

Mahsa Elyasi (Mahsa Sadat Elyasi Langarani)

E-mail: elyasila@ohsu.edu

Website: mahsae.github.io

RESEARCH INTERESTS

Machine learning and its applications in language and speech processing.

EDUCATION

Ph.D., Computer Science and Engineering

Center for Spoken Language Understanding, OHSU, Portland, OR, expected 2020

M.Sc., Computer Engineering, Artificial Intelligence

Sharif University of Technology, Tehran, IRAN, September 2012

B.Sc., Computer Engineering, Software Engineering

University College of Nabi Akram, Tabriz, IRAN, September 2010

POSITIONS

BioSpeech Inc., Portland, OR

March 2020 - Present

Speech Research contractor

- Word boundary alignment for three types of speech: normal, fast, and slow. Performed a word boundary detection using automatic pause detection, speech to text method followed by automatic fine tuning of word boundaries. Compared to baseline, our method improves F_1 measure from 63 to 95.

ObEN Inc., Pasadena, CA

Spring 2017

Speech Research Intern

- TTS adaptation: Developed an intonation adaptation model to transform the perceived identity of a DNN-based TTS (Merlin) system to that of a target speaker. Resulting in higher similarity compared to baseline.
- TTS: developed a discrete cosine transform intonation model for generating fundamental frequency (F_0) contour in Mandarin. Achieving higher naturalness compared to rule-based approach.
- Data preparation: Developed a script to speed up the labeling process and error checking which saved one week for the team.

ObEN Inc., Pasadena, CA

Summer 2016

Speech Research Intern

- Speech-To-Music: developed a bidirectional LSTM syllabification method to generate syllable boundaries given speech signal and phonetic labels. Achieving F_1 measure of 95.
- TTS: developed a superpositional intonation model for English. Achieving higher naturalness compared to intonation generated with HTS.

Sensory, Inc., Portland, OR

Fall 2015

Speech Research Intern

- Implemented intent classification to determine the intent expressed in a given query for an NLU system. Trained logistic regression in one-vs-rest setting. Achieved on average F_1 measure of 85.

Center for Spoken Language Processing, OHSU, Portland, OR 2012 - Present

Graduate Research Assistant

- Experimenting effect of fundamental frequency (F_0) contour in speaker group classification using combination of CNN and RNN. 1D CNN was used in lower layers in a deep architecture to capture supra-segmental patterns of the F_0 contour. Applying a Recurrent layer on top of the Convolutional helped the

model to capture these temporal pattern. In a balanced data set, achieved an average accuracy of 91

- Proposed a quantitative model for analysis and synthesis of English intonation. Developed and invented various Machine learning method to examine the performance and potential of the proposed intonation model in a variety of speech processing applications, including TTS, TTS adaptation, and speaker group classification.

PUBLICATIONS **M.S. Elyasi Langarani**, J. van Santen, Investigating prosodic unit effects of fundamental frequency dynamics in clear and conversational speech (under submission).
M.S. Elyasi Langarani, J. van Santen, Prosody based dialect classification using NMF and sparsity criteria (under submission).
M.S. Elyasi Langarani, J. van Santen, Recurrent Convolutional Neural Network for Classification of Speaker Groups based on Prosodic Information, 12th Women in Machine Learning Workshop (WiML), 2017.
M.S. Elyasi Langarani, J. van Santen, Automatic, model-based detection of pauseless phrase boundaries from fundamental frequency and duration features, 9th ISCA Speech Synthesis Workshop, 2016.
M.S. Elyasi Langarani, J. van Santen, Foot-based Intonation for Text-to-Speech Synthesis using Neural Networks, Speech Prosody 2016.
M.S. Elyasi Langarani, J. van Santen, Speaker Intonation Adaptation for Transforming Text-To-Speech Synthesis Speaker Identity, ASRU 2015.
M.S. Elyasi Langarani, J. van Santen, S.H. Mohammadi, A. Kain, Data-driven Foot-based Intonation Generator for Text-to-Speech Synthesis, Interspeech 2015.
M.S. Elyasi Langarani, J. van Santen, Modeling fundamental frequency dynamic-sin hypokinetic dysarthria, SLT 2014.
M.S. Elyasi Langarani, E. Klabbers, J. van Santen, A Novel Pitch Decomposition method for the Generalized Linear Alignment Model, ICASSP 2014.
M.S. Elyasi Langarani, H. Veisi, H. Sameti: The effect of phase information in speech enhancement and speech recognition. ISSPA 2012.
S.H. Mohammadi, H. Sameti, **M.S. Elyasi Langarani**, , A. Tavanaei, KNNDIST: A Non-parametric distance measure for speaker segmentation, Interspeech 2012.

ACTIVITIES **Reviewer**, Interspeech, WiML, Speech prosody
Organizing Committee, Volunteer at Interspeech 2012 conference.
Student Member, ISCA, IEEE Signal Processing Society
Award
Top student in bachelor of Science class (GPA: 3.83)
Nominated by OHSU for a HHMI fellowship 2015

COMPUTER SKILLS *Languages:* Python, R, C++.
ML Toolkits: Keras, TensorFlow.
Speech Toolkits: Merlin, Festival, HTS, Praat, TextGrid.
Academic skills: Data cleaning, Amazon mechanical turk, A/B testing, Error analysis, Statistical reliability.

REFERENCES **Jan van Santen**, Professor, Center for Spoken Language Understanding, Oregon Health and Science University, vansantj@ohsu.edu.
Alexander Kain, Associate Professor, Center for Spoken Language Understanding, Oregon Health and Science University, kaina@ohsu.edu.
Abeer Alwan, Professor and Vice Chair, Electrical and Computer Engineering, University of California, Los Angeles (UCLA) alwan@ee.ucla.edu .