Raghavendra R Bilgi | ✉ **rrbilgi@gmail.com** | **+91-9663047755** | **LinkedIn**

Summary

- ⋆ Currently working as Chief Engineer at Samsung R&D Institute India, Bangalore
- ⋆ Proficient is Speech Recognition, Deep Learning, Machine Learning techniques
- ⋆ Successfully developed and deployed Speech Recognition solution for S-Voice Personal Assistant
- ⋆ MS from the Indian Institute of Technology, Madras with specialization in Speech Recognition

Experience

Chief Engineer, Intelligent Systems, S-Voice ASR, Dec 2012 - Present
Samsung R&D Institute, Bangalore, India

My primary contribution is in developing commercial quality Speech Recognition system for Samsung S-Voice, Voice Assistant. We have successfully developed and deployed Speech Recognition for two major english locale and currently working on another region specific english locale. I have worked on all the modules of Speech Recognition solution and helped to setup a pipeline for developing and improving speech recognition solution for different locales. I have also worked on developing high quality speech synthesis system by combining unit selection based technologies and statistical parametric synthesis techniques.

- ⋆ Acoustic Modeling : Experience in developing and deploying Deep Neural Network based acoustic model
- ⋆ Language Modeling : Data collection and preparation, model optimization, handling OOVs, personalization of Language model, live data update
- ⋆ Lexicon Preparation : Preparation, development and deployment of lexicons for ASR
- ⋆ Embedded Solution : Model training, model compression and optimization, Confidence scoring module, Decoder optimization
- ⋆ Data Collection : Data preparation, Data collection, Data Evaluation, Co-Ordination
- ⋆ NLU, Dialog Manager : Collaborate with NLU team and ASR Integration
- ⋆ Text to Speech : Added the statistical cost computation and integrated VQ based unit compression.

Teaching Assistant, Electrical Department, Dec 2009 - Nov 2012
Indian Institute of Technology-Madras, Chennai, India

- ⋆ For Courses: Digital Signal Processing (DSP), Analog and Digital Signal Processing, Speech Processing

Software Engineer Jul 2005 - Dec 2009
Hewlett-Packard, Bangalore

I was a member of the team working on Distributed File System targeted for Cloud Computing environment. Primary development was in C, C++, Python on Linux based operating system.

Education

Master of Science (MS) by Research, Department of Electrical Engineering,
Indian Institute of Technology, Madras (IITM), 2012
Thesis: Rapid Speaker and Environment Adaptation in Feature Space for Speech Recognition
Thesis Advisor: Prof. S. Umesh

Bachelor of Engineering (B.E), Department of Electronics and Communication
National Institute of Engineering, Mysore, 2005

Skills

Domain Knowledge	:	Speech Recognition, Text to Speech, Deep Learning, Machine Learning, Natural Language Processing
Programming	:	C, C++, Python, Bash, Scheme
Tools	:	Kaldi, SRILM, OpenFst, CMU-Sphinx, Tensorflow, Keras, Festival, HTS, HTK
Operating Systems	:	Linux, Windows

Publications

- D. S. Pavan Kumar, **R. Bilgi** and S. Umesh, "Non-negative subspace projection during conventional MFCC feature extraction for noise robust speech recognition," Communications (NCC), 2013 National Conference on, New Delhi, India, 2013, pp. 1-5.
- Bharghav. Ch, Neethu. M. Joy, **R. Bilgi** and S. Umesh, "Subspace modeling technique using monophones for speech recognition," Communications (NCC), 2013 National Conference on, New Delhi, India, 2013, pp. 1-5.
- **R. Bilgi**, Vikas Joshi, S. Umesh, G. Luz, B. Carmen, Robust Speech Recognition through the selection of Speaker and Noise transforms- Proceedings of ICASSP 2012, Kyoto, Japan
- V. Joshi, **R. Bilgi**, S. Umesh, G. Luz, B. Carmen, Noise and Speaker Compensation in Log Filter Bank Domain- Proceedings of ICASSP 2012, Kyoto, Japan
- V. Joshi, **R. Bilgi**, S. Umesh, G. Luz, B. Carmen, Sub-band Level Histogram Equalization for Robust Speech Recognition-Proceedings of Interspeech 2011, Florence, Italy
- V. Joshi, **R. Bilgi**, S. Umesh, B. Carmen, G. Luz, Efficient Approach to Speaker and Noise Normalization-Proceedings of Interspeech 2011, Florence, Italy