

DATABASES FOR SPEAKER RECOGNITION: ACTIVITIES IN COST250 WORKING GROUP 2

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ABSTRACT

Working Group (WG) 2 of the COST250 Action “Speaker Recognition in Telephony” has dealt with databases for speaker recognition. The present final report gives an overview of the activities in this WG, and presents its main results. The first result is an overview of 36 existing databases that has been used in speaker recognition research. Those include both public and proprietary databases. As part of the overview, some of the variability represented in those databases is analyzed. The second result is the publicly available Polycost database, a telephony-speech multi-session database with 134 speakers from all around Europe. Together with pre-defined experiment specifications, this database is a useful resource to aid in the assessment of speaker recognition systems in general, and in comparing systems across sites, in particular.

1 INTRODUCTION

The availability of good speech databases is crucial for development and assessment in all branches of speech research. In the beginning of the COST250 Action, in January 1995, Working Group 2 (WG2) was formed to deal with questions related to the availability, design, collection and production of speech databases for speaker recognition. The present final report gives an overview of the activities and results of WG2.

The three main activities were first to list and characterize speech databases that were available to COST250 participants; then to design, record and produce the now publicly available Polycost database; and finally, to do case studies of several database projects. Section 2 of this report gives an overview of many of the existing databases that has been used in speaker recognition research, while section 3 presents a basis for detailed characterization of speaker recognition databases. The characteristics of the Polycost database is then given in section 4, along with an outline of the various stages traversed to create this database. A list of publications where Polycost has been used in research is also given. Section 5 summarizes the case study presentations made during WG2 technical sessions. The Working Group’s contribution to the research field of speaker recognition is finally summarized in section 6.

2 DATABASE SURVEY

The first task set out for WG2 was to survey what databases were available to each COST250 partner; what was their characteristics, and was there a database that could potentially be used by all partners? The first step was to define a basis for the survey in terms of a series of questions, and compile them into a questionnaire. Secondly, the questionnaire was distributed among partners, and finally, answers to the

questionnaire was collected and compiled into a comprehensive report. This report covers 18 databases and was finalized in June 1996 (Lindberg et al., 1996).

Table 1 contains a comprehensive (not necessarily exhaustive) list of speech databases that have been used in speaker recognition research around the world, with an emphasis on data from European laboratories. In many, not all, cases the databases are available to other researchers. The list includes databases from the above mentioned survey, but also several other databases that were either missing from the survey or have been produced after the survey was completed. For databases that were developed further since June 1996, information in the table has been updated with respect to the original survey.

It is interesting to look at the variability the various databases in Table 1 represent. Most of the databases are multi-session databases that capture temporal intra-speaker variability. Several contain recordings from the same speaker during three months or more (often motivated by an investigation in (Furui, 1986) where it is reported that the size of a speaker's feature subspace increases during the first three months of measurement and is fairly constant after that). In some cases, speakers have been recorded during more than a year: 1½ years in PolyVar and Gandalf, and 2 years in the CSLU Speaker Recognition Corpus (references to database publications are collected in Table 1).

Many multi-session databases contain telephone handset variability in addition to temporal intra-speaker variability. In GAUDI, for example, female target speakers were recorded four times through dual microphones in a studio, and five times over telephone lines from different locations and handsets each time. In Gandalf, speakers placed half the calls from a single "favorite" handset and half the calls from 2-9 other handsets during the first four months of the recording period. With this design, temporal intra-speaker variability and handset variability can be studied in parallel. In the Switchboard-II, 'LIMSI/CNET', CSLU and SpeechDat speaker recognition databases, special attention has also been paid to handset variation. Handset variability is often accompanied with variability in caller environment.

A single-session database can be useful when studying a particular (non-temporal) variability in isolation. A few of the databases in Table 1 have been designed to include such variability; two with intra-speaker variability and several with variability in recording conditions:

- VeriVox: voluntary intra-speaker variation from weak, strong, slow, fast, and denasalised speech, and involuntary variation from speech in noise and speech under stress (Karlsson et al., 1998),
- 'Disguised voices': the effect of nine different techniques for voice disguise,
- LLHDB, HTIMIT and SR4X: telephone handset variation,
- TSID: radio transmitter and receiver variation,
- NTIMIT and CTIMIT: telephone circuit variation (in fixed and mobile networks respectively).

Note that a single-session database is not useful for estimating the absolute level of performance for a speaker recognition system in a practical application because it will never include normal temporal intra-speaker variability.

Two of the databases in Table 1 contain speech from the same speaker in two languages. The AHUMADA-Mataro subset of the GAUDI database contains speech from bilingual speakers of Castilian Spanish and Catalan, and speakers in Polycost speak both English and their mother tongue.

Regarding inter-speaker variability, it is of course desirable to include many speakers to achieve a good sampling of a speaker population. At the same time, it is often desirable to get a good sampling of individual speakers over multiple sessions to study intra-speaker variability. With only limited resources for database creation, a trade-off between the number of speakers, number of sessions and total cost is often necessary. A common design technique, used in many databases (SpeechDat, Switchboard-1, PolyVar, 'LIMSI/CNET', SIVA, Broertjes-Polyphone, GAUDI, and Gandalf), is to record a smaller set of speakers in many sessions and a separate, larger set of speakers in a single session. With this technique, one can achieve both good sampling of inter-speaker variability in potential impostor speakers, and of intra-speaker variability in client speakers. Note that the same effect may be achieved if a multi-session speaker recognition database is used together with a database designed for speech recognition purposes. This is the case with, for instance, the SpeechDat and Broertjes-Polyphone databases.

Name	Creator(s)	Distributor	Language	Cat. 1 male	Cat. 1 female	Cat. 2 male	Cat. 2 female	Recording	Digits	Words	Sentences	Spontaneous	References	'96 survey
EUROM-1, Danish	Tele Danmark, CPK	ELRA	da	5	5	30	20	mic	•	•	•	•	[40]	•
Polycost	COST250	ELRA	en + 16	74	60	-	-	tel	•	•	•	•	[31,51]	
Brent	BT		en GB	50	50	-	-	tel	•	•	•	•	[5,62]	•
Millar	BT		en GB	63	-	-	-	mic	•	•	•	•	[4]	
SpeechDat (FDB+SDB)	GPT Limited	ELRA	en GB	60	60	2500	2500	tel	•	•	•	•	[34,54]	
XM2VTS	Univ. of Surrey, M2VTS-project		en GB	295	-	-	-	mic, video	•	•	•	•	[6,53]	
CSLU Speaker Recognition Corpus	OGI/CSLU	OGI	en US	¹ 47	¹ 53	-	-	tel	•	•	•	•	[11]	
Disguised voices	Hollinell, Meverly		en US	-	-	7	5	tel, mic	•	•	•	•	[40]	•
KING-92	ITT	LDC	en US	51	0	-	-	tel, mic	•	•	•	•	[10,27]	
LLHDB	MIT-LL	LDC	en US	-	-	24	29	mic ³⁾	•	•	•	•	[66]	
SR4X	OGI/CSLU	OGI	en US	-	-	36	-	tel	•	•	•	•	[57]	
Switchboard-1 (incl. SPIDRE subset)	Texas Instruments, NIST, LDC	LDC	en US	² 22	² 23	² 280	² 218	tel	•	•	•	•	[10,26]	
Switchboard-2, phase I	LDC	LDC	en US	358	299	-	-	tel	•	•	•	•	[10]	
Switchboard-2, phase II	LDC	LDC	en US	679	-	-	-	tel	•	•	•	•	[10]	
TIMIT (+N/C/H/FFM-TIMIT)	MIT, SRI, TI (+ others)	LDC	en US	-	-	630	-	mic (+tel)	•	•	•	•	[10,66]	
TSID	MIT-LL, LDC	LDC	en US	-	-	31	4	mic, radio	•	•	•	•	[38]	
YOHO	ITT, Oklahoma State Univ.	LDC	en US	106	32	-	-	mic	•	•	•	•	[10]	•
GAUDI (incl. Ahumada)	Univ. Politéc. Madrid and Mataro		es, ca	104	101	120	130	tel, mic	•	•	•	•	[61,69]	
TelVoice	Univ. of Vigo		es	39	20	-	-	tel	•	•	•	•	[67,68]	
M2VTS	UCL, M2VTS-project	ELRA	fr BE	25	12	-	-	mic, video	•	•	•	•	[6,65]	
LoCoMic	IDIAP		fr CH	22	-	-	-	mic	•	•	•	•	[6]	
Polycode	IDIAP		fr CH	10	10	-	-	tel	•	•	•	•		•
PolyVar (incl. SpeechDat subset)	IDIAP	ELRA	fr CH	43	28	42	30	tel	•	•	•	•	[10,43]	•
'LIMS/CNET'	CNET, LIMS, Vecsys		fr FR	100	-	1000	-	tel	•	•	•	•	[24,37]	
RECLOC	CRIN/INRIA		fr FR	17	7	-	-	mic	•	•	•	•	[41,52]	•
SpeechDat (FDB+SDB)	Matra Communication	ELRA	fr FR	60	60	2500	2500	tel	•	•	•	•	[34,54]	
SUBTV ⁴⁾	CRIN/INRIA		fr FR	-	-	161	63	mic	•	•	•	•	[40]	•
SIVA	FUB	ELRA	it	207	229	128	127	tel	•	•	•	•	[10,20]	•
Broertjes+Polyphone ⁵⁾	KUN+KPN Research		nl	100	0	2616	2434	tel	•	•	•	•	[9]	•
SESP	KPN Research		nl	23	22	1	0	tel	•	•	•	•	[7,9]	•
SESP 2	KPN Research		nl	84	64	-	-	tel	•	•	•	•	[8]	
SESP III	KPN Research		nl	-	-	-	-	tel	•	•	•	•		
Russian speech database	STC/St. Petersburg	ELRA	ru	54	35	-	-	mic	•	•	•	•	[16]	
Gandalf	KTH		sv	48	38	51	32	tel	•	•	•	•	[45,49]	•
VeriVox	KTH, VeriVox-proj.		sv	-	-	50	0	mic	•	•	•	•	[35]	

Table 1. An overview on 36 databases sorted according to language. 'Cat. 1' include speakers recorded in more than one session, and 'Cat. 2' speakers a single session. '96 survey' indicates details for a database are included in the 1996 database survey. 1) recording is under way – the number of speakers will increase; 2) No. of category 1-speakers includes targets in the SPIDRE subset. More speakers are recorded in multiple sessions; 3) Each speaker was recorded through 10 different telephone handsets without passing the signal through a telephone circuit; 4) SUBTV: 'short utterance-based talker verification'; 5) Broertjes contains Polyphone-like sessions so the two can be used to complement each other.

Other interesting aspects of inter-speaker variability is the inclusion of close relatives among speakers, and of human or technical mimicry. In the Broertjes, Brent, Gandalf and SpeechDat databases, pairs of close relatives, such as twins, siblings, father-son and mother-daughter, have been included. Mimicry has been included in the CSLU database, where in each call a speaker is asked to imitate a given prompt phrase. In the literature are also studies where speech from one person has been transformed by technical means to impersonate another speaker. In for instance (Lindberg and Blomberg, 1999) two speakers from Gandalf were used as target speakers, and in (Genoud and Chollet, 1999) 18 target speakers from PolyVar were used.

Recent work by (Campbell Jr. and Reynolds, 1999) gives a good and more detailed overview of currently publicly available databases for speaker recognition evaluation. Another database overview is (Godfrey et al., 1994).

3 DATABASE CHARACTERIZATION

An important secondary outcome of work with the database survey is a series of questions for characterizing a speaker recognition corpus. The questionnaire was updated during 1998 with several new questions, and the complete updated questionnaire is included as Appendix A. Questions are divided into the following topics: 1) name and availability, 2) general information, 3) speaker material (including questions on the number of speakers, inter-speaker variation, intra-speaker variation, and impostor characterization), 4) speech contents, 5) post-processing, 6) recording equipment, 7) recording environment, 8) publications, and 9) other information. An example of a corpus description based on the questionnaire is included in section 4.3, where the Polycost database is described.

Other work on the characterization of databases was presented in (Falcone, 1995b). A strategy for how to characterize speaker recognition databases was presented, including measures of acoustic properties over time and frequency. Special attention was dedicated to possible ad hoc measures using results from a standardized speaker recognition system (a reference system). More information is available in (Falcone and Contino, 1995). The ideas around the use of a reference system have been further developed in WG4 (Falcone, 1999).

4 POLYCOST

A conclusion from the database survey was that none of the existing databases was both 1) suitable as a common research corpus within COST250, and 2) could be made available to everybody. It was decided that some effort should be dedicated to creating a common database for the COST250 partners. It was argued that if all partners had access to a common database, it would increase the possibilities for comparing work done by different partners. A database together with standardized experiment specifications for it would allow collaborative research to a greater extent than if each partner worked with its own data, protocols and algorithms. Such a database would therefore be a good support for work in the other working groups. Finally, creating the database would be a practical exercise that should give useful insights into many of the problems that the Working Group was set up to deal with. To this end the Polycost database (Petrovska et al., 1998) (Hennebert et al., 1999) (Campbell Jr. and Reynolds, 1999) was subsequently created.

The following sections describe each of the steps in creating the Polycost database. They also give a short summary of the properties of the database and cite several references where it has been used in experiments. The latest information on Polycost is published on the Polycost home page at <http://circwww.epfl.ch/polycost>.

4.1 DESIGN, RECORDING AND DISTRIBUTION

During 1996 the Polycost database was designed, recorded and distributed among COST250 members as a first preliminary CD-ROM version. The design was based on the Polycode database previously recorded at IDIAP (Switzerland). Recording was done in Switzerland at IDIAP and at the Signal Processing Laboratory (LTS) at EPFL. The recording platform developed at IDIAP was based on a Sun XTL platform and recorded data off an ISDN subscriber line. Speakers were recruited from each of the participating countries. The first preliminary version was produced at KTH (Sweden) with sponsoring from Telia InfoMedia AB (Sweden). For the final distribution of Polycost a contract has been made with the European Language Resources Association (ELRA). The database will be distributed for commercial and research purposes to a wide audience at a low cost. Part of the revenue from distribution is used to finance the annotation work at Circuits and Systems Laboratory (CIRC) at EPFL.

4.2 POST PROCESSING

The annotation work was done during 1997 to 1999 (Petrovska et al., 1996) (Hennebert, 1997) (Hennebert et al., 1998). The mother tongue utterances were transcribed by several COST250 partners. The utterances spoken in English were annotated at CIRC/EPFL and KTH. The initial goal was to provide word-level transcriptions together with word segmentation information, all to be produced at EPFL (Petrovska et al., 1996). This goal was later revised to include only transcriptions; producing the segmentation information required more manual labor than expected, and would take too long time to complete, given the resources available at EPFL. Segmentation information is thus provided for only a subset of the files in the database while a transcription is available for every file.

4.3 DESCRIPTION

This section gives a condensed description of Polycost based on the questionnaire for characterizing speaker recognition corpora referred to in section 3.

1. **Name and availability:** Polycost was recorded in Switzerland during January-March 1996. A first release on two CD-ROM was made in June 1996 within COST250, and has been publicly available from ELRA at a low price since 1998. The next release, including annotations, is planned for the end of 1999.
2. **General information:** the corpus contains speech data and was designed for general-purpose experiments in speaker recognition. It is also suitable for language and accent identification and speech recognition experiments.
3. **Speaker material:** 74 male and 60 female speakers. **Inter-speaker variation:** Most of the speakers are actively involved in speech research. 85% were between 20 and 39 years old. Hence, the speaker group exhibits relatively small variation in age, profession and educational background. On the other hand, variation in language and accent background is large. A summary judgement of inter-speaker variation on a scale 1-5 would therefore be around 3. **Intra-speaker variation:** Three speakers were recorded in 2-5 sessions, 105 speakers in 6-10 sessions, and 25 speakers in 11-15 sessions. Most speakers were recorded during 2-3 months. The effective duration of speech in each call is approximately 60 seconds. **Impostor characterization:** casual impostors only.
4. **Speech contents, language:** All speakers are recorded both in English and in their mother tongue. 85% are non-native speakers of English. 17 languages are represented: nine with five speakers or more (French, English, Dutch, Turkish, Swedish, Italian, Danish, Spanish and Portuguese) and eight with less than five speakers (Catalan, Arabic, Russian, Polish, Macedonian, Lithuanian, Galician and German). **Text material:** Out of 14 files in each session, 12 are text-dependent and two are text-independent. Among the text-dependent files are 10 sequences of digits (including one telephone number) and two sentences. All are fixed from session to session and are spoken in English. Text-independent items are spoken in the speaker's mother tongue: one in response to a written question and the other as a free monologue.
5. **Post-processing:** All files have an orthographic transcription. Six of the digit sequences also have a verified word-level segmentation.

6. **Recording equipment:** All sessions are recorded from telephone calls made from unknown types of handsets to a Euro-ISDN digital subscriber line. Around 80% of the speakers made all their calls from the same handset each time.
7. **Recording environment:** most calls were made from a home or office environment with varying (uncontrolled) levels of noise.
8. **Publications:** (Petrovska et al., 1998) and (Hennebert et al., 1999) describe the database, while (Melin and Lindberg, 1996) and (Melin and Lindberg, 1999) define baseline experiments. Table 2 lists references with recognition results.
9. **Other information:** Special features are the use of international telephone lines, non-native speakers of English, and speech in both English and the speaker's mother tongue.

4.4 PUBLICITY

Outside of COST250, the Polycost database has been presented at the Workshop on Speaker Recognition and its Commercial and Forensic Applications (RLA2C) in Avignon, April 1998 (Petrovska et al., 1998). An extended version of that paper has further been submitted to a special issue of Speech Communication (Petrovska et al., 1999). Polycost has been included in a recent overview of public speaker recognition corpora (Campbell Jr. and Reynolds, 1999). The latest information on Polycost is published on the Polycost home page at <http://circwww.epfl.ch/polycost>.

4.5 EXPLOITATION

A set of four standard baseline experiments has been defined for Polycost. These were first presented as a version 1.0 in (Melin and Lindberg, 1996). Following initial results (Nordström et al., 1998), the specification was subsequently revised to a version 2.0 (Melin and Lindberg, 1999). The specification of the baseline experiments is further treated within the scope of WG4 (Falcone, 1999).

Results from experiments on Polycost have been reported at COST250 meetings (Olsen and Lindberg, 1999) and at major international conferences. An overview of those reports is given in Table 2.

BE	Task description	Version 1.0	Version 2.0
1	Text-dependent verification, sentence	(Hernando, 1998), (Hernando and Nadeu, 1998), (Nordström and Melin, 1998), (Nordström et al., 1998)	(Nordström et al., 1998)
2	Text-dependent verification, 10 digits	(Melin, 1998), (Nordström and Melin, 1998), (Melin et al., 1998), (Nordström et al., 1998)	(Nordström et al., 1998), (Melin and Lindberg, 1999)
3	Text-independent verification	(Nordström and Melin, 1998), (Nordström et al., 1998), (Durou, 1999)	(Nordström et al., 1998)
4*	Text-independent identification	(Ambikairajah and Hassel, 1996), (Hassel and Ambikairajah, 1997), (Durou and Magrin-Chagnolleau, 1997), (Durou, 1998), (Altincay and Demirekler, 1998), (Demirekler, 1999), (Altincay and Demirekler, 1999), (Magrin-Chagnolleau and Durou, 1999)	

Table 2. An overview of publications that report on results from baseline experiments (BE) 1-4 with Polycost. *Note that experiments on speaker identification (last row) are made with variations of BE4, with the main deviation from BE4 specification being the choice of different target speaker subsets.

5 CASE STUDIES

During the course of the COST 250 Action, several presentations were made on the design, recording procedure, and post-processing techniques used in several speaker recognition database projects. Those

presentations were quite suitable as case studies for the Working Group. A summary of each presentation is given in the following sections, along with references to more information on the various databases.

5.1 SIVA

The SIVA Italian telephone-speech database was designed for experiments on speaker recognition. It contains as many as 691 speakers recorded in 1-26 sessions each. The design of the database, the hardware and software setup of the recording system, and the definition and realization of a pilot experiment were described in this presentation (Falcone, 1995a). Further information about SIVA is available in (Falcone and Gallo, 1996).

5.2 GANDALF

The Gandalf telephone speech database is designed for experiments on speaker recognition with special attention to intra-speaker and handset variability. It contains 86 client speakers recorded in 17-29 sessions over a period of 1½ years, and 83 impostor speakers. Besides audio files, a lot of information related to speakers and sessions have been collected, such as the type of telephone handset, a characterization of background noise, and occurrences of head colds and sore throats. In this presentation (Melin, 1995), the database design and experiences from the first stage of data collection was communicated. More information on this database is available in (Melin, 1996).

5.3 SPEECHDAT

SpeechDat (II) is an EU-funded project (LE2-4001) with the goal of producing large telephone speech databases in several European languages (Höge et al., 1997). Three categories of databases are recorded: 5000 speaker fixed telephone network, 1000 speaker mobile telephone network and 120-speaker/20-session speaker verification database. This presentation (Lindberg, 1996) described the design and specification of the speaker verification databases. More information is available in (Nataf, 1996).

5.4 M2VTS

In part of a presentation (García-Plaza and Fernández, 1996) from a representative of the M2VTS project (ACTS, AC102), the initial efforts of creating a multi-modal database for speaker recognition was described. The database contains the speech signal plus video sequences from three viewing angles. At the time of this presentation, 37 French subjects had been recorded, but up to 300 subjects was planned for. More information on this database is available in (Pigeon and Vandendorpe, 1997). An extension database, XM2VTS, has later been created with 300 British English speakers (Messer et al., 1999).

5.5 POLYCAST

The annotation method used for Polycast digit utterances was presented (Petrovska et al., 1996). With this method, only 16% of the seven-digit utterances and 30% of the ten-digit utterances had to be manually annotated. Annotation in this case involves both transcription and marking word boundaries in time. In summary, a connected digit recognizer was iteratively trained on parts of the data. This recognizer was then applied to all digit utterances, and for those utterances where the recognized sequence was identical to the manuscript sequence, the recognition result was taken as the correct annotation for the utterance. Results for the remaining utterances were used as templates for manual annotation.

5.6 SESP

SESP is a Dutch telephone-speech database designed for experiments on speaker recognition. It contains speech from 45 speakers recorded in 21-32 sessions each. Each speaker placed calls from a variety of handsets and from many types of locations. A substantial proportion of the calls came from foreign countries. During the presentation (Kuitert, 1996), the database was described and some preliminary results from the CAVE project were given. SESP was the main research corpus during the second half of the

CAVE project (LE1-1930) (Bimbot et al., 1998). SESP and an extension, SESP 2, are used extensively in the PICASSO project (LE4-8369) (Bimbot et al., 1999).

5.7 GAUDI/AHUMADA

GAUDI (including the AHUMADA and AHUMADA-Mataro subsets) is a Castilian Spanish speaker recognition corpus designated to commercial and forensic tasks (in the Mataro subset, subjects speak Catalan in addition to Castilian Spanish). It was designed with special attention to intra-speaker variability and external variability, and contains 104 male target speakers recorded in six sessions and 101 female target speakers recorded in nine sessions, plus 120 male and 130 female speakers recorded in a single session (impostors). The design and collection of the database was described (Ortega-García, 1999), and results were presented for a text-independent verification task with a 25-speaker subset and a GMM-based recognition system. Influences on the error rate from changes in speaking-style and microphone type were demonstrated. This database is further described in (Ortega-García et al., 1998), (Ortega-García et al., 1999) and (Satué-Villar and Faúndez-Zanuy, 1999).

6 CONCLUSION

The COST250 Action has been an important forum for the discussion of ideas on the design and implementation of databases for speaker recognition. It has given its participants an insight into several problems involved in the creation of such databases, and into various approaches to solving those problems. The Action has also provided its participants and the wider research community with a good overview of several existing, public and proprietary, databases. Last, but not least, the Action has provided the research community with Polycost, a new public database created for research in speaker recognition. It is our hope and belief that this database will be a useful speech resource that will contribute to the advances of state-of-the-art in speaker recognition long after the end of the COST250 Action.

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CHARACTERIZATION OF SPEAKER
RECOGNITION CORPORA

1 NAME AND AVAILABILITY

- 1A. Name of the corpus:
- 1B. Production date:
- 1C. Creator(s):
- 1D. Country (recording site):
- 1E. Currently available (check one or two)
- to anybody who wants to buy it
 - with the following restrictions:.....
 - not available by default, but availability may be negotiated with (contact information):.....
 - is not currently available, but will/may be available around (date):..... from (supplier):..... with the following restrictions:.....
 - a proprietary corpus that is not and never will be available to anyone else.
- 1F. Source for publicly available corpus (if other than ELRA or LDC, please give full contact information):
.....
.....
- 1G. Price for publicly available corpus:
- 1H. Size (Number of CDs, or size in MB):

2 GENERAL INFORMATION

- 2A. The corpus contains:
- speech
 - pictures
 - video (sequence of pictures)
 - other biometric data:
- 2B. It was created for the following purpose(s):
.....
.....
- 2C. It is (also) suitable for the following type of experiments:
.....
.....

3 SPEAKER MATERIAL

3.1 NUMBER OF SPEAKERS

Speakers in a corpus can be placed in one of two categories depending on how many different recording sessions they have made.

CATEGORY 1: Speakers who have made *more than one* recording session. For these speakers, one session can be used for enrollment and one or more for verification tests (client speakers).

CATEGORY 2: Speakers who have made *only one* recording session (impostor speakers).

3A. How many speakers in

	Male	Female
Category 1:
Category 2:

3.2 INTER-SPEAKER VARIATION

		% of the category population	
		Category 1	Category 2
3B.	Well motivated speakers: % %
3C.	People working with speech research: % %
3D.	Age		
	≤ 20 % %
	21-30 % %
	31-40 % %
	41-50 % %
	51-60 % %
	61-70 % %
	71-80 % %
	> 80 % %

3E. Your judgement of inter-speaker variation on a scale 1-5 (1-little variation, 5-much variation):

Motivation: (consider for instance variation on: dialects, profession, educational background):

.....

.....

3.3 INTRA-SPEAKER VARIATION

During what time period(s) have the category 1 speakers been recorded? If it varies, specify minimum and maximum period for the subject, and the most typical period.

3F. Number of sessions per speaker (category 1):

#sessions	#speakers
2- 5
6-10
11-15
16-20
21-25
25-	

3G. Time span during which category-1 speakers have been recorded:

time span	#speakers
1 day or less
2-7 days
2-4 weeks
2-3 months
4-12 months
more than one year

3H. Total effective duration of recorded speech per category-1 speaker:

Minimum minutes
Typical minutes
Maximum minutes

3I. Comments on intra-speaker variation:

.....

3.4 CHARACTERIZATION OF IMPOSTORS

Speakers that can be used for impostor tests (not necessarily only category 2 speakers) may have different characteristics. They may be

CASUAL IMPOSTORS: speakers like any other speaker in the corpus, who just say a set of utterances.

RELATIVES: speakers who are closely related to one or more speakers in the corpus and, hence, may be anatomically similar. Examples: brothers, twin brothers, or father-son.

DEDICATED IMPOSTORS: speakers who mimic another speaker's voice (and/or visual appearance). They may be the "entertainer" kind who makes impressions of other people; they may have access to a speaker verification system for training their voice to be similar in the sense of the verification system; or they may somehow artificially generate speech samples to be similar to a target speaker.

- 3J. Occurrence of pairs of particular impostor/target speaker:
- Relatives (same-gender):
- [] twins: pairs
 - [] brothers, sisters, father-son, mother-daughter: pairs
- Dedicated impostors
- [] entertainer kind mimickers: pairs
 - [] trained with speaker recognition system: pairs
 - [] artificially generated: pairs
(method:)
- 3K. Comment on impostor characteristics:
-
-
-
-

4 SPEECH CONTENTS

4.1 LANGUAGE

- 4A. Language(s) spoken:
- 4B. Proportion of subjects that speak their mother tongue:%

4.2 TEXT MATERIAL

Definition: A speech/image file can be used in a text-dependent recognition test if the vocabulary from which test words are drawn is included in some corresponding enrollment material. Such test files are called text-dependent test files. This includes files with test words embedded in carrier phrases.

- 4C. Portion of test files
- [] text-dependent: %
 - [] text-independent: %
- 4D. Specification of text-dependent test files (state typical number of files per session in each category)
- | | fixed | Text-prompted | audio-prompted |
|-----------------|-------------|---------------|----------------|
| digit sequence: | files | files | files |
| single word: | files | files | files |
| sentence: | files | files | files |
- 'fixed' files contain the same text in each session; 'text-prompted' and 'audio-prompted' files contain new text in each session.

4E. Specification of text-independent test files (state typical number of files per session in each category)

Non-spontaneous:

[] text-prompted: files

[] audio-prompted: files

Spontaneous

[] monologue: files

[] answer to question: files

[] conversation: files

5 POST-PROCESSING

5A. What annotation is available for speech data? Specify for how large portion of all speech files a given annotation type is available. 'transcription' refers to a symbolic description and 'segmentation' refers to boundary timing information. Automatic segmentation has not been edited manually.

[] orthographic transcription: %

[] phonemic transcription: %

[] phonetic transcription: %

[] automatic segmentation: %

[] verified segmentation %

5B. What annotation is available for image data:

.....
.....

6 RECORDING EQUIPMENT

6.1 AUDIO RECORDINGS

Specify the type of microphone or telephone handset used in a portion of the total number of calls/sessions. Entries under 6A and 6B together must sum to 100, unless utterances were recorded with multiple devices in a stereo fashion.

6A. Broadband microphone:

[] Head-set %

[] Desktop-mounted microphone %

[] Other broadband microphone %

WG2 FINAL REPORT - APPENDIX A

- 6B. Telephone handset:
 - "Normal" (electret): %
 - Carbon button: %
 - Cordless (not mobile/cellular) % (for instance DECT)
 - Cellular phone, digital % (for instance GSM)
 - Cellular phone, analog % (for instance NMT)
 - Other (specify in 6C) %
 - Unknown telephone %

- 6C. Comment on microphones/handsets:

- 6D. Specify how many speakers made their calls/recordings from how many different phones or microphones:

number of handsets	portion of speakers
<input type="checkbox"/> unknown %
<input type="checkbox"/> 1 %
<input type="checkbox"/> 2-3 %
<input type="checkbox"/> 4-10 %
<input type="checkbox"/> >10 %

- 6E. Telephone recordings were made from:
 - Analog subscriber line
 - Euro-ISDN digital subscriber/trunk line
 - Other digital subscriber or trunk line

- 6F. Microphone recordings were made onto:
 - Analog tape
 - Tape or computer with digital storage (for instance DAT)

6.2 IMAGE RECORDINGS

Please describe recording conditions for the image part of the data. Where possible, try to follow the same outline as for the description of audio recording conditions. (imagine the following analogies: microphone-camera, number of handsets-number of views)

- 6G. Camera equipment:

- 6H. The variety of views (camera angles):

- 6I. Storage (analog video tape, digital direct, etc.):

7 RECORDING ENVIRONMENT

7A. Characterize the caller environment

	% of all sessions
<input type="checkbox"/> anechoic chamber %
<input type="checkbox"/> sound-treated room %
<input type="checkbox"/> quiet home/office %
<input type="checkbox"/> noisy home/office %
<input type="checkbox"/> noisy public place (station, street, etc.) %
<input type="checkbox"/> extreme noise (cockpit, etc.) %
<input type="checkbox"/> unknown %

7B. Information given in 7A is based on

- instructions/conditions given to speaker ("Please call from...")
- answer from speaker ("I called from ...")
- "a posteriori" estimates ("We think he called from ..." :-)

7C. For image data, describe light conditions:

.....
.....

8 PUBLICATIONS

8A. Publications that describe the corpus (conference or journal paper; or web-document):

.....
.....
.....
.....
.....

8B. Publications that describe experiments thoroughly enough to be reproducible by somebody else:

.....
.....
.....
.....
.....

8C. Publications with results produced with this corpus:

.....
.....
.....
.....
.....

9 OTHER INFORMATION

9A. Other important information not covered by the above questions:

.....
.....
.....
.....
.....

9B. Other corpora may contain data that can be used as development data, additional impostors, or to train non-client models. List such corpora and their recommended use:

.....
.....
.....
.....
.....

9C. Major discrepancies between design specification and implementation of this corpus:

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.....
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.....
.....

9D. Things the creators would like to have done differently:

.....
.....
.....
.....
.....

9E. Interesting experiences made during the creation of this corpus:

.....
.....
.....
.....
.....