Track A: Chandrasekhar Seelamantula, Isabel Trancoso, Mariapaola D'Imperio, Haizhou Li

Track B: Prasanta Kumar Ghosh, Richard Stern, Mathew Magimai Doss, Vikramjit Mitra, Guruprasad S

Welcome tea 0900 hrs Assemble at 0920 hrs
Presentation 20 mins Feedback 10 mins

DURATION	TRACK A.	TRACK B.
(presentation+feedback)	(Room 1)	(Room 2)
0930 hrs - 1000 hrs	Presentation 1A	Presentation 1B
1005 hrs - 1035 hrs	Presentation 2A	Presentation 2B
1040 hrs - 1100 hrs	TEA/COFFEE BREAK	
1100 hrs - 1130 hrs	Presentation 3A	Presentation 3B
1135 hrs - 1205 hrs	Presentation 4A	Presentation 4B
1210 hrs - 1240 hrs	Presentation 5A	Presentation 5B
1245 hrs - 1400 hrs	LUNCH BREAK	
1400 hrs - 1430 hrs	Presentation 6A	Presentation 6B
1445 hrs - 1515 hrs	Presentation 7A	Presentation 7B
1515 hrs - 1540 hrs	TEA/COFFEE BREAK	
1545 hrs - 1615 hrs	Presentation 8A	Presentation 8B
1620 hrs - 1650 hrs	Presentation 9A	Presentation 9B

1A: Zhehuai Chen - Linguistic Search Optimization for Deep Learning Based LVCSR

2A: Rashmi Kethireddy - Acoustic Modelling for Spontaneous Speech Recognition

3A: Karthik Pandia D S - Discovering steady-state and transient regions in speech

4A: Manjunath K E - Study of Multilingual Phone Recognition using Indian Languages

5A: Chiranjeevi Yarra - Temporal analysis of spoken language learning for automatic tutoring

6A: Kamini Sabu - Automatic assessment of children's oral reading for prosodic fluency

7A: Kishore Kumar Botsa - Multiple Emotion Recognition from a Speech Utterance using Deep Neural Architectures

8A: Meredith Moore - Designing Voice-Assistive Technologies: Enhancing the Quality and Intelligibility of Pathological Speech

9A: Nisha Meenakshi Ganesan - Analysis of whispered speech and its conversion to neutral

1B: Madhu Kamble - Energy Separation Algorithm based Features for Replay Spoof Detection

2B: Sushmita Thakallapalli - Unsupervised Techniques for Blind Audio Source Separation

3B: Jesin James - Modelling prosodic features for Empathetic speech of a Healthcare Robot

4B: Nikhil Mohanan - Joint Speech Dereverberation and Denoising Using Constrained NMF

5B: Nirmesh Shah - Voice Conversion Strategies for Parallel and Nonparallel Data Cases

6B: Sishir Kalita - Objective assessment of cleft lip and palate speech intelligibility

7B: Ankur Patil - Automatic Speech Recognition: Low Resource and Noise Robustness

8B: Vikram C M - Analysis and Detection of Stop Consonant Production Errors in Cleft Lip and Palate Speech

9B: Hardik Sailor - Auditory Representation Learning