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Objective_

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

Skills

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

DSP Fourier Analysis, IIR/FIR Filter Design, MIR features, segmentation, audio source separation, ARM NEON SIMD Training & maintenance of spoken language-id & LVCSR models in Kaldi (DNN-HMM, LF-MMI) with n-gram LMs Speech Machine Learning DNN, RNN & seq2seq modeling in TF, Keras & PyTorch, clustering & classification, scikit-learn, PCA/ICA/NMF

> Audio Extensive knowledge of audio engines, codecs, psychoacoustics, effects, mixing, film-audio production & post PyTorch, Torchaudio, librosa, numpy, ffmpeg, CoreAudio, AudioUnit, Stagefright, Audioflinger

Audio Frameworks

Infrastructure Horizontally scaling Kaldi decoders in AWS EC2. Apache {Kafka, Spark}, Spinnaker, Huggingface. Misc ML Ops. Intangibles I ship! Fast learner with strong engineering communication, team work ethic & emotional intelligence

Experience

Netflix Seattle, WA

SENIOR SOFTWARE ENGINEER - AUDIO ALGORITHMS

Nov. 2018 - Present

- Algorithm design for audio content analysis & understanding. Research, prototyping & product-ization
- · Training & maintaining Audio ML models for scaled Post-production & Globalization tasks
- Construction of large-scale audio-video training datasets

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

- · 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

Cupertino, CA

SENIOR AUDIO SOFTWARE ENGINEER

Feb. 2006 - Nov. 2011

Jan. 2012 - Aug. 2014

- 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

May 2005 - Jan. 2006

- 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 Composition, Music Cognition, Aesthetics, Acoustics & Signal Processing

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

Selected Recent Publications _

2024 DJ Tool Retrieval Using Speech Activity, Music Str	ucture And CLAP Embeddings, ISMIR LBD 2024
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- 2024 Leveraging counterfactual cross-modal pairs for audiovisual representation learning, CVPR 2024
- 2024 **Ìròyìnspeech: A multi-purpose yorùbá speech corpus**, LREC-COLING 2024
- 2023 A generalized bandsplit neural network for cinematic audio source separation, IEEE OJSP-ICASSP 2023
- 2022 BibleTTS: a large, high-fidelity, multilingual, and uniquely African speech corpus, Interspeech 2022
- 2022 A large TV dataset for speech and music activity detection, EURASIP-JASMP 2022
- 2020 Masakhane Machine Translation For Africa, AfricanNLP Workshop (ICLR 2020)
- 2018 Attentive Seq2Seq Learning for Diacritic Restoration of Yorùbá Text, Interspeech 2018
- 2018 The Marchex 2018 English Conversational Telephone Speech Recognition System, arXiv
- 2017 Semi-Supervised Model Training for Unbounded Conversational Speech Recognition, arXiv

Patents

2023	US 18/505,081 , Methods and systems for learning language-invariant audiovisual representations	Netflix
2020	US 10,810,995 B2, Automatic speech recognition (ASR) model training	Marchex
2019	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 TECHNOLOGY ORGANIZATION

2018 - PRESENT

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