HUMAN AND MACHINE RECOGNITION OF SPEECH SOUNDS IN NOISE

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ABSTRACT
In this paper, we report on studies of human perception and machine recognition of speech signals in noise. As a case study, the perception of the linguistic feature 'place of articulation' for consonants in adverse conditions is examined through a series of perceptual experiments. The experiments examined the effects of additive noise on the perception of Consonant-Vowel (CV) syllables. Results are analyzed in terms of vowel context and SNR. Results show a striking vowel-context effect. In general, the /Ca/ context is more noise robust than other contexts. Another important result is that certain categories which cue a feature acoustically and perceptually in quiet conditions (such as nasal murmurs for /n, m/) do not correlate well with the perceptual robustness of the feature in noise. A Hidden Markov Model (HMM)-based automatic speech recognition (ASR) system was then constructed to identify the features at various signal-to-noise ratios. Modifications to a standard ASR system were made that were inspired by the results of the perceptual experiments. The modifications improved recognition performance by 20-40 percent in noise.

Keywords: noise-robust speech recognition; human perception in noise; vowel-context effects.

1. INTRODUCTION
Although noise is frequently the limiting factor in communication, most previous studies that examined the perceptual importance of acoustic cues in signaling phonetic contrasts have been based on experiments conducted in quiet. This study focuses on the perception of the place of articulation for syllable-initial nasal consonants /m, n/ in adverse conditions. These sounds are typically characterized by an initial segment (a murmur), which has most of its energy in the low-frequency region, and by distinct formant transitions into the neighboring vowel. The place of maximum constriction is at the lips for /m/, whereas it is at the alveolar ridge for /n/. Hence, the spectral characteristics of these two sounds, in the murmur region and formant transitions, are different. In quiet, nasal place of articulation is thought to be signaled by both the murmur and formant transitions into the adjacent vowel [3, 4, 7]. The only study that examined place perception in noise is [6] in which the perception of place, manner, and voicing of syllable-initial consonants (including the nasals) were examined. Unfortunately, the study was limited to the vowel /a/.

We examine the perceptual role of both acoustic features (murmur and formant transitions) in identifying nasal place through an extensive series of perceptual experiments. The experiments examine the effects of additive white Gaussian noise, and additive speech-shaped noise, on place perception. The experiments also examine human perception of altered /CV/ syllables, whereby the murmur or the formant transitions are removed, in the presence of AWGN. An automatic speech recognition (ASR) system was then constructed to take into account the results of the perceptual experiments. System performance was compared to a baseline ASR system.

2. PERCEPTUAL EXPERIMENTS

2.1 Stimuli and Protocol
Stimuli consisted of CV syllables where the consonant was either /m/ or /n/, and the vowel was /a/, /i/, or /u/. Eight tokens of each syllable were recorded by two male and two female talkers of American English, resulting in a total of 192 syllables. The sampling rate was 16 kHz and the speech was coded with 16 bits. Perceptual experiments were a combination of identification and adaptive forced choice tasks, and were conducted in a sound-isolated chamber. Four healthy-hearing subjects participated in the experiments.

2.2 Additive Noise Experiments
In these experiments, white Gaussian noise (WGN) or speech shaped noise (SSN), modeled after the specifications of [1], was added, digitally, to the speech stimuli. The level of the noise varied in 5 dB steps. The noisy signals were 150 msec longer than the speech tokens. The speech tokens were placed 150 msec after the onset of the noise so that artifacts caused by the sudden onset of noise are avoided. The Signal-to-Noise Ratio
20 msec long for the other syllables. Shorter signals are more difficult to hear especially in the presence of noise [2]. Spectrograms of the syllables /na/ and /ni/ as spoken by a male talker are shown in Fig. 3. Notice the longer formant (especially F2) transition in /na/. In addition, F2, which carries important place information has the highest amplitude (relative to F1) in /Ca/ syllables, and the least in /Ci/ syllables.

Figure 1: Average percent correct identification for nasal place in the presence of additive white Gaussian noise.

(SNR) was calculated based on the average energy of the speech signal and the calculation precluded silence in the speech segments, if any.

2.2.1 Results. Figures 1 and 2 summarize the results of the AWGN and SSN experiments, respectively. Notice the strong vowel-context effect, with /Ca/ being the most robust in the presence of noise and /Ci/ being the least robust. In the AWGN case, and at -10 dB SNR, percent correct recognition is above 80 for /Ca/ syllables. For /Ci/ syllables, on the other hand, place perception is difficult even at a high SNR of 5 dB. We speculate that /Ca/ is the most robust in noise because formant transitions are longer than they are in the other syllables. In our database, /na/ transitions were about 50-60 msec long, while they were about 15-

Figure 2: Average percent correct identification for nasal place in the presence of additive speech shaped noise.

The type of noise also affects perception. For example, a comparison of Figures 1 and 2 reveals that, at the same SNR, nasals are more difficult to perceive correctly in the presence of WGN than in the presence of SSN. This could be explained by the fact that speech-shaped noise is low-pass and as such, high-frequency
spectral cues can contribute to place perception if these
cues are not masked by noise. This is especially true
for /Ci/ syllables since F2 in this case is high (above
2000 Hz). The results clearly imply that the study of
Miller and Nicely [6] does not generalize to all vowel
contexts and to different noise shapes.

2.3 Examining the Role of the Murmur and Formant Transitions in Noise
To better quantify the role of the murmur and formant
transitions on nasal perception, the following experi-
ment was undertaken. Subjects were asked to identify
nasal consonants in three different types of speech
tokens: (1) CV syllables; (2) CV syllables without the
murmur; and (3) CV syllables with 150 msec of the
formant transition in the following vowels removed. The
speech tokens were then added to WGN and presented
to listeners. An adaptive procedure based on the trans-
fomed up-down method by [5] was implemented. A
correct response results in a reduction in threshold and
an incorrect response results in a threshold increase.
The convergence of the threshold occurs when there
are 75% correct responses.

2.3.1 Results. Table 1 illustrates experimental re-
sults. A threshold increase implies that the sound can
be identified reliably only if the the additive noise is
lower than it was for the baseline case. For /Ca/ and
/Cu/, removing the nasal murmur raises the threshold
by about 2-3 dB, while removing the transition results
in raising the threshold by about 24 dB for /Ca/ and
12 dB for /u/. Thresholds could not be found (proce-
dure did not converge) for /Ci/ syllables when either
the murmur or the formant transition was removed.
These results clearly indicate that, in the presence
of AWGN, formant transitions seem to play a critical role
in identifying place for /Ca/ and /Cu/ syllables. In
/Ci/ syllables, since the formant transitions are short
and the amplitudes of F2 are relatively weak, the exis-
tence of both the murmur and the formant transitions
is important for identifying place.

<table>
<thead>
<tr>
<th>vowel</th>
<th>CV</th>
<th>w/o murmur</th>
<th>w/o transition</th>
</tr>
</thead>
<tbody>
<tr>
<td>/a/</td>
<td>-12.5</td>
<td>-10.6</td>
<td>12</td>
</tr>
<tr>
<td>/u/</td>
<td>-7.5</td>
<td>-4.8</td>
<td>5.4</td>
</tr>
<tr>
<td>/i/</td>
<td>8.9</td>
<td>N/A</td>
<td>N/A</td>
</tr>
</tbody>
</table>

Table 1: The 75% correct threshold in dB for identifying
nasal place in AWGN.

3. VARIABLE FRAME RATE (VFR) METHOD

3.1 The Algorithm As implied in the perceptual
experiments, changes in spectral characteristics are im-
portant cues for discriminating and identifying speech
sounds. These changes can occur over very short time
intervals. Computing frames every 10 ms, as commonly
done in recognition systems, is not sufficient to capture
such dynamic changes. To illustrate this point, Figure
4 shows plots of MFCC vectors along a 100 ms segment
surrounding the formant transition region in a /ma/
syllable. The frame length is 20 ms, but the frame step
size is 10 ms in (a) and 2.5 ms in (b). Note that the
murmur and steady-state region of the vowel are re-
presented by (perhaps an unnecessarily large) number
of MFCC vectors, while the critical formant transition
region (13 ms) is only represented by one vector with
a 10 ms frame step size and 2 (distinct) vectors when
the step size is reduced to 2.5 ms.

We propose a Variable Frame Rate (VFR) algo-
rithm [9]. The algorithm results in an increased number
of frames for rapidly-changing segments with relatively high energy and less frames for steady-state segments. Using MFCC feature vectors, the variable frame rate algorithm is implemented as shown in Figure 5. First, speech is analyzed with frame lengths of 25 ms (Hamming window) and a step size of 2.5 ms. We refer to these frames as the dense frames. Second, the difference \( d(i) \), where \( i \) is the time index, between every two adjacent dense frames is calculated. The average of these differences is then calculated over the whole utterance. Third, based on the weighted differences, some frames are kept and others are discarded. In particular, dense frames around a formant transition will be kept, while at the steady part of the signal, frames will be picked sparsely. It is important to note that the distance \( d(i) \) is calculated as the energy weighted Euclidean MFCC distance; first the Euclidean distance of the MFCC vectors of two adjacent frames are calculated, then it is weighted by \( (E-b) \), where \( E \) is the log energy of that frame, and \( b \) is a constant offset. This is different from the method proposed in where the Euclidean MFCC distance was used. Energy weighting is important so that segments which exhibit changes but are low in energy are discarded, since they may not be noise robust. Our previous experiments have shown a clear relationship between the energy of formant transitions and perceptual noise robustness. In addition, our pilot ASR experiments using Euclidean MFCC distance did not yield high recognition accuracy in noise. The two parameters \( a \), the threshold, and \( b \), log energy offset, are chosen experimentally. The choice of \( a \) will determine the average data rate. For example, if \( a \) is 4 (ratio of the 10 ms step size and the dense step size of 2.5 ms), then the resulting total number of frames will be nearly the same as that in a front-end with a frame step size of 10 ms. If \( a \) is larger than 4, then the average data rate will be less than 100 frames per second and vice versa. In our implementation, \( a \) was chosen to be 6.8. The log energy offset \( b \) was set to be the average \( E \) (over the entire utterance) divided by 1.5.

3.2 An Example of VFR Analysis Figure 6 illustrates how frames are picked for the utterance /ma/ as spoken by a male speaker. Part (a) shows a time waveform of the utterance. The upper part of (b) plots \( d(i) \), the weighted feature distance between two adjacent frames - with a step size of 2.5 ms - and the lower part shows the result of the frame-picking algorithm where each bar indicates that a frame has been chosen for recognition. Note that near the transition region from the consonant to the vowel \( d(i) \) is large. For this example, 50 out of 200 dense frames are picked. Around the transition region, all the dense frames (spaced by 2.5 ms) are kept while in the steady-state part of the vowel, only 3-4 frames out of 20 frames are selected corresponding to a step size which is larger than 10 ms.

Figure 4: MFCC vectors around the transition of /ma/.
(a) Window step = 10 ms. (b) Window step = 2.5 ms.

- Calculate MFCC vectors with a 25 ms frame length and a 2.5 ms step size.
- Calculate \( d(i) \), weighted Euclidean distance of MFCC vectors, between frame \( i \) and frame \( i+1 \).
- Calculate the average distance \( d \) from \( d(i) \), this can be done locally or, in our case, to the whole utterance.
- Calculate the threshold \( \Theta \) for "frame picking" as \( \Theta = \alpha d \), where \( \alpha \) is a parameter that determines the average frame rate.
- Perform "frame picking". Locally accumulate \( d(i) \) from time \( n \) \( (A=A+d(i), i=n, n+1, n+2, \ldots, n+k, \ldots) \). Whenever \( A \) exceeds \( \Theta \) at time \( n+k \) choose the frame at \( n+k \), reset and restart the accumulation from \( n+k \).

Start this process from \( n=1 \), and repeat to the last frame of the utterance, the resulting frames are the variable frame rate MFCC vectors (VRMFCC).

Figure 5: Flow chart of computing variable frame rate MFCC vectors.
Figure 6: (a) The wave form of /ma/. (b) The "frame picking" results. The upper panel is the normalized d(i) the lower panel is the frames that are picked out from the dense frame.

3.3 Recognition with the VFR Front End

The variable frame rate method was used in ASR experiments using the nasal database described in Section 2, and the TIDIGITS database. In the experiments, the performance of the recognition system with two feature vectors were compared: MFCC, and MFCC vectors with peak enhancement [8] (hereafter referred to as MFCCP). First and second derivatives of these features were used. Training was done using clean data while testing was done with either clean or noisy data. Results for the nasal recognition experiment are shown in Table 2. Clearly, the variable frame rate approach together with a method for enhancing spectral peaks, gives the best performance at low SNRs. The VFR method was also used with the database Studio Quality Speaker Independent Connected Digit Corpus (TIDIGITS). Each left to right digit HMM model had 4 states, 2 mixtures, and a diagonal covariance matrix. 80 utterances from 80 speakers, (40 male and 40 female) were used to train each model. Test data were from the other 32 speakers (half male and half female).

We compared MFCC and MFCCP with their variable frame rate versions. The results are shown are summarized in Table 3. The results clearly illustrate that applying the VFR method to each feature vector improves recognition performance especially at low SNRs. Increasing time resolution for rapidly changing segments, while keeping the time resolution low for steady parts, results in improved robustness.

<table>
<thead>
<tr>
<th></th>
<th>Clean</th>
<th>15 dB</th>
<th>5 dB</th>
<th>0 dB</th>
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<tr>
<td>MFCC</td>
<td>90</td>
<td>89</td>
<td>88</td>
<td>87</td>
</tr>
<tr>
<td>MFCCP</td>
<td>96</td>
<td>91</td>
<td>77</td>
<td>78</td>
</tr>
<tr>
<td>VFRMFCCP</td>
<td>100</td>
<td>96</td>
<td>81</td>
<td>71</td>
</tr>
</tbody>
</table>

Table 2: Recognition accuracy for the nasal database for different front ends.

<table>
<thead>
<tr>
<th>SNR</th>
<th>20 dB</th>
<th>15 dB</th>
<th>5 dB</th>
<th>0 dB</th>
</tr>
</thead>
<tbody>
<tr>
<td>MFCC</td>
<td>97</td>
<td>87</td>
<td>71</td>
<td>56</td>
</tr>
<tr>
<td>MFCCP</td>
<td>98</td>
<td>96</td>
<td>93</td>
<td>78</td>
</tr>
<tr>
<td>VFRMFCCP</td>
<td>97</td>
<td>97</td>
<td>96</td>
<td>89</td>
</tr>
</tbody>
</table>

Table 3: Recognition accuracy for MFCC, MFCCP, and VFRMFCCP front ends using the TIDIGITS database.

SUMMARY AND CONCLUSION

In this paper, we investigated the perception of place of articulation for the nasal consonants /m, n/ in adverse conditions which included AWGN, and additive speech-shaped noise. In addition, identification thresholds for the two consonants in AWGN were measured using an adaptive procedure. These thresholds were measured for the entire syllable, and for the syllable with either the murmum or formant transitions removed. Results show a strong vowel-context effect. For example, for /Ca/ and /Cu/ syllables, the formant transitions seem to play a bigger role in place identification (in the presence of additive noise) than the murmum. For /Ci/, both the murmum and the formant transitions appear to play an important role in identifying place. To investigate whether or not placing a larger emphasis on formant transitions would improve machine recognition performance, a recognition system was constructed which was sensitive to dynamic changes of the signal over short periods of time. The system had better performance than a baseline ASR system especially at low SNRs.

ACKNOWLEDGMENTS

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REFERENCES