

Characterizing and Processing Robot-Directed Speech

Paulina Varchavskaia, Paul Fitzpatrick, Cynthia Breazeal

AI Lab, MIT, Cambridge, USA
paulina,paulfitz,cynthia@ai.mit.edu,

Abstract. Humanoid robots are more suited to being generalists rather than specialists. Hence when designing a speech interface, we need to retain that generality. But speech recognition is most successful in strongly circumscribed domains. We examine whether some useful properties of infant-directed speech can be evoked by a robot, and how the robot’s vocabulary can be adapted. (need 200 words)

Keywords: *speech recognition, ASR, word-spotting*

1 Introduction

A natural-language interface is a desirable component of a humanoid robot. In the ideal, it allows for natural hands-free communication with the robot without necessitating any special skills on the human user’s part. In practice, we must trade off flexibility of the interface with its robustness. Contemporary speech understanding systems rely on strong domain constraints to achieve high recognition accuracy [23]. This paper makes an initial exploration of how ASR techniques may be applied to the domain of robot-directed speech with flexibility that matches the expectations raised by the robot’s humanoid form.

A crucial factor for the suitability of current speech recognition technology to a domain is the expected perplexity of sentences drawn from that domain. Perplexity is a measure of the average branching factor within the space of possible word sequences, and so generally grows with the size of the vocabulary. For example, the basic vocabulary used for most weather-related queries may be quite small, whereas for dictation it may be much larger and with a much less constrained grammar. In the first case speech recognition can be applied successfully for a large user population across noisy telephone lines [22], whereas in the second a good quality headset and extensive user training are required in practice. It is important to determine where robot-directed speech lies in this spectrum. This will depend on the nature of the task to which the robot is being applied, and the character of the robot itself. For this paper, we will consider the case of Kismet [5], an “infant-like” robot whose form and behavior is designed to elicit nurturing responses from humans. Among other effects, the youthful character of the robot is expected to confine discourse to the here-and-now.

Sections 4 and 5 look at these effects in more detail. Sections 6 and 7 look at methods that don’t rely on such cooperative forms of speech. We expect both mechanisms to play a role in practical language modeling for a general-purpose humanoid robot.

2 Background

Recent developments in speech research on robots have followed two basic approaches. The first approach builds on techniques developed for command-and-control style interfaces. These systems employ the standard strategy found in ASR research of limiting the recognizable vocabulary to a particular predetermined domain or task, so as to ensure a manageable size of the vocabulary. For instance, the ROBITA robot [16] interprets command utterances and queries related to its function and creators, using a fixed vocabulary of 1,000 words. Within a fixed domain fast performance with few errors becomes possible, at the expense of any ability to interpret out-of-domain utterances. But in many cases this is perfectly acceptable, since there is no sensible response available for such utterances even if they were modeled.

A second approach adopted by some roboticists [19, 17] is to allow adjustable (mainly growing) vocabularies. This introduces a great deal of complexity, but has the potential to lead to more open, general-purpose systems. Vocabulary extension is achieved through a label acquisition mechanism based on a learning algorithm, which may be supervised or unsupervised. This approach was taken in particular in the development of CELL [19], Cross-channel Early Language Learning, where a robotic platform called Toco the Toucan is developed and a model of early human language acquisition is implemented on it. CELL is embodied in an active vision camera placed on a four degree of freedom motorized arm and augmented with expressive features to make it appear like a parrot. The system acquires lexical units from the following scenario: a human teacher places an object in front of the robot and describes it. The visual system extracts color and shape properties of the object, and CELL learns on-line a lexicon of color and shape terms grounded in the representations of objects. The terms learned need not be pertaining to color or shape exclusively - CELL has the potential to learn any words, the problem being that of deciding which lexical items to associate with which seman-

tic categories. In CELL, associations between linguistic and contextual channels are chosen on the basis of maximum mutual information. Also in [17], a Pioneer-1 mobile robot was programmed with a system to cluster its sensory experiences using an unsupervised learning algorithm. In this way the robot extends its vocabulary by associating sets of sensory features with the spoken labels that are most frequently uttered in their presence.

3 Our approach

We share the goal of automatically acquiring new vocabulary, but our methods are different. We rely on contemporary ASR systems.

but wish to do so using conventional speech recognition systems.

In our approach, we try to stay within the ASR paradigm as much as possible

The rest of this paper examines two broad classes of robot-directed speech which are relevant to vocabulary extension. In the first case, we consider speech that is cooperative – speech that intentionally or not has properties that allow us to cast the machine learning problem as supervised. This is examined in Section 4. In the second case, we consider neutral speech for which this is not the case, and the problem is essentially unsupervised. This is examined in Section 6.

4 Supervised extension

As mentioned earlier, natural speech may be the most convenient medium for a human to interact with a humanoid robot, including in the case of communicating vocabulary extensions. The latter case is a specialized kind of interaction, where the human plays the role of a teacher, with the accompanying modifications in the discourse with which the robot is addressed. When interacting with a young-appearing robot such as Kismet in particular, we can expect that the speech input may have specialized characteristics similar to those of infant-directed speech (IDS). This section examines some of the properties of IDS so they may inform our expectations of the nature of Kismet-directed speech signal.

4.1 Properties of infant-directed speech

In this paper we examined the following two questions regarding the nature of infant-directed speech:

- Does it include a substantial proportion of single-word utterances? Presenting words in isolation would solve the difficult word-segmentation problem.
- How often, if at all, is it clearly enunciated and slowed down in an unnatural way? Overarticulated speech may be helpful to infants, but may be detrimental to artificial speech recognizers.

Isolated words Whether isolated words in parental speech help infants learn has been a matter of some debate. It has been shown that infant-directed utterances are usually short with longer pauses between words (e.g., research cited in [21]), but also that they do not necessarily contain a significant proportion of isolated words [1]. Another study [6] presents evidence that isolated words are a reliable feature of infant-directed speech, and that infants’ early word acquisition may be facilitated by their presence. In particular, the authors find that the frequency of exposure to a word in isolation is a better predictor of whether the word will be learned, than the total frequency of exposure. This suggests that isolated words may be easier for infants to process and learn. Equally importantly for us, however, is the evidence for a substantial presence of isolated words in IDS: 9% found in [6] and 20% reported in [1]. If Kismet achieves its purpose of eliciting nurturing behavior from humans, then maybe we can expect a similar proportion of Kismet-directed speech to consist of single-word utterances. This hypothesis will undergo a preliminary evaluation here.

Enunciated speech and “vocal shaping” The tendency of humans to slow down and overarticulate their utterances when they meet with misunderstanding has been reported as a problem in the ASR community [12]. Such enunciated speech degrades considerably the performance of speech recognition systems which were trained on natural speech only. If we find that human caretakers tend to address Kismet with overarticulated speech, its presence becomes a problem to be addressed by the robot’s perceptual system.

In infant-directed speech, we might expect overarticulation to occur in an instructional context, when a caretaker deliberately introduces the infant to a new word or corrects a mispronunciation. Another possible strategy is that of “shaping” of the infant’s pronunciation by selecting and repeating the mispronounced part of the word until a satisfactory result is reached. There is anecdotal evidence that parents may employ such a strategy.

4.2 Preliminary exploration

To facilitate some preliminary exploration of this area, experiments were conducted in which subjects were instructed to try to teach a robot words. While the response of the robot was not the focus of these experiments, a very basic vocabulary extension was constructed to encourage users to persist in their efforts. The system consisted of a simple command-and-control style grammar. Sentences that began with phrases such as “say”, “can you say”, “try” etc. were treated to be requests for the robot to repeat the phonetic sequence that followed them. If, after the robot repeated a sequence, a positive

phrase such as “yes” or “good robot” were used, the sequence would be entered in the vocabulary. If instead the human’s next utterance was similar enough to the first, it was assumed to be a correction and the robot would repeat it. Because of the relatively low accuracy of phoneme-level recognition, such corrections are the rule rather than the exception.

Maybe also: Kismet’s “production” system (so sophisticated).

5 “Supervised” results - preliminary analysis of the input signal

The purpose of this preliminary study is to suggest new ways of improving the speech interface on the robot based on a better knowledge of the properties of speech directed at this particular robot. This section presents the results of the study and a discussion of the method used with directions for similar future research.

We have analyzed video recordings of 13 children aged from 5 to 10(?) years old interacting with the robot. Each session lasted approximately 20 minutes. In two of the sessions, two children are playing with the robot at the same time. In the rest of the sessions, only one child is present with the robot.

The recordings were originally made for Sherry Turkle’s research on children’s perception of technology and identity.

5.1 Preliminary data analysis

We have looked to establish, in particular, whether any of the following strategies are present in Kismet-directed speech:

- single-word utterances (words spoken in isolation)
- enunciated speech
- vocal shaping
- vocal mimicry of Kismet’s babble

A total of 831 utterances were transcribed from the 13 sessions of children playing with the robot. Of these, 303 utterances, or 36.5% consisted of a single word said in isolation. 27.4% of transcribed utterances (228) contained enunciated speech. An utterance was counted as “enunciated speech” whenever deliberate pauses between words or syllables within a word, and vowel lengthening were used. The count therefore includes the very frequent examples where a subject would ask the robot to repeat a word, e.g. “Kismet, can you say: GREEN?”. In such examples, GREEN would be the only enunciated part of the utterance but the whole question was counted as containing enunciated speech. In the whole body of data we have discovered only 6 plausible instances (0.7%) of vocal shaping. Finally, there were 23 cases of children imitating the babbling sounds that Kismet made, which accounts for 2.8% of the transcribed utterances.

These coarse figures mask the finding that there was a very wide distribution of strategies among different subjects. In the following, deviations from the mean are mentioned to give an idea of the wide range of the data. They are not meaningful otherwise, since we have not observed any Gaussian distributions. The total number of utterances varied from session to session in the range between 19 and 169, with a mean of 64 (standard deviation of 44, based on a sample of 13) utterances per session. The percentage of single-word utterances had a distribution among subjects with a mean at 34.8 and a standard deviation of 21.1. These numbers come from counts of single-word utterances including instances of greetings, such as “Hello!”, and attention-bidding using the name of the robot, i.e. “Kismet!”. The statistics change if we exclude such instances from the counts, as can be seen in Table 1.

Specifically, if we exclude both greetings and the robot’s name from counts of single-word utterances, we get a distribution centered around 20.3% with a standard deviation of 18.5%. This still accounts for a substantial proportion of all recorded Kismet-directed speech. Examining the other distributions, we find that the mean proportion of enunciated speech is 25.6% with a deviation of 20.4%. The percentage of vocal imitation of Kismet’s babble has a mean of 2.0% and a deviation of 4.0%, which is again a very large variation. The same pattern holds for the proportion of vocal shaping utterances: a mean of 0.6% with a standard deviation of 1.1%. Thus, children in this dataset used varied strategies to communicate with the robot, and there does not seem to be enough evidence to suggest that the strategies of vocal shaping and imitation play an important part in it.

5.2 Discussion

The results presented above seem encouraging. However, before we draw any meaningful conclusions from the analysis, we must realize that in this instance, the process of gathering the data and the method of analysis had several shortcomings. The data itself, as was mentioned earlier, came from recordings of interactions set up for the purposes of an unrelated sociological study of children. (AM I SUPPOSED TO CITE SHERRY? BUT HOW?).

The interaction sessions were not set up as controlled experiments, and do not necessarily represent spontaneous Kismet-directed speech. In particular, on all occasions but one, at some point during the interaction, children were instructed to make use of the currently implemented command-and-control system to get the robot to repeat words after them. In some cases, once that happened, the subject was so concerned with getting the robot to repeat a word that anything else simply disappeared from the interaction. On three occasions, the subjects were instructed to use the “say” keyword

subject	# utterances	# single-word utterances	%	# single-word greetings	# kismet utterances	%
1	94	65	69.2	0	30	37.2
2	19	9	47.4	1	2	31.6
3	128	69	54.0	11	46	9.3
4	37	17	46.0	2	7	21.6
5	26	9	34.7	3	0	23.1
6	61	14	23.0	9	0	8.2
7	34	2	5.9	1	0	2.9
8	73	43	58.9	0	0	58.9
9	169	39	23.1	8	9	13.0
10	32	17	53.1	0	2	46.9
11	56	7	12.5	3	1	5.4
12	33	5	15.2	5	0	0.0
13	69	7	10.1	3	0	5.8
total	831	303		46	97	
mean			34.8			20.3
deviation			21.1			18.5

Table 1. Percentage of isolated words in Kismet-directed speech

as soon as they sat in front of the robot. When subjects are so clearly focused on a teaching scenario, we can expect the proportion of isolated words, for instance, to be unnaturally high.

Note also that as of now, we have no measure of accuracy of the transcriptions, which were done by hand by one transcriber, from audio that sometimes had poor quality. Given the focus of the analysis, only Kismet-directed speech was noted from each interaction, excluding any conversations that the child may have had with other humans who were present during the session. Deciding which utterances to transcribe was clearly another judgement call that we cannot validate here yet. Finally, since the speech was transcribed by hand, we cannot claim a scientific definition of an utterance (e.g., by pause duration) but must rely on one person’s judgement call again.

However, this preliminary analysis shows promise in that we have found many instances of isolated words in Kismet-directed speech, suggesting that Kismet’s environment may indeed be scaffolded for word learning. We have also found that a substantial proportion of speech was enunciated. This would present problems for the speech recognizer, but at the same time opens new possibilities. For an improved word-learning interface, it may be possible to discriminate between natural and enunciated speech to detect instances of pronunciation teaching (this approach was taken in the ASR community, for example in [12]). On the other hand, the strategy of vocal shaping was not clearly present in the interactions, and there were few cases of mimicry. More (and better) research would determine how reliable or not these features of Kismet-directed speech may be.

We plan in the future to conduct much more controlled studies to explore further the nature of the speech input to the robot. The setup will involve filming 20 minutes of interaction between an adult subject and Kismet.

The subjects will be told that they are controls in an experiment with children and that they should play freely with the robot while we record the interaction. All care will be taken not to constrain the subjects’ speech patterns artificially by asking them to teach Kismet words. The data will be sequenced and transcribed independently by two people, so the transcripts may be compared and analyzed for agreement and error. An utterance will be defined as continuous speech between two pauses which last for longer than a threshold. We will be looking to address in depth the question of how word teaching scenarios are different from other kinds of interaction with the robot, by examining the prosody, vowel and pause lengthening (enunciation) and repetitions in the speech input. We will also be interested in finding out whether Kismet-directed speech has a high proportion of topics related to the robot’s immediate environment, for the purposes of attaching meaning to the words that the robot learns.

6 Unsupervised vocabulary extension

This section develops a technique to bootstrap from an initial vocabulary (perhaps introduced by the methods described in Section 4) by building a model of unrecognized parts of utterances. The purpose of this model is both to improve recognition accuracy on the initial vocabulary and to automatically identify candidates for vocabulary extension. This work draws on research in word spotting and speech recognition. In word spotting, utterances are modeled as a relatively small number of keywords floating on a sea of unknown words. In speech recognition, an occasional unknown word may punctuate utterances that are otherwise assumed to be completely in-vocabulary. Despite this difference in viewpoint, in some circumstances implementations of

the two may become very similar. When transcribed utterances are available for a domain, word spotting benefits from the more detailed background model this can support [13]. The manner in which the background is modeled in these cases is reminiscent of speech recognition. For example, a large vocabulary with good coverage may be extracted from the corpus, so that relatively few words in an utterance remain unmodeled. In this case, the situation is qualitatively similar to OOV (out-of-vocabulary) modeling in a conventional speech recognizer, except that the vocabulary is strictly divided into “filler” and “keyword”.

We will bootstrap from a relatively weak background model for word-spotting, where OOV words dominate, to a much stronger model where many more word or phrase clusters have been “moved to the foreground” and explicitly modeled. With this increase in vocabulary comes an increase in the potency of language modeling, boosting performance on the original vocabulary.

The remainder of this section shows how a conventional speech recognizer can be convinced to cluster frequently occurring acoustic patterns, without requiring the existence of transcribed data.

A speech recognizer with a phone-based OOV model is able to recover an approximate phonetic representation for words or word sequences that are not in its vocabulary. If commonly occurring phone sequences can be located, then adding them to the vocabulary will allow the language model to capture their co-occurrence with words in the original vocabulary, potentially boosting recognition performance. This suggests building a “clustering engine” that scans the output of the speech recognizer, correlates OOV phonetic sequences across all the utterances, and updates the vocabulary with any frequent, robust phone sequences it finds. While this is feasible, the kind of judgments the clustering engine needs to make about acoustic similarity and alignment are exactly those at which the speech recognizer is most adept. This section describes a way to convince the speech recognizer to perform clustering almost for free, eliminating the need for an external module to make acoustic judgments.

The clustering procedure is shown in Figure 1. An *n*gram-based language model is initialized randomly, or trained up using whatever data is available - for example, a small collection of transcribed utterances. Unrecognized words are explicitly represented using a phone-based OOV model, described in the next section. The recognizer is then run on a large set of untranscribed data. The phonetic and word level outputs of the recognizer are compared so that occurrences of OOV words are assigned a phonetic transcription. A randomly cropped subset of these are tentatively entered into the vocabulary, without any attempt yet to evaluate their significance (e.g. whether they occur frequently, whether they are dangerously similar to a keyword, etc.). The hy-

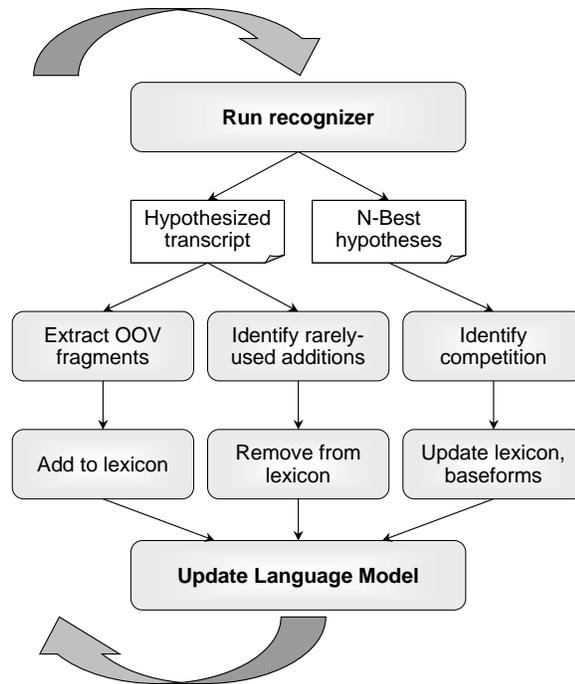


Fig. 1. The iterative clustering procedure.

potheses made by the recognizer are used to retrain the language model, making sure to give the newly added vocabulary items some probability in the model. Then the recognizer runs using the new language model and the process iterates. The recognizer’s output can be used to evaluate the worth of the new “vocabulary” entries. The following sections detail how to eliminate vocabulary items the recognizer finds little use for, and how to detect and resolve competition between similar items.

Extracting OOV phone sequences Recognizer is that developed by the SLS group at MIT [8]. The recognizer used the OOV model developed by Bazzi in [3]. This model can match an arbitrary sequence of phones, and has a phone bigram to capture phonotactic constraints. The OOV model is placed in parallel with the models for the words in the vocabulary. A cost parameter can control how much the OOV model is used at the expense of the in-vocabulary models. This value was fixed at zero throughout the experiments described in this paper, since it was more convenient to control usage at the level of the language model. The bigram used in this project is exactly the one used in [3], with no training for the particular domain.

Recovering phonemic representations It is useful to convert the extracted phone sequences to phonemes if they are to be added as baseforms in the lexicon. Although the sequences could be kept in their original form by creating a dummy set of units for the baseforms that are passed verbatim by the phonological rules, con-

verting to phonemes adds some small amount of generalization over allophones to the sequence’s pronunciation, and reduces the amount of competing forms that have to be dealt with later (see Section 0). I make the conversion in a naive way, classifying single or paired phonetic units into a set of equivalence classes that correspond to phonemes. For example, taps and cleanly enunciated stops are mapped to the same phoneme, with explicit closures being dropped. Although the procedure does not capture some contextual effects, it achieves perfectly adequate performance (see Section 0).

Phoneme sequences are given an arbitrary name and added to the list of vocabulary and baseforms. To ensure that the language model assigns some probability to these new vocabulary items the next time the recognizer runs, a collection of randomly generated sentences is added to those output of the recognizer used in re-training.

Dealing with rarely-used additions If a phoneme sequence introduced into the vocabulary is actually a common sound sequence in the acoustic data, then the recognizer will pick it up and use it. Otherwise, it just will not appear very often in hypotheses. After each iteration a histogram of phoneme sequence occurrences in the output of the recognizer is generated, and those below a threshold are cut.

Dealing with competing additions Very often, two or more very similar phoneme sequences will be added to the vocabulary. If the sounds they represent are in fact commonly occurring, both are likely to prosper and be used more or less interchangeably by the recognizer. This is unfortunate for language modeling purposes, since their statistics will not be pooled and so will be less robust. Happily, the output of the recognizer makes such situations very easy to detect. In particular, this kind of confusion can be uncovered through analysis of the N-best utterance hypotheses.

If we imagine a set of N-best hypotheses aligned and stacked vertically, then competition is indicated if two vocabulary items exhibit both of these properties:

- ▷ Horizontally repulsive - if one of the items appears in a single hypothesis, the other will not appear in its vicinity.
- ▷ Vertically attractive - the items frequently occur in the same part of a collection of hypotheses for a particular utterance.

Since the utterances in this domain are generally short and simple, it did not prove necessary to rigorously align the hypotheses. Instead, items were considered to be aligned based simply on the vocabulary items preceding and succeeding them. It is important to measure

both the attractive and repulsive conditions to distinguish competition from vocabulary items that are simply likely or unlikely to occur in close proximity.

Accumulating statistics about the above two properties across all utterances gives a reliable measure of whether two vocabulary items are essentially acoustically equivalent to the recognizer. If they are, they can be merged or pruned so that the statistics maintained by the language model will be well trained. For clear-cut cases, the competing items are merged as alternatives in the baseform entry for a single vocabulary unit. A better alternative might have been to use class n-grams and put the items into the same class, but this works fine. For less clear-cut cases, one item is simply deleted.

Here is an example of this process in operation. In this example, “phone” is a keyword present in the initial vocabulary. These are the 10-best hypotheses for the given utterance:

“what is the phone number for victor zue”

```
<oov> phone (nahmber) (mihterz) (yuw)
<oov> phone (nahmber) (mihterz) (zyuw)
<oov> phone (nahmber) (mihterz) (uw)
<oov> phone (nahmber) (mihterz) (z uw)
<oov> phone (ahmberf) (mihterz) (zyuw)
<oov> phone (ahmberf) (mihterz) (yuw)
<oov> (axfaanah) (mberfaxr) (mihterz)
(z yuw)
<oov> (axfaanah) (mberfaxr) (mihterz)
(yuw)
<oov> phone (ahmberf) (mihterz) (z uw)
<oov> phone (ahmberf) (mihterz) (uw)
```

The “<oov>” symbol corresponds to an out of vocabulary sequence. The phone sequences within parentheses are uses of items added to the vocabulary in a prior iteration of the algorithm. From this single utterance, we acquire evidence that:

- ▷ The entry for (ax f aa n ah) may be competing with the keyword “phone”. If this holds up statistically across all the utterances, the entry will be destroyed. The keyword vocabulary is given special status, since they represent a link to the outside world that should not be modified.
- ▷ (n ah m b er), (m b er f axr) and (ah m b er f) may be competing. They are compared against each other because all of them are followed by the same sequence (m ih t er z) and many of them are preceded by the same word “phone”.
- ▷ (y uw), (z y uw), and (uw) may be competing

All of these will be patched up for the next iteration. This use of the N-best utterance hypotheses is reminiscent of their application to computing a measure of recognition confidence in [11].

Testing for convergence For any iterative procedure, it is important to know when to stop. If we have transcribed data, we can track the keyword error rate on that data and halt when the increment in performance is sufficiently small.

If there is no transcribed data, then we cannot directly measure the error rate. We can however bound the rate at which it is changing by comparing keyword locations in the output of the recognizer between iterations. If few keywords are shifting location, then the error rate cannot be changing above a certain bound. We can therefore place a convergence criterion on this bound rather than on the actual keyword error rate. It is important to just measure changes in keyword locations, and not changes in vocabulary items added by clustering. Items that do not occur often tend to be destroyed and rediscovered continuously, making comparisons difficult.

7 Experiments in unsupervised vocabulary extension

The unsupervised procedure described in the previous section is intended to both improve recognition accuracy on the initial vocabulary, and to identify candidates for vocabulary extension. This section describes experiments that demonstrate to what degree these goals were achieved. To facilitate comparison with other ASR systems, results are quoted for a fairly typical domain called LCSInfo [9] developed by the SLS group at MIT. This domain consists of queries about personnel – their addresses, phone numbers etc.

7.1 Experiment 1: Qualitative Results

This section describes the candidate vocabulary discovered by the clustering procedure. Numerical, performance-related results are reported in the next section.

Results given here are from a clustering session with an initial vocabulary of five keywords (email, phone, room, office, address), run on a set of 1566 utterances. Transcriptions for the utterances were available but not used by the clustering procedure. Here are the top 10 clusters discovered on this very typical run, ranked by decreasing frequency of occurrence:

1	n ah m b er	6	p l iy z
2	w eh r ih z	7	ae ng k y uw
3	w ah t ih z	8	n ow
4	t eh l m iy	9	hh aw ax b aw
5	k ix n y uw	10	g r uw p

These clusters are used consistently by the recognizer in places corresponding to: "number, where_is, what_is, tell_me, can_you, please, thank_you, no, how_about, group," respectively in the transcription. The first, /n ah m b er/, is very frequent because of "phone number", "room number", and "office number". Once it appears as a cluster the language model is immediately able to improve recognition performance on

those keywords. Other high-frequency clusters correspond to common first names (Karen, Michael).

Every now and then a "parasite" appears such as /d h ax f ow n/ (from an instance of "the phone" that the recognizer fails to spot) or /iy n eh l/ (from "email"). These have the potential to interfere with the detection of the keywords they resemble acoustically. But as soon as they have any success, they are detected and eliminated as described in Section [sect]. It is possible that if a parasite doesn't get greedy, and for example limits itself to one person's pronunciation of a keyword, that it will not be detected, although I didn't see any examples of this happening.

Many simple sentences can be modeled completely after clustering, without need to fall back on the generic OOV phone model. For example, the utterances:

```
What is Victor Zue's room number
Please connect me to Leigh Deacon
```

are recognized as:

```
(w ah t ih z) (ih t er z uw) room (n ah m b
er)
(p l iy z) (k ix n eh k) (m iy t uw) (l iy d
iy) (k ix n)
```

All of which are entries in the vocabulary and so contribute to the language model. All the discovered vocabulary items are assigned one or more baseforms as described in Section [sect]. For example, the nasal in /n ah m b er/ is sometimes recognized, sometimes not, so both pronunciations are added to a single baseform.

7.2 Experiment 2: Quantitative Results

For experiments involving small vocabularies, it is appropriate to measure performance in terms of Keyword Error Rate (KER). I take this to be:

$$KER = \frac{F + M}{T} * 100 \quad (1)$$

with:

F : Number of false or poorly localized detections

M : Number of missed detections

T : True number of keyword occurrences in data

A detection is only counted as such if it occurs at the right time. Specifically, the midpoint of the hypothesized time interval must lie within the true time interval the keyword occupies. I take forced alignments of the test set as ground truth. This means that for testing it is better to omit utterances with artifacts and words outside the full vocabulary, so that the forced alignment is likely to be sufficiently precise.

The experiments here are designed to identify when clustering leads to reduced error rates on a keyword vocabulary. Since the form of clustering addressed in this paper is fundamentally about extending the vocabulary,

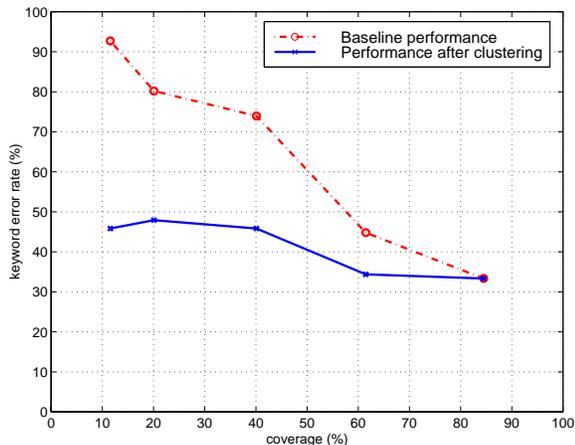


Fig. 2. Keyword error rate of baseline recognizer and clustering recognizer as total coverage varies.

we would expect it to be useless if the vocabulary is already large enough to give good coverage. We would expect it to offer the greatest improvement when the vocabulary is smallest. To measure the effect of coverage, the full vocabulary was made smaller and smaller by incrementally removing the most infrequent words. A set of keywords were chosen and kept constant in the vocabulary across all the experiments so the results would not be confounded by properties of the keywords themselves (for example, the most common word "the" would make a very bad keyword since it is often unstressed and loosely pronounced). The same set of keywords were used as in the previous section.

Clustering is again performed without making any use of transcripts. To truly eliminate any dependence on the transcripts, an acoustic model trained only on a different dataset was used. This reduced performance but made it easier to interpret the results.

Figure 2 shows a plot of error rates on the test data as the size of the vocabulary is varied to provide different degrees of coverage. The most striking result is that the clustering mechanism reduces the sensitivity of performance to drops in coverage. In this scenario, the error rate achieved with the full vocabulary (which gives 84.5% coverage on the training data) is 33.3%. When the coverage is low, the clustered solution error rate remains under 50% - in relative terms, the error increases by at most a half of its best value. Straight application of a language model gives error rates that more than double or treble the error rate.

As a reference point, the keyword error rate using a language model trained with the full vocabulary on the full set of transcriptions with an acoustic model trained on all available data gives an 8.3% KER.

7.3 Experiment 3: Kismet Domain

An exploratory experiment was carried out for data drawn from robot-directed speech collected for the Kismet robot. This data comes from an earlier series of recording sessions [7] rather than the ones described in Section 4. Early results are promising – semantically salient words such as “kismet”, “no”, “sorry”, “robot”, “okay” appear among the top ten clusters. But this work is in a very preliminary stage.

8 Discussion and Conclusions

Paper is a collection of stuff, not a unified whole. Work in progress. Hard hat area. Divers alarms and excursions. Exeunt all pursued by the furies.

This paper does not address the crucial issue of binding vocabulary to meaning. One line of research under way is to use transient, task-dependent vocabularies to communicate the temporal structure of processes. Another line of research looks more generally at how a robot can establish a shared basis for communication with human through learning expressive verbal behaviors as well as acquiring the humans’ existing linguistic labels.

Problem of affective speech – too much darned prosody. Good thing about prosody: it may help distinguish a word teaching scenario from normal conversation. The robot could operate in different modes then (mentioned in section 5.2).

Ultimate research interests: How can a robot establish a shared basis for communication? Informed by infant language research. Establishing a mechanism for a robot to vocalize its behavioral and internal state in a consistent manner understandable to humans.

THIS CAME FROM 4.1:

Vocal imitation and referential mapping Parents tend to interpret their children’s first utterances very generously and often attribute meaning and intent where there may be none [4]. It has been shown, however, that such a strategy may indeed help infants coordinate meaning and sound and learn to express themselves verbally. Pepperberg [18] formalized the concept into a teaching technique called referential mapping. The strategy is for the teacher to treat the pupil’s spontaneous utterances as meaningful, and act upon them. This, it is shown, will encourage the pupil to associate the utterance with the meaning that the teacher originally gave it, so the student will use the same vocalization again in the future to make a similar request or statement. The technique was successfully used in aiding the development of children with special needs.

For the purposes of the research reported here, we are not concerned with the meaning of words yet. However, one of the purposes of vocabulary extensions is to

build a shared basis for meaningful communication between the human and the robot, and referential mapping may be one of the promising lines of development. We are therefore interested in finding out how often humans spontaneously treat Kismet's utterances as meaningful. WHY DO I THINK THAT One way of doing this is to look at how often they are imitated by the humans. ?

Acknowledgements

The authors would like to thank Sherry Turkle, Jen Audley, Anita Chan, Tamara Knutsen, Becky Hurwitz, and the MIT Initiative on Technology and Self, for making available the video recordings that were analyzed in this paper.

Parts of this work rely heavily on speech recognition tools and corpora developed by the SLS group at MIT.

Funds for this project were provided by DARPA as part of the "Natural Tasking of Robots Based on Human Interaction Cues" project under contract number DABT 63-00-C-10102, and by the Nippon Telegraph and Telephone Corporation as part of the NTT/MIT Collaboration Agreement.

References

- [1] R.N. Aslin, J.Z. Woodward, N.P. LaMendola, and T.G. Bever. Models of word segmentation in fluent maternal speech to infants. In J.L. Morgan and K. Demuth, editors, *Signal to Syntax: Bootstrapping From Speech to Grammar in Early Acquisition*. Lawrence Erlbaum Associates: Mahwah, NJ, 1996.
- [2] E.G. Bard and A.H. Anderson. The unintelligibility of speech to children: effects of referent availability. *Journal of Child Language*, 21:623–648, 1994.
- [3] I. Bazzi and J.R. Glass. Modeling out-of-vocabulary words for robust speech recognition. In *Proc. 6th International Conference on Spoken Language Processing*, Beijing, China, October 2000.
- [4] P. Bloom. *How Children Learn the Meaning of Words*. Cambridge: MIT Press, 2000.
- [5] C. Breazeal. *Sociable Machines: Expressive Social Exchange Between Humans and Robots*. PhD thesis, MIT Department of Electrical Engineering and Computer Science, 2000.
- [6] M.R. Brent and J.M. Siskind. The role of exposure to isolated words in early vocabulary development. *Cognition*, 81:B33–B44, 2001.
- [7] C. Breazeal and L. Aryananda. Recognition of affective communicative intent in robot-directed speech. In *Proceedings of Humanoids 2000*, Cambridge, MA, September 2000.
- [8] J. Glass, J. Chang, and M. McCandless. A probabilistic framework for feature-based speech recognition. In *Proc. International Conference on Spoken Language Processing*, pages 2277–2280, 1996.
- [9] J. Glass and E. Weinstein. Speechbuilder: Facilitating spoken dialogue systems development. In *7th European Conference on Speech Communication and Technology*, Aalborg, Denmark, September 2001.
- [10] A.L. Gorin, D. Petrovksa-Delacrtaz, G. Riccardi, and J.H. Wright. Learning spoken language without transcriptions. In *Proc. IEEE Automatic Speech Recognition and Understanding Workshop*, Colorado, 1999.
- [11] T.J. Hazen and I. Bazzi. A comparison and combination of methods for oov word detection and word confidence scoring. In *Proc. International Conference on Acoustics*, Salt Lake City, Utah, May 2001.
- [12] J. Hirschberg, D. Litman, and M. Swerts. Prosodic cues to recognition errors. In *ASRU*.
- [13] P. Jeanrenaud, K. Ng, M. Siu, J.R. Rohlicek, and H. Gish. Phonetic-based word spotter: Various configurations and application to event spotting. In *Proc. EUROSPEECH*, 1993.
- [14] P.W. Jusczyk. *The Discovery of Spoken Language*. Cambridge: MIT Press, 1997.
- [15] P.W. Jusczyk and R.N. Aslin. Infants' detection of the sound patterns of words in fluent speech. *Cognitive Psychology*, 29:1–23, 1995.
- [16] Y. Matsusaka and T. Kobayashi. Human interface of humanoid robot realizing group communication in real space. In *Proc. Second International Symposium on Humanoid Robots*, pages 188–193, 1999.
- [17] T. Oates, Z. Eyler-Walker, and P. Cohen. Toward natural language interfaces for robotic agents: Grounding linguistic meaning in sensors. In *Proceedings of the 4th International Conference on Autonomous Agents*, pages 227–228, 2000.
- [18] I. Pepperberg. Referential mapping: A technique for attaching functional significance to the innovative utterances of an african grey parrot. *Applied Psycholinguistics*, 11:23–44, 1990.
- [19] D.K. Roy. *Learning Words from Sights and Sounds: A Computational Model*. PhD thesis, MIT, September 1999.
- [20] C. Trevarthen. Communication and cooperation in early infancy: a description of primary intersubjectivity. In M. Bullowa, editor, *Before Speech: The beginning of interpersonal communication*. Cambridge University Press, 1979.
- [21] J.F. Werker, V.L. Lloyd, J.E. Pegg, and L. Polka. Putting the baby in the bootstraps: Toward a more complete understanding of the role of the input in infant speech processing. In *Signal to Syntax: Bootstrapping From Speech to Grammar in Early Acquisition*.
- [22] V. Zue, J. Glass, J. Plifroni, C. Pao, and T.J. Hazen. Jupiter: A telephone-based conversation interface for weather information. *IEEE Transactions on Speech and Audio Processing*, 8:100–112, 2000.
- [23] V. Zue and J.R. Glass. Conversational interfaces: Advances and challenges. *Proceedings of the IEEE, Special Issue on Spoken Language Processing*, Vol. 88, August 2000.