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## Objective\_

An applied research & software development role in audio, speech or natural language processing

## Skills

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

**DSP** Fourier Analysis, IIR/FIR Filter Design, segmentation, MIR features, source separation, ARM NEON SIMD Speech Training LVCSR models in Kaldi (DNN-HMM, LF-MMI), n-gram LMs, building labeling & model scoring tools

Machine Learning DNN, RNN & seq2seq modeling in Tensorflow, Keras & PyTorch, clustering & classification, PCA/ICA/NMF **Audio** Solid knowledge of audio codecs, effects, mixing, psychoacoustics, audio production & post-production

Audio Frameworks LibROSA, CoreAudio, AudioUnit, Stagefright, Audioflinger

Infrastructure Horizontally scaling fleets of Kaldi decoders in AWS EC2, Load Balancers, Auto-Scaling & Target groups Intangibles I ship! Fast learner with strong engineering communication, team work ethic & emotional intelligence

# **Experience**

Marchex Seattle, WA

Sept. 2014 - PRESENT STAFF ENGINEER

- · Applied research (from prototype to production) in narrowband conversational speech recognition
- · Audio DSP algorithm development for language & Telephony SPAM identification
- Acoustic & Language model corpus preparation. Labeling tooling, scoring & data pipeline development
- Training of custom DNN-HMM models in Kaldi, Experiments with E2E ASR
- AWS EC2 {LB, ASG, EFS, EBS}. 24/7 on-call support of fleet processing 1M calls/day

**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 included: Hybrid complex QMF, MDCT, spectral extension, Adaptive hybrid transform (AHT)
- Dialogue clarity, surround virtualization, volume leveling & bass enhancement
- · 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 tools for Soundtrack Pro v2, Final Cut Studio 2 & 3.

- · Integrated & maintained Logic Pro audio effects within the STP Audio-Unit hosting environment.
- · Collaborated with core-audio team on optimizations to their resampler and audio analysis algorithms
- 2010-2011: Developed & shipped an auto-synch feature for ingested media & an audio file configuration classifier for Final Cut X
- Integrated & maintained custom 64-Bit versions of Logic Pro audio effects within the FCP X Audio-Unit hosting environment

2KSports San Rafael, CA

AUDIO SOFTWARE ENGINEER

May 2005 - Jan 2006

Feb. 2006 - Nov. 2011

- 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)

Sept. 1999 - June. 2001

· Interdisciplinary Masters in 20th century Music Composition, Cognition, Aesthetics, Audio Engineering & Music DSP

**Carnegie Mellon University** 

Pittsburgh, PA

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

Sept. 1993 - May. 1997

#### École polytechnique fédérale de Lausanne (EPFL)

Lausanne, Switzerland

3RD YEAR EXCHANGE STUDENT IN COMPUTER AND COMMUNICATION SCIENCES

July 1995 - August 1996

## **Publications**

2018 Attentive Sequence-to-Sequence Learning for Diacritic Restoration of Yoruba Text. IN LERSPE	2018	Attentive Sequence-to-Sequence Learning for Diacritic Restoration of Yorùbá Text, II	NTERSPEECH
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- 2018 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)

Member Jan 2012 - PRESENT

#### Niger-Volta Language Technologies Institute

FOUNDER OF OPEN-SOURCE WEST AFRICAN LANGUAGE TECHNOLOGY 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 of Journal SEAMUS 2008 - 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