# Rutuja Ubale

@ rutuja@ucla.edu +1(310)-756-4598 % https://rutujaubale.github.io/ Google Scholar in https://in.linkedin.com/in/rutujaubale • https://github.com/RutujaUbale **EDUCATION** ₩ Sep 2015 - Dec 2016 **m** University of California, Los Angeles GPA: 3.6 / 4.00 Master of Science in Electrical Engineering Advised by Prof. Dr. Abeer Alwan Courses: Speech Processing, Natural Language Processing, Data Science, Machine Learning, Statistical Programming, Big Data: Modeling and Mining the Web, Graphs and Network Flows, Linear Programming **The State of Technology, University of Pune, India** ₩ Jul 2011 - May 2015 Bachelor of Technology in Electronics Engineering GPA: 9.23 / 10.00 Honors in VLSI Design Rank: 3/111 Courses: Digital signal processing, Pattern Recognition, Digital Image Processing, Optimization Techniques, Embedded Systems, Data structures and algorithms, Coding and Data Compression, Computer Programming Deep Learning, Machine Learning, Artificial Intelligence, Speech Processing, Natural Language Processing, Dialog Systems, Multimodal Processing, Computer Vision, Data Science. **WORK EXPERIENCE** ☐ Educational Testing Service (ETS) | Associate Research Engineer (Speech and NLP) Feb 2017 - present Dialog, Multimodal and Speech Research Center (DIAMONDS) San Francisco, CA - Voice and Face Biometrics (Speaker Recognition) https://www.ets.org/accelerate/ai-portfolio/biometric-learner-verification - Spoken Dialogue Systems, Natural Language Understanding https://www.ets.org/accelerate/ai-portfolio/spoken-dialogue-system - Natural Language Understanding of code-switched speech - Automated Scoring of Dialogue and Monologue Responses for Language Learning and Assessment - Deep Learning for Automated Spoken Response Scoring https://www.ets.org/accelerate/ai-portfolio/speechrater - Multimodal Scoring (using speech, text and video) of Presentation videos - End-to-end modeling from raw audio waveforms using Deep Learning - Native Language Identification (Identification of native language of the speaker from his speech in a second language). - Metaphor detection in essays - Automated scoring of scientific visual models for children's science skill assessment. Educational Testing Service (ETS) | Research Intern - NLP **#** Jun 2016 - Sep 2016 San Francisco, CA

Dialog, Multimodal and Speech Research Center (DIAMONDS)

- Spoken Language Understanding for Conversational Dialogue Systems

Speech Processing and Auditory Perception Lab (SPAPL) | Student Researcher University of California Los Angeles (UCLA)

- Noise Robust Speaker Identification in limited training data environments

- Understanding and modeling kids' speech (M.S. Project - Fall 2016)

## **HONORS & AWARDS**

## YFRSW Scholarship at Interspeech 2016

Workshop for Young Female Researchers in Speech Science & Technology Funded by NSF, Microsoft & Google.

 Scholarship recipient, selected to participate at the workshop for women undergraduate and masters students working in speech science and technology at the Interspeech 2016 conference in San Francisco (September 2016).

# **Pune Municipal Corporation Scholarship**

**2009**, 2011

San Francisco, CA

Los Angeles, CA

₩ Sep 2016

Academic excellence in Higher Secondary Certificate and Secondary School Certificate examinations.

## **TECHNICAL SKILLS**

Programming: Python, R, MATLAB, C, C++, JavaScript, SQL, HTML, CSS, Bash

- Tools: Tensorflow, Keras, Kaldi, Apache Spark, WEKA, SKLL, NLTK, Gensim, OpenFace, Voicebox, VoiceSauce, CVX, VMware, MS Office
- Operating Systems: MS Windows, Linux, Mac OS

## **PUBLICATIONS**

- 1. **R. Ubale**, V. Ramanarayanan, Y. Qian, K. Evanini, C. W. Leong, C. M. Lee. "Native language identification from raw waveforms using deep convolutional neural networks with attentive pooling." in (to appear) Proceedings of 2019 IEEE Workshop on Automatic Speech Recognition and Understanding (ASRU), Dec. 2019. IEEE.
- 2. Y. Qian, **R. Ubale**, P. Lange, K. Evanini. F. Soong, "Spoken Language Understanding of Human-Machine Conversations for Language Learning Applications." in *Journal of Signal Processing Systems (accepted for publication November 2019)*.
- 3. C. W. Leong, K. Roohr, V. Ramanarayanan, M. P. Martin-Raugh, H. Kell, **R. Ubale**, Y. Qian, Z. Mladineo, and L. McCulla. "To Trust, or Not to Trust? A Study of Human Bias in Automated Video Interview Assessments." in (to appear) Workshop on Interpreting and Explaining Visual Artificial Intelligence Models, International Conference on Computer Vision (ICCV 2019), Nov. 2019. IEEE.
- 4. C. W. Leong, K. Roohr, V. Ramanarayanan, M. P. Martin-Raugh, H. Kell, R. Ubale, Y. Qian, Z. Mladineo, and L. McCulla. "Are Humans Biased in Assessment of Video Interviews?" in Adjunct of the 2019 International Conference on Multimodal Interaction (ICMI '19). ACM, New York, NY, USA, Article 9, 5 pages.
- 5. Y. Qian, P. Lange, K. Evanini, R. Pugh, **R. Ubale**, M. Mulholland, and X. Wang. "Neural approaches to automated speech scoring of monologue and dialogue responses." in *Proceedings of The 44th IEEE International Conference on Acoustics, Speech, and Signal Processing (ICASSP)*, 2019. IEEE.
- 6. **R. Ubale**, Y. Qian, K. Evanini. "Exploring end-to-end attention-based neural networks for native language identification." in *Proceedings of the IEEE Workshop on Spoken Language Technology (SLT 2018)*, pp. 84-91, 2018. IEEE.
- 7. Y. Qian, R. Ubale, M. Mulholland, K. Evanini, X. Wang. "A prompt-aware neural network approach to content-based scoring of non-native spontaneous speech." in *Proceedings of the IEEE Workshop on Spoken Language Technology (SLT 2018)*, pp. 979-986, 2018. IEEE.
- 8. Y. Qian, **R. Ubale**, P. Lange, K. Evanini. "From Speech Signals to Semantics Tagging Performance at Acoustic, Phonetic and Word Levels." in *Proceedings of the 11th International Symposium on Chinese Spoken Language Processing (ISCSLP 2018)*, 2018. IEEE.
- 9. Z. Ni, **R. Ubale**, Y. Qian, M. Mandel, S. Yoon, A. Misra, D. Suendermann-Oeft. "Unusable Spoken Response Detection with BLSTM Neural Networks." in *Proceedings of the 11th International Symposium on Chinese Spoken Language Processing (ISCSLP 2018)*, 2018. IEEE.
- 10. V. Ramanarayanan, D. Pautler, P. Lange, E. Tsuprun, **R. Ubale**, K. Evanini, and D. Suendermann-Oeft. "Toward Scalable Dialog Technology for Conversational Language Learning: A Case Study of the TOEFL® MOOC." in *Proceedings of Interspeech* 2018, pp. 1960-1961, 2018.
- 11. K. Evanini, M. Mulholland, R. Ubale, Y. Qian, R. Pugh, V. Ramanarayanan, and A. Cahill. "Improvements to an Automated Content Scoring System for Spoken CALL Responses: The ETS Submission to the Second Spoken CALL Shared Task." in *Proceedings of Interspeech* 2018, pp. 2379-238, 2018.
- 12. C. W. Leong, L. Liu, **R. Ubale**, and L. Chen. "Toward large-scale automated scoring of scientific visual models." In *Proceedings* of the Fifth Annual ACM Conference on Learning at Scale, 2018.
- 13. Y. Qian, **R. Ubale**, V. Ramanarayanan, P. Lange, D. Suendermann-Oeft, K. Evanini, and E. Tsuprun, "Exploring ASR-free end-to-end modeling to improve spoken language understanding in a cloud-based dialog system," in *Proceedings of 2017 IEEE Workshop on Automatic Speech Recognition and Understanding (ASRU)*, Dec. 2017. IEEE.
- 14. Y. Qian, K. Evanini, P. L. Lange, R. A. Pugh, **R. Ubale**, F.K. Soong, "Improving native language (L1) identification with better VAD and TDNN trained separately on native and non-native English corpora," in *Proceedings of 2017 IEEE Workshop on Automatic Speech Recognition and Understanding (ASRU)*, Dec. 2017. IEEE.

15. Y. Qian, **R. Ubale**, V. Ramanarayanan, P.L. Lange, D. Suendermann-Oeft, K. Evanini and E. Tsuprun, "Towards End-to-End Modeling of Spoken Language Understanding in a Cloud-based Spoken Dialog System," in *Proceedings of SEMDIAL 2017 (SaarDial) Workshop on the Semantics and Pragmatics of Dialogue (pp. 160-161).* 

## **PATENTS**

C. W. Leong, L. Liu, **R. Ubale**, and L. Chen. Platform for automated scoring of scientific visual models (US Patent Pending, filed 2019).

Y. Qian, **R. Ubale**, V. Ramanarayanan, P. Lange, D. Suendermann-Oeft, K. Evanini, and E. Tsuprun. Exploring ASR-free end-to-end modeling to improve spoken language understanding in a cloud-based dialog system (US Patent Pending, filed 2018).

Y. Qian, K. Evanini, P. L. Lange, R. A. Pugh, **R. Ubale**. Native Language Identification with Time Delay Deep Neural Networks trained separately on native and non-native English corpora (US Patent Pending, filed 2018).

## **PROFESSIONAL ACTIVITIES**

Reviewer for Association for Computational Linguistics, 2019.

Reviewer for North American Chapter of the Association for Computational Linguistics: Human Language Technologies (NAACL-HLT), 2019.

Reviewer for IEEE Spoken Language Technology Workshop (SLT), 2018.

Reviewer for Interspeech, 2018.

Reviewer for 20th ACM International Conference on Multimodal Interaction (ICMI), 2018, 2019.

## **TALKS AND PRESENTATIONS**

## **Educational Testing Service Research** | Intern Presenter

Statistical framework for real time implementation of the Spoken Language Understanding component in the HALEF spoken dialog system

**Interspeech 2016** | Workshop for Young Female Researchers in Speech Science & Technology Drunk-Text Detection

## **THESIS**

## Accent Correction System Based On Feedback Stimuli Generation

🛗 Jan 2014 - Apr 2015

Developed an accent correction system in which non-native English speaker's speech is corrected by identifying deviations in the person's current speech from the desired accent using PSOLA and Regression Analysis.

[Tools: MATLAB]

## **OPEN SOURCE CONTRIBUTIONS**

halef-SETU - Python Package

% https://pypi.python.org/pypi/halef-SETU/

Statistical Engine for Text Understanding (SETU)

% https://sourceforge.net/p/halef/halef-SETU

halef-SETU provides an easy wrapper around SKLL models for statistical language understanding as well as an easy to use API based on Flask

## **SELECTED COURSE PROJECTS**

## Area function construction of children's /a/ vowel from 3D ultrasounds

Sep 2016 - Nov 2016

The goal of this project is to facilitate the understanding of children's production of the vowel /a/ through computation of area function - a measure that is representative of a speaker's vowel or consonant production through volumetric imaging.

[Tools: MATLAB]

## **Noise Robust Speaker Identification System**

May 2016 - Jun 2016

Built a system using cepstral features, normalization and filtering techniques to achieve robustness to noise conditions. [Tools: MATLAB, VoiceBox, VoiceSauce]

## **Drunk-Text Detection**

**Mar** 2016 - Mar 2016

Given a Tweet, the goal is to identify if it was written under the influence of alcohol or not. Performed text-based analysis to determine level of inebriation. Multiclass Support Vector Machine (SVM) classifier is trained using stylistic, LDA and n-gram features to detect drunk tweets.

[Tools: Python, Scikit-learn, PyEnchant, NLTK, Tweepy, Re, Pandas, Numpy]

## Acoustic correlates to speaker identity

## Feb 2016 - Mar 2016

Analyzed the performance of different acoustic features - formants, spectral slope parameters, voice source features, MFCC, LPCC and DTW using SVM and KNN classifiers for predicting perceptual dissimilarity for a pair of sounds and determining whether a pair of speech sounds are from the same speaker or not.

[Tools: MATLAB, VoiceSauce]

# Identifying patterns of brain activity in larval zebrafish

₩ Jul 2015 - Aug 2015

Implemented distributed Principal Component Analysis to identify patterns across time-varying images.

[Tools: Python, MLlib]

# **MISCELLANEOUS**

Languages: English – Fluent, Marathi – Native proficiency, Hindi – Native proficiency, French – Basic proficiency (speak, read, write with basic competence), Spanish - Basic proficiency (speak, read, write with basic competence)