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Objective_

An applied research & software development role in audio, speech and natural language technologies

Skills

Languages Python, C/C++, Objective-C, bash, LaTeX, Java, Scala, MATLAB

DSP Fourier Analysis, IIR/FIR Filter Design, MIR features, segmentation, source separation, ARM NEON SIMD **Speech** Training & maintenance of custom LVCSR models in Kaldi (DNN-HMM, LF-MMI), with n-gram LMs

Machine Learning DNN, RNN & seq2seg modeling in TF, Keras & PyTorch, clustering & classification, scikit-learn, PCA/ICA/NMF

Audio Solid knowledge of audio engines, codecs, effects, mixing, psychoacoustics, audio production & post-production

Audio Frameworks LibROSA, numpy, CoreAudio, AudioUnit, Stagefright, Audioflinger Infrastructure Horizontally scaling Kaldi decoders in AWS EC2. Apache {Kafka, Spark}

Intangibles I ship! Fast learner with strong engineering communication, team work ethic & emotional intelligence

Experience_

Marchex Seattle, WA

STAFF ENGINEER Sept. 2014 - Nov. 2018

- Algorithm design, prototyping, implementation, performance analysis & optimization
- Develop audio-dsp algorithms for language & telephony SPAM identification
- Construct corpora for conversational speech recognition. Design automatic data selection algorithms
- · Train, deploy & maintain custom DNN-HMM models in Kaldi. Experiment with novel end-to-end ASR architectures
- Perform 24/7 on-call support of fleet processing 1M calls/day. Extensive experience with AWS EC2 {LB, ASG, EFS, EBS}

Dolby Laboratories San Francisco, CA

STAFF AUDIO APPLICATIONS ENGINEER Jan. 2012 - Aug. 2014

- · Worked on embedded systems implementations/ports of Dolby's perceptual audio coding & audio post-processing
- · Algorithms: MDCT, spectral extension, hybrid complex QMF, dialog, bass & volume leveling, surround virtualization
- Extensive experience conducting acoustic measurements & critical listening
- · Mentored junior engineers & regularly lectured on Dolby's audio-dsp algorithms to APAC FAE staff

Apple Cupertino, CA

SENIOR AUDIO SOFTWARE ENGINEER

• 2006 - 2009: Developed system level audio & effect processing features for Soundtrack Pro v2, Final Cut Studio 2 & 3

- Integrated, fixed bugs and maintained audio-unit hosting code for Logic Pro audio effects within ST Pro, Final Cut Pro X
- 2010-2011: Developed & shipped an audio auto-synch tool & a multi-channel audio file configuration classifier for FCP X
- · Collaborated with the CoreAudio team on optimizations to their resampler & audio analysis algorithms

2KSports San Rafael, CA

AUDIO SOFTWARE ENGINEER

- Developed C/C++ tools & libraries for the XBOX360 launch of NBA2k6, NHL2k6, College Hoops 2K6
- Performed support tasks to game development teams using the audio library

Education ___

Dartmouth College Hanover, NH

A.M. IN MUSIC COMPOSITION (ELECTRO-ACOUSTIC MUSIC)

Carnegie Mellon University

Sept. 1999 - June 2001

Feb. 2006 - Nov. 2011

May 2005 - Jan. 2006

Interdisciplinary Masters in 20th century Composition, Music Cognition, Aesthetics, Acoustics & Signal Processing

B.Sc. IN ELECTRICAL AND COMPUTER ENGINEERING, WITH A DOUBLE MAJOR IN MATHEMATICS/COMPUTER SCIENCE

Pittsburgh, PA Sept. 1993 - May 1997

École polytechnique fédérale de Lausanne (EPFL)

3RD YEAR EXCHANGE STUDENT IN COMPUTER AND COMMUNICATION SCIENCES

Lausanne, Switzerland

July 1995 - August 1996

Publications

2018	The Marchex 2018 English Conversational Telephone Speech Recognition System, an	rXiv
2010	The Marchex 2010 English Conversational relephone Speech Recognition System, at	$ \wedge $

- 2018 Attentive Seq2Seq Learning for Diacritic Restoration of Yorùbá Text, Accepted for Interspeech 2018
- 2017 Audio Spectrogram Factorization for Classification of Telephony Signals, arXiv
- 2017 Semi-Supervised Model Training for Unbounded Conversational Speech Recognition, arXiv
- 2009 Introduction to Digital Signal Processing (review), Computer Music Journal
- 2001 Riddim: A Rhythm Analysis Tool Based On Independent Subspace Analysis, MA Thesis

Patents Issued

2015	US 9,485,354, Identifying call features and associations to detect call traffic SPAM	Marchex
2015	US 8,929,561, System and Method for Automated Audio Mix Equalization	Apple
2014	US 8,842,842, Detection of audio channel configuration	Apple
2013	US 8,621,355, Automatic synchronization of media clips	Apple

Extracurricular

Institute of Electrical and Electronics Engineers (IEEE), Audio Engineering Society (AES), International Speech and Communication Association (ISCA)

Member 2012 - PRESENT

Niger-Volta Language Technologies Institute

FOUNDER OF OPEN-SOURCE WEST AFRICAN LANGUAGE PROJECT 2018 - PRESENT

Unicode Common Locale Data Repository (CLDR) Project

Yorùbá language contributor 2007 - 2008

The Society for Electro-Acoustic Music in the United States (SEAMUS)

Managing Editor for Journal SEAMUS 2008-2012

DC Productions (Independent record label)

Owner, songwriter, recording artist, producer, remixer, A&R, staff writer 2003-2012

Natural Languages ____

French

Full Professional Proficiency, written & spoken

- University of Lausanne coursework in Literature & Film
- ILR proficiency of 4+

Yorùbá

CONVERSATIONAL PROFICIENCY

• Junior secondary school reading/writing level

Spanish

BASIC PROFICIENCY

• Fulfill traveling needs & conduct myself in a polite manner