzzy theory to the field of speech recognition , and puts forward a speaker-independent fuzzy speech recognition method .

## 2.Speaker-independent fuzzy speech recognition [2] ~ [5]

### 2.1 The summaization of the basic method

The procedure for computer speech recognition is as follows : First , the computer transfirs the acoustic waves sent by a human which stand for a word meaning into numerical electronic signals , and extracts the features of the speech from these signals as input patterns . Then , according to a matching principle , it matches these input patterns with the patterns in the reference dictionary which were stored in the computer before , and selects the moot similar the pattern from them as the final recognized result . In terms of different reference dictionaries speech recognition can be classified into specified speaker speech recognition and speaker-independent speech recognition . What we call "specified speaker speech recognition " means that , before the computer recognizes a human's speech , we need to store the standard patterns of that human's speech in the computer , and form a reference dictionary . Then , the computer can only recognize correctly that human's speech . When the speaker is changed , the reference dictionary needs to be changed , too . Speaker -independent speech recognition has no limitation on the speaker which means that the reference dictionary is universal .

There are some differences in the speech generation organs of different speakers , so even though the word is the same , the speech features of different humans are different . Therefore , when we set up the reference dictionary of a speaker-independent speech recognition system , we can't analyse only one or two humans' speech , we must analyse a lot of humans' speech features . Since the internale storage space is limited , it is not

possible to register many human's speech features of each word . Usually, the way of setting up the reference dictionary of the speaker-independent speech recognition system is to extract a lot of humans' speech features first , and then , to classify them , and finally , to take the average value of the different classes . This kind of classification is not precise , and it brings a great influence to the recognized rate . Therefore , this paper uses the fuzzy classification method in setting up the reference dictionary , i , e . classifying reasonably the speech features of many humans speaking the same word into different classes ,then , taking the average value within each class . Before classification , in order to make the monolization of the length of the different human's speech feature for the same word and the internal length of relativity in this word , we adopted the linear matching and the DP matching method . The purpose of this is to find out the precise and real resemblance relation which is needed when classifying features .

In this paper the speech signal feature used , is Fourier transform coefficients . As mentioned above , human speech is affected by many fuzzy factors , therefore these coefficients do not give an absolutely accurate expression in this system . We use the fuzzy set which is defined in the value field of the Fourier transform coefficients to describe the speech features . In terms of the description , we defined the resemblance degree of the recognition procedure for this system .

## 2.2 The setting up of reference dictionary

The interval used to measure and analyse the speech signals which vary with time is called a frame , the length of the interval is called the frame length , and the distance between the frames is called the frame interval . Since the speech signals which vary with time can be thought of as constants within 5~20ms , and we measure and analyse the speech signal every 10ms or so , the speech signal's essential feature can be truly reflected . Therefore , the value taken for the frame length is 5~20ms , and for the frame interval is 10ms or so usually .

Suppose , the H features of H humans speaking the same word is $\{x_1, x_2 ....., x_H\}$. In order to resonably classify these features into several classes by using fuzzy classification , we need first to set up the fuzzy resemblance relations among the H features , and we use the following

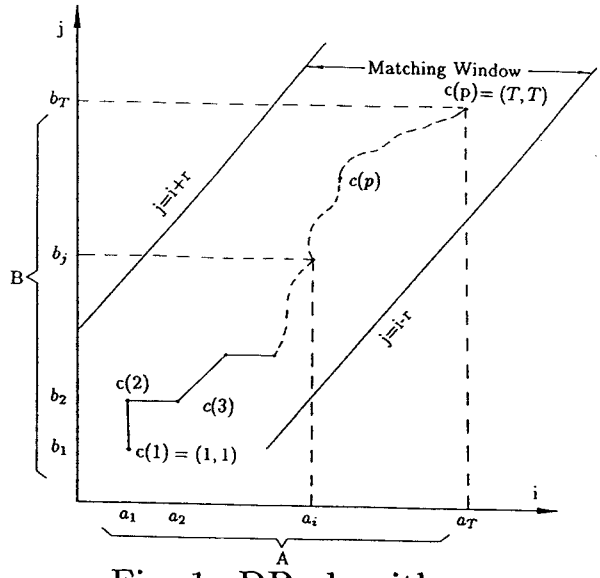

Fig. 1. DP algorithm

formula in this system[1] :

$$r_{i,j} = e^{-D(x_i,y_j)} \tag{1}$$

Where , $D(x_i, x_j)$ is the absolute value distance between $x_i$ and $x_j$ , and this paper , in terms of the speech features , adopts the DP distance by using the DP method after linear matching [3]. Linear matching makes the H different length features monolized , i, e . to make the frame number of each feature equal to the average frame number of the H feature of the word .The method is simple , so we won't mention it here . The DP distance is the distance after eliminating the influences of the variation of each corresponding length within the features . By the following algorithm , we know that the DP distance can reflect precisely and truly the differences among different speaker speech features for the same word .

Suppose A and B are arbitrarily two features in feature $\{x_1, x_2..., x_H\}$ after the linear matching . T is the frame number , then A and B can be expressed as :

$$A = \{a_1, a_2, ..., a_T\}$$

$$B = \{b_1, b_2, ..., b_T\}$$

where , $a_i(1 \leq i \leq T)$ is the Fourier transform coefficient of the ith frame in the feature A , and $b_j$ is the Fourier transform coefficient of the jth frame in the feature B .

The basic idea of the DP method is to find a path F = c(1) c(2).....c(p) from the matching window in Fig(1) , and to have the right side of equation (2) take the minimum value , and this minimum value is the DP distance between feature A and B , and is expressed as D(A , B) .

$$D(A,B) = Q^{-1} \bigwedge_{F} \left[ \sum_{p=1}^{P} d(c(p) \cdot \omega(p)) \right] \tag{2}$$

where ,

$$\omega(p) = (i(p) - i(p-1)) + (j(p) - j(p-1))$$

$$d(c(p)) = d(i,j) = |a_i - b_j|$$

$$Q = 2T$$

In order to find the discrete functional extrema value , we adopt dynamic programming . The method of doing this is as follows :

(1) The initial value setting :

$$g(1,1) = 2d(1,1) = 2d(a_1, b_1), i = 1$$

$$g(1,0) = g(0,1) = 0$$

(2) To take $j_1 = max(1, i-r), j_2 = min(i+r, T)$ to have j increased from $j_1$ to $j_2$ , its increment is 1 , and we use the recursion formula (3) to calculate g(i ,j).

$$g(i,j) = min \begin{bmatrix} g(i,i-1) & +d(i,j) \\ g(i-1,j-1) & +2d(i,j) \\ g(i-1,j) & +d(i,j) \end{bmatrix} \tag{3}$$

(3) if $i \leq T$ , then take i+1 as the new value of i to execute (2) , if i=T , then $D(A,B) = g(T,T)/Q$ and the calculation of (3) is finished .

From formula (2) , we find out the DP distance between $x_i$ and $x_j (1 \leq i, j \leq T)$ , where $x_i$ and $x_j$ are the different features of two human's pronounciation of the same word , then by using formula (1) , we can get the fuzzy resemblance relation between these features . After that , we do the fuzzy classification by using the maximum tree method

[7], and classify reasonably the H feature into a certain number of classes , and find the average value within each class . Finally , express these average values by using fuzzy set , and store them in the computer as a standard pattern in the reference dictionary . Considering the real-time of computer processing , the specific method of fuzzy expression is to suppose the total number of the words in the reference dictionary to be I , and by fuzzy classification , classify the reference patterns of each word into K classes , take $a_{ik}{}^{(t,l)}$ as the arithmetic mean value of the corresponding lth Fourier trans-form coefficient of the tth frame . Where , $1 \leq i \leq I$) , $1 \leq k \leq K$ , $1 \leq t \leq T$ , $1 \leq l \leq L_a$ , $L_a$ is the total number of the Fourier transform coefficients in the tth frame . The corresponding standard pattern in the reference dictionary in the computer is a fuzzy set , and the definition of the membership function is as follows :

$$\mu_{a_{ik}{}^{(t,l)}}(m) = \begin{cases} 1 - (1/5)|m - a_{ik}{}^{(t,l)}| & , \ |m - a_{ik}{}^{(t,l)}| \leq 5 \\ 0 & , \ otherwise \end{cases} \tag{4}$$

If the feature of every word is processed by the above method the reference dictionary of the speaker-independent fuzzy speach recognition system mentioned by this article is formed . So we can see that the reference dictionary for this system is the multi-structure expressed by fuzzy set . i . e . each word corresponds to some standard fuzzy pattern . Therefore , this system has developed the structure of the speaker-independent speech recognition system of the usual multi-structure reference dictionary .

### 2.3 The definition of the resemblance degree .

In what way the inputed speech patterns match with the standard pattern in the reference dictionary , is the problem which is going to be solved in this section . The method adopted by this system is first to do the linear matching to the inputed recognized speech feature , and to have its frame number equal to the frame number of the ith word . The Fourier coefficients in each frame are expressed by using fuzzy set , and , in this way , form the fuzzy pattern X(i) of the waiting to be recognized speech which matches the standard pattern of the ith word ,and then , use formula (5) to match it to the standard pattern of the ith word in the reference dictionary .

Suppose , the lth Fourier coefficient in the tth frame of the feature of the waiting to be recognized speech is $a^{(t,l)}$ . The fuzzy expression calculated by formula (4) is $\mu_{a(t,l)}(m)$ then the resemblance degree of $\mu_a{}^{(t,l)}(m)$ and the standard fuzzy pattern of the Ith word in the reference dictionary $S_{x(i),i}$ is defined as follows:

$$S_{x(i),i} = \bigvee_{k=1,2..:\mathrm{k}} \left[ 1/(L \cdot T) \sum_{t=1}^{T} \sum_{l=1}^{l} P_k(t,l) \right] \tag{5}$$

where ,

$$P_k(t,l) = \bigvee_{m} \left[ \mu_{a(t,l)}(m) \bigwedge \mu_{a_{i,k}{}^{(t,l)}}(m) \right]$$

$$L = L_a \bigwedge L_{x(i)}$$

After we have calculated the corresponding resemblance degree of each word in the reference dictionary by using formula (5) , if there is a $I_0$th word in I words , and it satisfies the following formula(6):

$$S_{x(I_0),I_0} = max[S_{x(1),1}, S_{x(2),2} \cdots, S_{x(I),I}] \tag{6}$$

then we think that the $I_0$th word and the waiting to be recognized patterns are the most similar , i . e . we think that the waiting to be recognized pattern and the $I_0$th word are the same , therefore , we get the $I_0$ as the recognized output results .

## 3. System constitution

This paper puts forward the system constitution of the speaker inpdependent fuzzy speech recognition method as shown in Fig(2) , where the broken line expresses the setting up procedure of the reference dictionary . The pre-processing part is implemented by a special purpose hardware processing unit . Its function is , after smoothly filtering the electronic signals of the speech , to transfer the analogue speech signals into the numerical speech signals (12 bit ) under the 10 kHz physical sampling frequency , ready for the computer's read-save . The maximum tree principle is adopted in fuzzy classification [1] , and we use the Kurskal

principle to find the maximum tree by using Fuzzy resemblance matrix . In the flow chart , the procedure from the speech feature extraction to the end is implemented on an FM-7 computer by using software . The Hamming window method is used in frame classification : the frame length is 12.8ms , and the period of frame is 10ms .

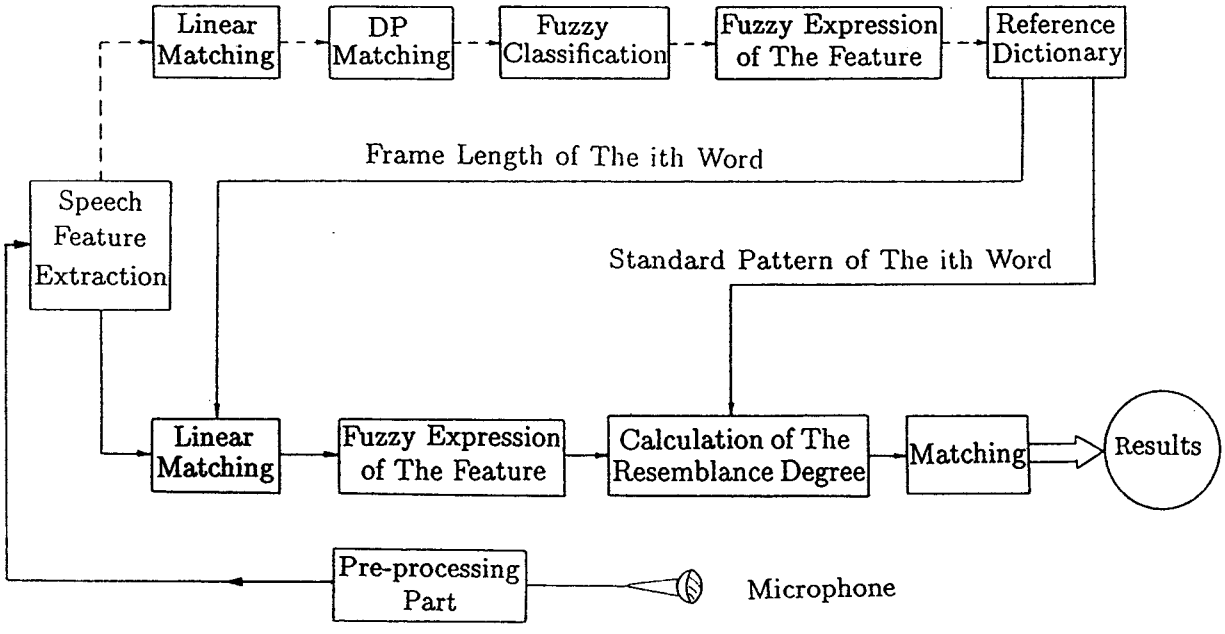


Fig. 2. System constitution of the speaker-inpdependent fuzzy speech recognition

## 4. The result of recognition

While setting up the reference dictionary , each word altogether 12 humans speech features including males and females , and registers 50 geographical terms . The speech environment is a quiet room without noise . The specific procedure is that the sound is transfered to the hardware interface board through a microphone , then , the computer stores the numerical speech signals , as the source voice data , on disk , ready for forming the reference dictionary . Fuzzy classification classifies the speech features of 12 humans speaking each word into three classes , i . e . the standard fuzzy pattern of each word in the reference dictionary is three . The fuzzy patterns expressed by fuzzy set in the reference dictionary are also stored on the disk ready for real-time invocation when needed for recognizing .

The recognition experiment is done by inviting 3 men and 3 women individually to do the practical analogue experiment . From the experiments , we find that the recognition rate is different for different people .The average recognition rate for females is a bit higher than that for males .The average recognition rate for this experiment is 98.3% . One other point which should be emphasized is that the six speakers all speak standard chinese .

## 5. Conclusions

This paper applies fuzzy theory into speech recognition technology , and puts forward a new method of speaker-independent fuzzy speech recognition .According to the experiments , we know that , because the fuzzy classification is used in setting up the reference dictionary , and the fuzzy set is adopted in speech features' description , the universality of the reference dictionary is improved . The recognition procedure is done by using the close degree between the fuzzy sets to recognize the results , therefore , this system gets a higher recognized rate . This paper is a very significant effort in the field of speech recognition .

The problems which still need to be solved , in order to make the whole system mone practical , are to realize the hardware description of the part algorithms and the modularization of the processing unit .

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