Abstract

Proper early acquisition of speech and language appears to be a necessary process to reach mature speech communication. In modelling the process of natural (and pathological) speech production and speech perception, we frequently concentrate on specific aspects of phonetic knowledge. But also to improve the performance of speech technological systems, an intelligent interpretation of the abundant, but sometimes also incomplete or absent, phonetic information is highly advisable. This keynote will use the above framework to discuss the progress made in speech science and technology over the past 30 years.

1. Introduction

Acquiring and implementing phonetic knowledge, in my opinion, is a unifying concept in many fields of speech research. Babies who make their first steps in acquiring their mother tongue, just as second language learners, are confronted with it. Any psycholinguist, psychoacoustician, or phonetician studying normal and pathological speech production and speech perception concentrates on certain signal attributes and will try to model the consequences for speech perception. Speech technologists generally first try to stay away from specific phonetic knowledge, but if they really have to gain the final few percents in performance, this might be the area where some profit lays. This brings me to the following subdivision of important fields:

- early speech and language acquisition (L1 and L2)
- speech production and perception (normal and pathological)
- speech technology

As far as (phonetic) knowledge is concerned I think it is appropriate to distinguish:

- rules, regularities, analogies, synonyms
- knowledge stored in databases

Knowledge is scattered and not uniquely defined. The physical characteristics of, for instance, a prominent word or of a question intonation are not defined once and for all. Rather, certain features have to be present or can be traded against others. Acquiring knowledge is a life-long process for humans, only a small part of this knowledge is already available at birth [5]. Some systems are adaptive, but most speech machines go through a one-time training cycle after which their ‘knowledge’ is fixed. Confronted with a slightly different new situation, the machine generally lacks the flexibility to adapt to this new situation and to interpret whatever (other) information might be available.

Also knowledge implementation takes place in many different forms. It is a way of defining ‘learning’ and long-term memory. It is also a way to formalize certain findings, such as a model for vowel reduction [3], or for predicting speech intelligibility [26], or in the form of a network that predicts prominence from acoustic signal characteristics [28].

2. Early speech and language acquisition

Although early speech and language acquisition is certainly not my primary field of expertise, I supervised several projects in this area. My colleagues Florien Koopmans-van Beinum and Jeannette van der Stelt developed and ingenious system to describe early sound productions in terms of source characteristics and articulatory movements, thus creating a natural continuum between early sound productions, babbling, and the production of meaningful (pseudo-)words [14]. Much phonetic knowledge is reflected in this approach.

I find it especially intriguing to see how results on certain phonetic tests, such as phoneme identification and same-different judgements, might be early indicators for dyslexia [22]. Also the recent development of early hearing screening with babies may have serious implications for the way we test and monitor the early speech development towards mature language processing. If, through objective tests such as brainstem audiometry BERA, but in the future perhaps also via new imaging techniques such as PET scan and functional MRI, serious hearing defects can be detected early, one should also provide the means to monitor, train and test the early speech and language development (early detection requires early intervention!). Hearing aids for very young hearing-impaired children will be beneficial for speech and language development. But for optimal adaptation one cannot rely on subjective testing only. Also in developing and adapting digital hearing aids, there might be a role for speech scientists. The same is true for cochlear implants. Also in the whole discussion about cochlear implants for very young babies we may have a responsibility.

3. Speech production and speech perception

At Eurospeech conferences and otherwise, there is not so much interest anymore in presenting basic phonetic data about speech analysis, speech production and speech perception. What I am thinking of are experiments and procedures like phoneme identification, similarity judgments, trading relations, bite block experiments, effects of local context, speaker normalization, gating experiments, shadowing, matching and the like. Of course the experimental procedure is only a means to study certain phenomena, of which the experimental results then hopefully enrich our vision on speech production and perception. The various Ph.D. projects that I supervised in this field (see sect. 7)
almost all produced interesting and sometimes challenging results. Let me just mention an alternative model for acoustic and lexical vowel reduction [3], a description of perceiving dynamic speechlike sounds [32] and an acoustic description of consonant reduction [25].

4. Speech technology

Speech technology is a fascinating field of research and of applications, that barely existed 30 years ago. From a scientific point of view I see it as the test bed for all our acquired knowledge of speech phenomena.

If we would be able to produce fully natural sounding synthetic speech from reading any text in the appropriate style, then we would have solved the text interpretation and speech generation problem. We would do even better if we knew how to optimize synthetic speech for noisy communication channels, reverberant conditions, and elder or non-native listeners.

If we would be able to fully understand every word spoken in any utterance from any speaker and would be able to properly interpret the meaning of that utterance in its context, then we would have solved the speaker adaptation, the word recognition and the speech understanding problem.

Similarly speech-to-speech translation, topic spotting, and document retrieval, just as multi-lingual applications, are challenging tasks to our limited knowledge.

If we would be able to properly accomodate laryngectomized speakers with an artificial voice, hearing-impaired and deaf listeners with alternative stimulation via optimized digital hearing aids, cochlear implants or otherwise, and blind readers with natural speech output, then we would have given a true contribution to understanding the needs for the handicapped.

5. Speech databases

(Annotated) speech corpora are our latest knowledge source. Both in the field of speech recognition and speech synthesis there were good reasons for collecting such material. Only gradually many people in the speech and language community realized that more can be done with such corpora than just training context-independent or context-dependent phoneme models, or just concatenating carefully selected smaller units to bigger chunks of synthetic speech. They happen to be valuable resources for duration modelling [31], for pronunciation modelling, for intonation modelling, etc. The popularity of the TIMIT database for speech research is at least partly related to its design and its accessibility.

As two representative examples let me shortly describe the 10-millions words Spoken Dutch Corpus (CGN) [17] and the much smaller but fully phonemically segmented IFA-corpus [24].

In a jointly funded project by the Dutch and Flemish government, about 1,000 hours of contemporary standard Dutch, as spoken by adults in the Netherlands and Flanders, will be collected over a 5-years period, starting June 1998. The fourth of seven releases is due in October 2001. Each release contains its share of audio files as well as one or more CD-ROMs with annotations, this latter part is updated with each release. Future distribution via DVDs is considered. The Dutch Language Union (http://www.taalunie.org/) holds all the rights and the distribution will be done via ELRA (http://www.icp.grenet.fr/ELRA/), it is quite unique that the consent of all speakers and of any other parties involved will have been acquired. In order to serve as many needs as possible from a wide variety of speech and language researchers with various backgrounds, the corpus will contain both monologues and dialogues, both high-quality speech as well as telephone speech, both spontaneous and read speech, etc. The entire corpus will be orthographically transcribed by hand using ’praat’ (http://www.fon.hum.uva.nl/praat/), and will be automatically lemmatized and annotated with part-of-speech information. A selection of one million words will be phonetically transcribed and syntactically annotated. One quarter of that will also be prosodically annotated. The CGN-website is a valuable source of information for all interested parties (http://lands.let.kun.nl/cgn/home.htm).

In this phase of the project, there is still more emphasis on how to collect and annotate all the material, then on how to make proper use of this CGN-corpus. A mid-term evaluation of the progress of the project will most probably be performed by BAS in Muenich.

In compiling the 50,000-words IFA-corpus we built on the experiences gained so far within the CGN-project, but we also introduced several of our own demands. Most notably, this corpus contains much more speech (half an hour on average per speaker), of a higher recording quality, from less Dutch speakers (4 male and 4 female). However, per speaker various speaking styles (informal story telling, reading, retelling a read story, reading lists of sentences, words and syllables from the narrative stories) are available for comparison. Furthermore, all speech is phonemically segmented and labeled. Initial phoneme labels and segment boundaries were introduced automatically (on the basis of the CELEX pronunciation lexicon and a phone-based HMM recognizer) and then hand corrected by trained labelers. All compiled data tables are fed into a PostgreSQL database for subsequent data mining. Some initial results about speaking rates and phoneme durations are already available [24], but more intricate questions can hopefully be solved using the powerful query language SQL. The corpus is freely available and accessible on line under the GNU General Public License (http://www.fon.hum.uva.nl/IFAcorpus/).

6. 30 Years ago

In September 2001 when Eurospeech 2001 - Scandinavia takes place in Aalborg, Denmark, it will be almost exactly 30 years ago that I had my first experience of being actively involved in a major speech conference. This was the 7th International Congress on Acoustics (ICA) in August 1971 in Budapest, Hungary. This event to me seems to be an appropriate landmark for comparing progress in our field since that time (for all authors mentioned below, see papers in the Proceedings of the 7th Int. Congress on Acoustics).

Many of the younger people will find it hard to imagine that in 1971 there was no text-to-speech synthesis-by-rule yet, diphones did not yet exist, only formant analysis-resynthesis using for instance Liljencrants’ and Fant’s OVE III. Rabiner’s digital hardware for speech synthesis, using a cascade of 2-pole digital filters, was able to generate sentences like ‘we were away a year ago’, ‘may we all learn a yellow lion roar’, and ‘few thieves are never sent to the judge’. Only isolated word recognition was possible with vocabulary sizes of some 50 words using template matching. Sakoe had just introduced dynamic processing for time normalization, Atal presented his initial ideas about predictive coding. The probabilistic
approach was yet unknown. Dreyfus-Graf advocated an artificial language that would simplify automatic recognition. Erman was the first to talk about telephone input and Neely about speech recognition in noise, but their word sets were still limited to 54 isolated words, carefully spoken by one male speaker, being Ken Stevens. I used a list of 50 Dutch words spoken by 5 males to produce a dimensional representation of bandfilter spectra for recognizing the more or less stationary phoneme-like segments.

Kozhevnikov presented a paper about the perception of amplitude modulated vowel-like stimuli. Chistovich talked about vowel discrimination and she also gave one of the plenary lectures in which she emphasized the importance of psychoacoustics for speech perception. Also then already Flanagan gave a keynote on focal points in speech communication research. Sundberg showed real time pitch extraction of folk music, whereas Matthews demonstrated music synthesis. As nowadays is very popular with all the ISCA Tutorial and Research Workshops and other satellite events surrounding Eurospeech, then also we already extended our presence in Hungary with a most enjoyable 3-days Speech Symposium in the city of Szeged, in the southern part of Hungary close to the Bugac Puszt. It was special because of the place and of the period in which it took place. It was also special because of the topic and because of the people involved.

Another early scientific meeting made a large impression on me for various reasons, this was the Symposium on ‘Auditory Analysis and Perception of Speech’ [10], organized by Gunnar Fant and Ludmilla Chistovich in August 1973 in, what then still was called, the city of Leningrad. It was special because of the topic and because of the place involved. One last memorable event that I want to mention here is the Symposium on ‘Invariance and Variability of Speech Processes’ [18], that was organized by Joe Perkell and Dennis Klatt in Cambridge, Massachusetts in October 1983. This was one year after I was appointed as a full professor in Phonetic Sciences at the University of Amsterdam. The search for systematic variability has lead my research ever since.

7. Joint research with Ph.D. students

I was privileged to be able to supervise since 1982 some 25 Ph.D. theses of which most were successfully completed, which in Holland implies the production of a good-looking booklet that is distributed world-wide to colleague scientists. Of course this work is generally also reported about at conferences, workshops and in the open literature. One reason for choosing such a wide theme for my keynote, was the fact that I wanted to emphasize that my scientific career only has systematic variability to thank for choosing such a wide theme for my keynote, was the fact that my scientific career only has been possible through intensive cooperative work with our students in many different fields of speech science and technology. Out of respect to all their hard work I include in this booklet that is distributed world-wide to colleague scientists.

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8. Developments over the last 30 years

Of course there has been a tremendous progress over the last 30 years. The almost 700 papers that will be presented at this Eurospeech conference alone, may well represent a larger amount of publications in our field than everything published in the whole of 1971. I expect that the number of people presently active in the field of speech science and technology will be well over 10,000 world wide. If 1,000 of them come together here in Aalborg and are a member of ISCA, then this is a powerful task force to solve communicative problems in our society.

However, the joint phonetic knowledge in the minds and in the PCs of these people is apparently far from sufficient to solve today’s communicative demands. Telephone communication is everywhere but the speech quality of GSMs is deplorable. Speech is considered to be the most natural form of communication, but a proper dialogue with a computer information system is still in an experimental phase. We are able to produce intelligible synthetic speech, but naturalness is at stake and speaker and speaking style characteristics cannot be properly controlled [19]. Automatic speech recognition has made tremendous progress over the last 30 years, but still much remains to be desired, such as greater robustness and quick adaptation. Speech and language technology could be used more in education, language learning and aids for the handicapped.

Also many ‘smaller’ questions remain to be solved in the domain of phonetic sciences.

- How do listeners normalize over speakers? Even young children can cope with this problem, but we still don’t know how they do it.

- How do listeners handle speech variation? Because of coarticulation, prosodic variation, speaking style, et c., no two utterances are ever the same. How do we interpret this variation? Is there any random variation or is there always a cause for any specific source of variation?
- What is a realistic front end processor that efficiently extracts all the relevant information from the speech signal, also for high-pitched voices or noisy speech?
- What are the processes in acquiring our mother tongue and foreign languages? What are the implications of a speaking or hearing defect, what of hearing aids and cochlear implants?

I feel privileged to have been part of this lively speech community for some 30 years and I have high expectations of progress in the years to come, partly because I have the impression that phonetic knowledge nowadays is more accessible and can more easily be implemented in descriptive models (e.g., computational phonetics [20]) and in technological systems.

9. References


