

Newsletter

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News

MLMI 2008

5th Joint Workshop on
Machine Learning and
Multimodal Interaction

8-10 September 2008
Utrecht, The Netherlands

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Video Editing in Meetings

A COOPERATION OF TECHNISCHE UNIVERSITÄT MÜNCHEN (TUM) AND BRNO UNIVERSITY OF TECHNOLOGY (BUT)

If you think of today's business meetings where a single participant is connected remotely to a meeting room, you have a problem with representing activities in the meeting room to a remote participant. One camera for the room is normally not enough because some of the important details would not be recorded.

If the meeting room is equipped with several cameras, current video conference systems show all video streams to the remote participant. This is useable if someone is sitting in front of a huge screen where all cameras can be shown simultaneously. But does it make sense to watch all the cameras at the same time? This problem is getting worse if a remote participant is using a small screen or even a mobile phone.

important than the current speaker. For this task an event detection system was developed at TUM which recognises events like standing-up, sitting down or taking notes from video and audio sources. These events with additional information from the audio channels are sent to the BUT's editing system.

There, a camera is selected depending on certain rules, which an editor has to take into account when cutting a video. The user has total control over the system, so that it is possible, that only a special participant is shown, or a selected camera is used more often.

The current system can be used for the efficient recording of past meetings, because only one video stream has to be stored. On the other side the system could be used for online video



The video editing system from TUM and BUT solves this problem. All meeting rooms in the AMI corpus are equipped with different cameras. There are close-ups for each participant, as well as some cameras, which are recording a overview of the meeting room.

Depending on what is currently happening in the meeting, the system selects the camera which represents the meeting in the best way. For example, a person who is shaking the head or nodding is possibly more

conferencing. Video editing would help to reduce the work load of the remote participant and the bandwidth of the data connection.

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Cover Story

Hybrid Multi-Step Disfluency Correction

Human speech production and articulation is full of syntactical and grammatical speech errors. The rate of these errors which are often referred to as **disfluencies** typically lies between 5% - 10% of our speech [1] and tend to increase with the speaker's cognitive load [2] (e.g., in business meetings).

While humans are able to filter disfluencies instantly, natural language processing (NLP) systems are often developed for written, grammatical text and thus not able to deal with speech disfluencies. Therefore, an intermediate system that at best should repair or at least mark the disfluent words before the speech gets processed is desirable. An example of an output of such a system is given in the above picture. There, the red marked words are superfluous and can be ignored by the reader while the black words form the speaker's real intended sentence.

This is where GroDi - Get rid of Disfluencies - steps in. GroDi is a hybrid disfluency detection and correction system that has been developed using the classification scheme developed on the AMI corpus (see [3]). The system is able to detect and correct a broad field of different disfluency types, such as repetitions, hesitations, stutterings, etc.

A thorough investigation of the annotated material (so far 40 meetings), showed a heterogeneity of the disfluencies with respect to how the different types can be detected and led to our hybrid detection approach: Easily detectable disfluencies are identified by a simple lexical rule-matching approach while the remaining disfluencies are tackled by sophisticated machine learning techniques. In fact, our system is able to detect more than 56% of

all occurring disfluencies and works faster than real-time. Our next planned step, is to implement a «Hub-version» of GroDi to support the upcoming AMIDA demos with the disfluency correction.

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Use of AMI technology in assessing Clinical Network Meetings

The CSIRO Australian E-Health Research Centre joined the AMIDA project in August 2007. Since then the project team has undergone a transformation and the new team has started on the research related to the recording and analysis of clinical network meetings. Researchers known to the AMIDA team, Iain McCowan and Darren Moore (both previously at IDIAP), have transferred to new labs within CSIRO.

The new research team members are Simon Locke (Post Doctoral Fellow, funded under the Australian Government International Sciences Linkage scheme), Hazel Harden (PhD student, funded by Griffith University and CSIRO) and David Wang (PhD student, funded by Queensland University of Technology and CSIRO) and David Hansen (as project manager).

The new team has started the research of using AMI technologies to support clinical network meetings. Clinical network meetings are used to lead health system change. They involve participants, mainly clinicians, from a variety of backgrounds, addressing a particular issue or need in health care. The meetings, which have shared leadership where organisational structure is irrelevant, are vital to changes in the delivery of health care. Hence, early identification of those teams which are likely to be successful can greatly enhance the final outcome of a project.

Hazel's PhD is focusing on social psychology aspects of team interaction, which will involve manual encoding of the video recording of team meetings. Hazel will be using the methodology of interaction process analysis, a technique for studying interaction in small groups. Coding schemes classify communication acts and deliver positivity/negativity, inquiry/advocacy, and other/self

ratios. Simon will focus on automating this interaction encoding as far as possible from recorded audio, either directly encoding or by looking for proxy signals in the conversation.

The team has purchased eight Apple iPod nano (3rd Gen.) equipped with Belkin TuneTalk amplifier and Sony ECM-C115 lapel microphone to record audio from each individual during meetings, for automatic encoding. As well as helping Hazel to manually encode video, Simon has synchronised the multiple audio streams using mutual information and is examining how to identify who is speaking from the multiple audio streams.



Future work by Simon will involve more complex analysis of the multiple audio streams. David Wang, during his PhD, will be researching speaker segmentation and clustering from a single audio stream, with help from the other team members.

The team will also investigate automated structuring and summarisation of medical concepts in edited speech transcripts using natural language processing techniques.

As well as our new researchers joining the team, the CSIRO Australian e-Health

Research Centre has just moved to its brand new site at the University of Queensland, Royal Brisbane and Women's Hospital. For further information visit: <http://aehrc.com>.

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ACM SIGIR Conference 20-24 JULY 2008, SINGAPORE

SIGIR Conferences

SIGIR is the major international forum for the presentation of new research results and for the demonstration of new systems and techniques in the broad field of information retrieval (IR).

The Conference included:

- Technical Sessions comprising plenary sessions and talks with topics of interest covering original research contributions relating to IR.
- Posters as an opportunity for researchers to present late-breaking and new ideas in a relaxed, interactive setting.
- Panels will consist of discussions on timely and controversial topics.
- Technical Demos will include leading edge work in every area of IR technology and its application.

- State-of-the-Art Tutorials by leading experts will precede the technical program. The full- and half-day offerings will span a wide variety of topics.
- Day-long Workshops on topics of great current interest to members of the IR research community will take place after the technical program.
- The Doctoral Consortium is a venue for doctoral students to present their research and receive feedback from members of the IR research community.

More information at: <http://ilps.science.uva.nl/SSCS2008>

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News and Upcoming Events

MLMI 2008

8-10 September 2008, Utrecht, The Netherlands <http://www.mlmi.info>

MLMI 2008 - 5th Workshop on Machine Learning and Multimodal Interaction

The MLMI series brings together researchers from the different communities working on the common theme of advanced machine learning algorithms applied to multimodal human-human and human-computer interaction.

Several events are also associated to MLMI 2008:

- The AMI Career Day will provide an opportunity for young scientists to talk to representatives of companies working on meeting technology and prepare the next steps of their careers.
- A special session on user requirements and evaluation of multimodal meeting browsers/assistants.
- A workshop on the evaluation of automatic speech recognition systems for Dutch.
- An interproject meeting on the evaluation of space-time audio processing
- A student poster session.

The workshop website has more information about the program, venue and satellite events: <http://www.mlmi.info>.

Looking forward to welcoming you in Utrecht:

- Andrei Popescu-Belis, Idiap research institute (Programme Co-chair)
- Rainer Stiefelhagen, University of Karlsruhe (Programme Co-chair)
- David van Leeuwen, TNO (Organization Chair)
- Anton Nijholt, University of Twente (Special Sessions Chair)



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