

ASR – Automatic Speech Recognition for European Portuguese with the Kaldi Framework

Mauricio Breternitz
Miguel Sales Dias
Pedro Santos
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Outline

Introduction

ASR – Automatic Speech Recognition

The Kaldi ASR Toolkit

European Portuguese ASR Development

Experimental Set UP, Results

Future Work

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Digital Living Spaces

- ✓ Ambient Assisted Living
- ✓ Smart cities e smart buildings
- ✓ Mobilidade sustentável
- ✓ Realidade virtual e aumentada para edifícios
- ✓ Sustentabilidade energética
- ✓ Projeto participativo e fabricação digital
- ✓ Análise do espaço, percecão e uso
- Projeto de espaços, do património às práticas vernaculars e às gramáticas da forma

21 researchers 5 PhD Students



Software Systems Engineering

- ✓ Engenharia de software empírica
- ✓ Cibersegurança, Gestão e Proteção de ativos digitais
- ✓ Big Data e Visual Analytics
- ✓ Gestão de fluxo de trabalho científico
- ✓ Governância das tecnologias da informação
- ✓ Smart cities e smart buildings
- ✓ Smart grids
- ✓ BPM Business Process Management

20 researchers
3 PhD Students



Information Systems

- ✓ Teorias e modelos de sistemas de informação
- Business Intelligence & Analytics
- ✓ Business Process Management
- ✓ IT Governance & Management
- ✓ E-Government
- ✓ IS aplicada a e-learning, gamification, marketing, hospitalidade e tourismo

31 researchers 11 PhD Students



Complexity and Computational Modelling

- ✓ Análise multi critério e teoria da decisão
- ✓ Geração automática de desenho/projeto
- ✓ Gramáticas da forma genéricas
- ✓ Estruturas de significado em redes de comunicação
- ✓ Análise de fenómenos sociais complexos

13 researchers 2 PhD Students



Brief BIO, Publications, Patents

PhD – Carnegie-Mellon, ECE

MSc – UNICAMP/Brazil

BSc – ITA-Brazil

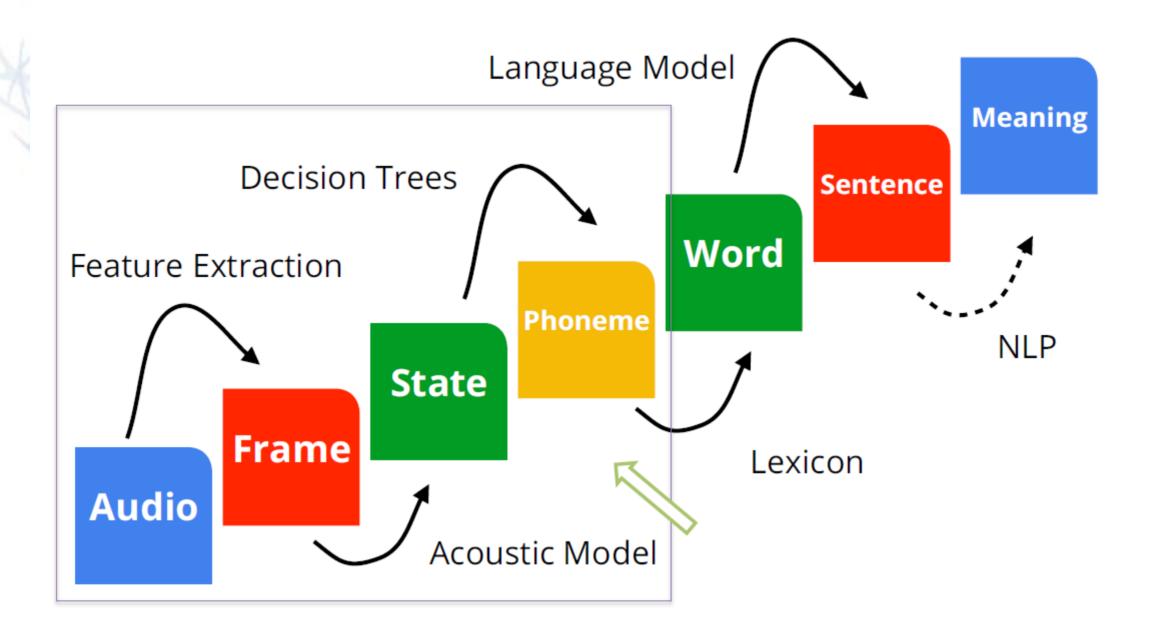
Work: IBM Research, Motorola, Times N, Intel Labs, AMD Research 48 U.S. Patents Issued, 54 U.S. Patents Pending

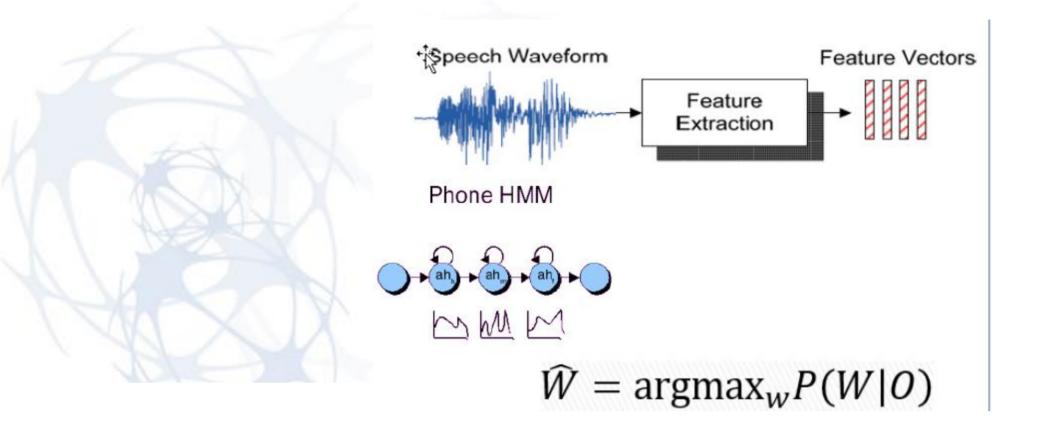
Publications

Citations 1587 H-index 24, i10-index 39
Computer Architecture, Computer Systems, Performance, Tuning
Big Data, Machine Learning

Creator / General Chair: International Workshop on Architectural / Microarchitectural Support for Binary Translation, joint with ISCA (ACM/IEEE International Symposium on Computer Architecture) and CGO.

Automated Speech Recognition





Automatic Speech Recognition

Find the most likely sentence (word sequence) W, which transcribes the speech audio A: $\widehat{W} = \underset{W}{\operatorname{argmax}} P(W|A) = \underset{W}{\operatorname{argmax}} P(A|W)P(W)$

- Acoustic model P(A|W)
- Language model P(W)

Training: find parameters for acoustic and language model separately

- Speech Corpus: speech waveform and human-annotated transcriptions
- Language model: with extra data (prefer daily expressions corpus for spontaneous speech)



KALDI ASR Framework

- State-of-the-art ASR framework
- Open-Source, Forum
- DNN Framework
- Components
 - processing a speech waveform,
 - language models and
 - transcriptions

EUROPEAN PORTUGUESE ASR Development:

- 1. Simple vocabulary (spoken digits)
 - Develop, tune scripts
- European Portuguese(progressively larger)
 corpus 4 versions of 154,558 utterances
 total 618,232 utterances
 - 800+ hours of transcribed speech
 - Adapted 'Fisher-English' -> LusoFisher
- 3. Second training and experiment
 - combination of all available utterances
 - total of 589,540 unique utterances.

Experimental Setup

- CPU
 - -Intel(R) Xeon(R) CPU E5-2660
 - 16 cores –
 - 16Gb memory –
 - 2Tbyte disk storage
- GPU: NVIDIA Corporation Titan XP (*)
 Software: CUDA 8.0
- Operating System: Ubuntu 16.04.03 LTS

(*)The authors acknowledge equipment donation by NVIDIA academic program and by Advanced Micro Devices

Experimental Corpus

CORPUS	Number of Utterances	
Corpus Subset	1000	
SpeechCorpus1	154558	
Ensemble Dataset	588162	
SpeechCorpus1 Data Augmented (4x)	618232	

Corpus Input Example

e vai lá estar toda a gente a ver se nós nos encontramos com todos

viagem se leva muito muito tempo se não pronto

Words (size 148249):

abafaneticamente 190 abafante 191 abafar 192 abafação 193 abafeira 194 abafo 195 abafura 196 abafável 197 abaganhado 198 abaganhar 199

•••

Text:

SA001EP2 maria inês pedroso SA001EP3 rosmaninho SA001EQ1 não SA001ET1 às sete SA001ET2 às oito SA001F01 olá inês é a célia que está a falar estou te a mandar esta mensagem a ver comigo a uma a uma festa logo à noite que é da da patrícia é os anos da patrícia e

Lexicon:

<UNK> SPN
a a
a aex
í i
à a
à-direita a d i dx aex j t aex
à-distância a d i sh t aexn s j aex
à-esquerda a sh k e dx d aex
àgua a g w aex

. . .

olá boa tarde o meu nome é célia aleixo e eu gostaria de saber qual é que

da partida do para do voo para para as canárias e se há neste mês se não há ou se e

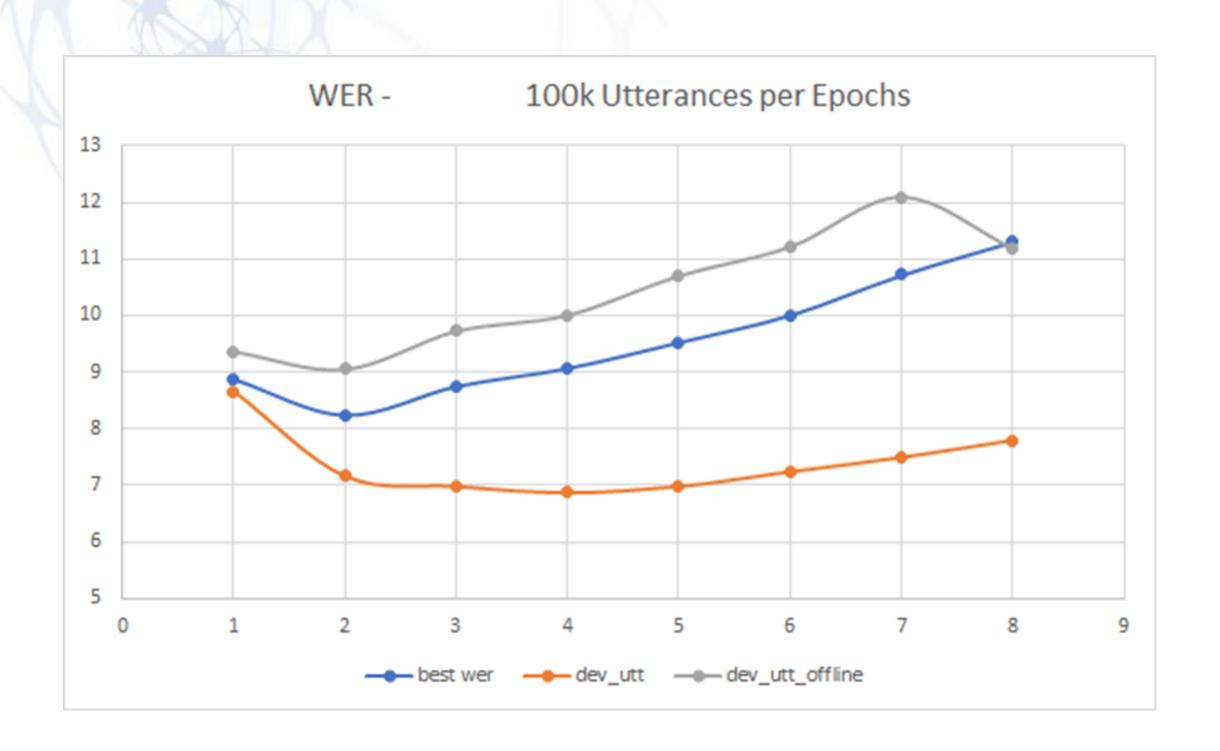
Training – Computational Requirements

	Total Elapsed	
	Processing	
CORPUS	Time(hours)	
SpeechCorpus1 Subset	12.5	
SpeechCorpus1	194	
SpeechCorpus1 Data		
Augmented (4x)	620	

Trained Model Size

CORPUS	Trained Model Size (Mbytes)		
SpeechCorpus1 Subset	47		
SpeechCorpus1	48		
Ensemble Dataset	97		
SpeechCorpus1 Data			
Augmented (4x)	91		

Training vs OverFitting



Results-WER

11 38 11 12 1		
CORPUS	DNN Best WER	Traditional Best WER
SpeechCorpus1 Subset	10.46	12.75
SpeechCorpus1	11.77	15.61
Ensemble Dataset	10.12	13.17
SpeechCorpus1 Data Augmented (4x)	15.99	18.23

<<Demo>>

Kaldi -> Library



Conclusion & Future Work

End-to-End ASR

- Research topics:
 - Full-stack: HW, SW, Application, System
 - Cross-layer optimization
 - Societal Applications



education



work





personal























End-To-End ASR

End-to-End Speech Recognition From the Raw Waveform

Neil Zeghidour^{1,2}, Nicolas Usunier¹, Gabriel Synnaeve¹, Ronan Collobert¹, Emmanuel Dupoux²

¹ Facebook A.I. Research, Paris, France; New York & Menlo Park, USA
² CoML, ENS/CNRS/EHESS/INRIA/PSL Research University, Paris, France {neilz, usunier, gab, locronan}@fb.com, emmanuel.dupoux@gmail.com

STATE-OF-THE-ART SPEECH RECOGNITION WITH SEQUENCE-TO-SEQUENCE MODELS

Chung-Cheng Chiu, Tara N. Sainath, Yonghui Wu, Rohit Prabhavalkar, Patrick Nguyen, Zhifeng Chen, Anjuli Kannan, Ron J. Weiss, Kanishka Rao, Katya Gonina, Navdeep Jaitly, Bo Li, Jan Chorowski, Michiel Bacchiani

Google, USA

{chungchengc,tsainath,yonghui,prabhavalkar,drpng,zhifengc,anjulironw,kanishkarao,kgonina,ndjaitly,boboli,chorowski,michiel}@google.com

ABSTRACT

Attention-based encoder-decoder architectures such as Listen, Attend, and Spell (LAS), subsume the acoustic, pronunciation and language model components of a traditional automatic speech recognition

on a large vocabulary continuous speech recognition (LVCSR) task. The goal of this paper is to explore various structure and optimization improvements to allow sequence-to-sequence models to significantly outperform a conventional ASR system on a voice search task.



Site ISTAR

http://istar.iscte-iul.pt/

Ciência-IUL ISTAR

https://ciencia.iscte-iul.pt/centres/istar-iul







Key Machine Learning Frameworks

CAFFE Torch Tensorflow

> Theano Keras



ML Services

Google ML API

Prediction API https://cloud.google.com/prediction/docs/

MICROSOFT COGNITIVE SERVICES

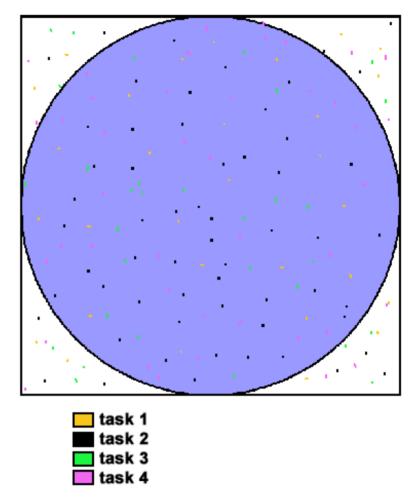
APIs https://www.microsoft.com/cognitive-services/en-us/apis
Vision, Speech, Language, Knowledge, Search



MapReduce in 30 seconds: Estimating Pl

- 1- pick random points in unit rectangle
- 2- count fraction inside circle

Area: $\pi/4$



Map-Reduce:

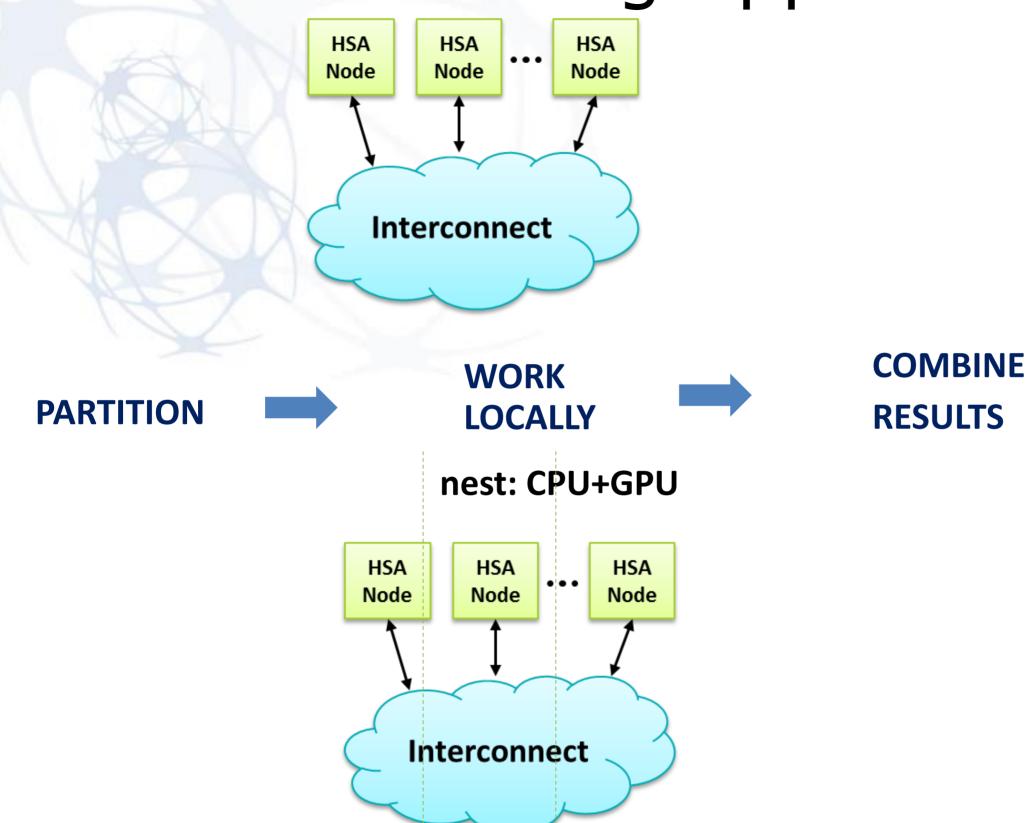
Map: random point inside? Issue k=1, v=1 else k=0,v=1

Reduce: count 0 keys and count 1 keys

Programmer: writes { map, reduce } methods, system does rest



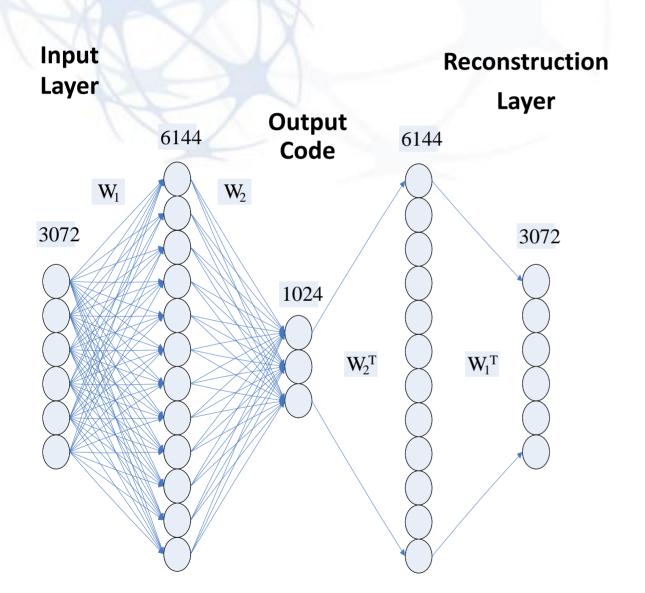
Nested Processing Approach

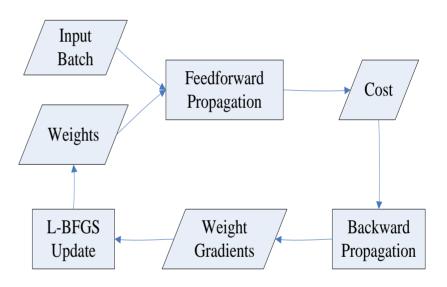




DNN Autoencoder

- Autoencoder + L-BFGS training (used by Google)
 - Encode the input and then reconstruct
 - A mix of CPU compute with GPU compute
 - Heavy CPU and GPU data interaction- good match for HSA





L-BFGS Training Algorithm

End to End Speech Recognition

Feature extraction stage

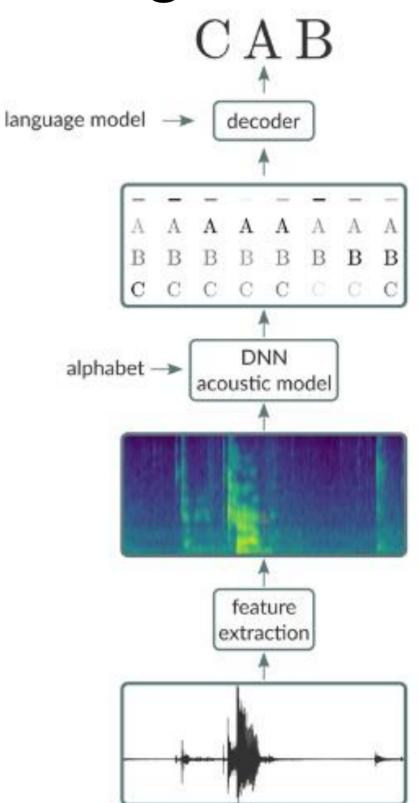
takes raw audio signals (from a wav file) and generates a sequence of feature vectors, with one feature vector for a given frame of audio input.

Acoustic model stage

takes sequences of feature vectors and generates probabilities of either character or phoneme sequences conditioned on the feature vector input.

Decoder stage

takes two inputs – the acoustic model's outputs as well as a language model – and searches for the most likely transcript given the sequences generated by the acoustic model constrained by the linguistic rules encoded in the language model.



TYPES OF MACHINE LEARNING

- Supervised (inductive) learning
 - Training data includes desired output e.g. Spectrum of supervision images are labeled with categories "face" "face" "dog "car" ...

Unsupervised learning

 Training data does not include desired the system autonomously sorts similar images From left to right, an increasing amount of information is

and classifies data

supplied to help classify the images into categories:

No classification

supplied for each image

Images classified in broad categories: human faces, koalas, cars

Detailed Categorization: parts of the image are Labeled and highlighted

Semi-supervised learning

- Training data includes a few desired outputs:

tem generalizes from labeled examples

MACHINE LEARNING: APPLICATION FRAMEWORK across MULTIPLE REGIONAL SCOPES

Point of Data Capture



Local/Regional Data Center

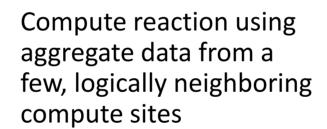


Scale-Out Data Center



Compute reaction nearby the place data is captured, using locally stored knowledge

Examples - Car warning system using front camera image; Factory Sensor Input



Example: Traffic management and scheduling of a city-wide fleet

Compute reaction using knowledge from a worldwide and long term memory

Example: Netflix movie recommendation engine

Exascale Technology Benefits

- "Big data is what happened when the cost of storing information became less than the cost of *making the decision* to throw it away."
 - George Dyson

- Big Data and Analytics
 - Machine Learning at Exascale
 - Commercial Applications
- High Performance Computing Systems
 - Memory
 - Storage
 - Communication

The National Strategic Computing Initiative* and DOE's role

Through Executive Order 13702 he signed July 29, 2015, President Obama established the National Strategic Computing Initiative (NSCI) to maximize the benefits of HPC for US economic competitiveness and scientific discovery.

DOE is a lead agency within NSCI with the responsibility that the DOE Office of Science and DOE National Nuclear Security Administration will execute a joint program focused on advanced simulation through a capable exascale computing program emphasizing sustained performance on relevant applications.

^{*} https://www.whitehouse.gov/the-press-office/2015/07/29/executive-order-creating-national-strategic-computing-initiative



ACADEMIC COLLABORATIONS

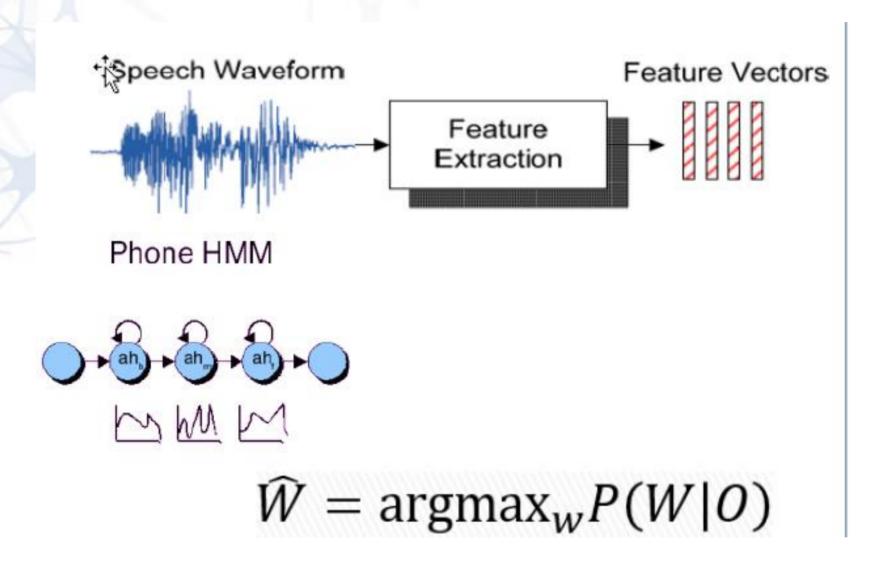
- DIVIDEND Distributed Heterogeneous Vertically IntegrateD ENergy Efficient Data centres
 - Marie Curie CHIST-ERA: Edinburgh, Belfast, Lancaster, EPFL, Timisoara, AMD, INRIA
 - Euros 2M
 - http://www.chistera.eu/projects/dividend
- EUPHONIC ITN 2018 Submission
- Pesquisador Visitante Especial UNICAMP
- Invited Lecturer:
 - U.Texas/Austin, Rice University, CMU-Silicon Valley

ASR System Overview

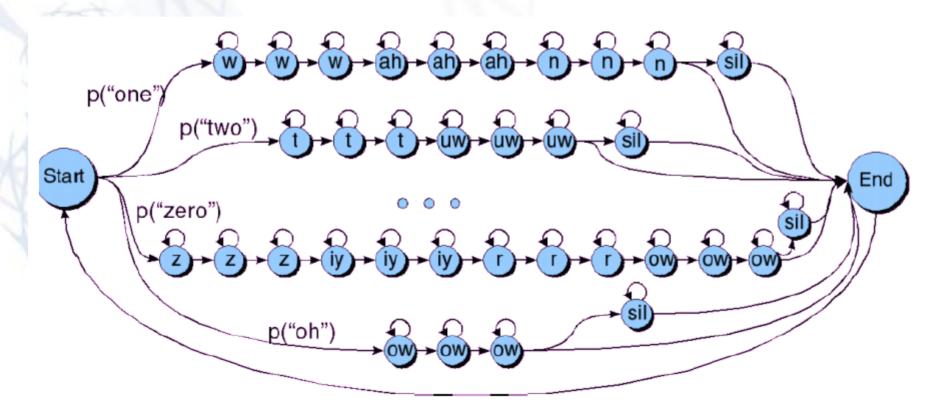
Courtesy of Bin Ma (2015)

ASR In Brief

Lexicon



ASR In Brief



- □ P(W|O) = P(O|W)P(W): P(O|W) probability of a feature sequence given a word sequence, called acoustic model (AM), P(W) word language model (LM).
- ■Each word is decomposed into phonemes according to a lexicon, and each phone is modeled by a 3-state left-right Hidden Markov Model.
- □ Conventionally, the HMM state emission probability P(o|s) is modeled by Gaussian Mixture Models (GMMs).
- □LM is usually a N-gram.











