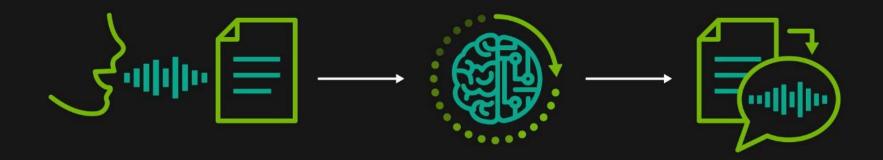
Nemo and Rasa Application in a Chatbot Environment

Natural Language Processing Project Andrea Gurioli, Giovanni Pietrucci e Mario Sessa



Objectives



Automatic Speech Recognition

Natural Language Processing

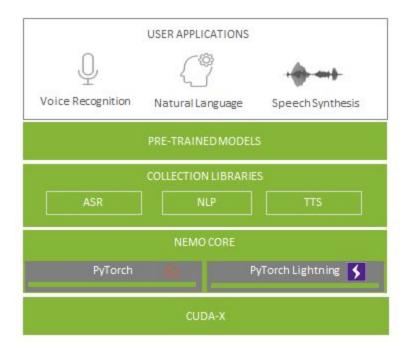
Text to Speech

NeMo Application Stack

NeMo provides different modules to build pre-trained models using collections libraries.

NeMo models are built on PyTorch and PyTorch Lightning libraries

It is possible to improves performances with the GPU-based library CUDA-X.



Automatic Speech Recognition Pipeline

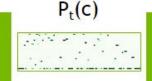


Preprocessing

- Data augmentation
 - speed perturbation (resampling)
 - · additive Gaussian noise
 - time / frequency masks (cutout)
- · Windowing 20-25 ms, stride 10 ms
- · FFT, log, mel scale
- Normalization

spectrogram

Neural Acoustic Model



training

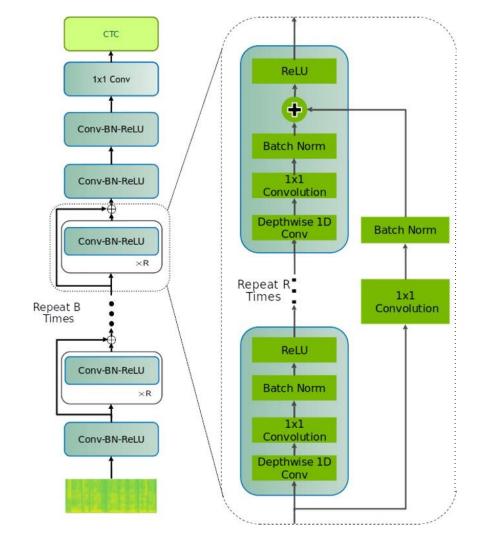
CTC Loss

- Jasper
- Wav2Letter+
- DeepSpeech 2

inference

Decoder

- greedy (argmax)
- beam search
- beam search with language model rescoring



Quartznet

Small WER and high transcription speed. It is based on Jasper model.

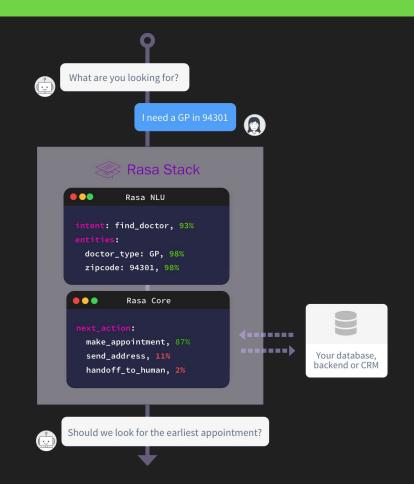
Its composition consists in 1
Convolutional layer followed by
a sequence on N blocks
repeated Z times

Each block has a depthwise 1D
Convolutional, Pointwise
Convolutional and a
Normalization layer with a ReLu
activation function

Rasa Framework

Rasa is an open-source conversational AI tools formed by two modules:

- **1. Rasa NLU:** It consists in the intent classification and entity extrapolation
- 2. Rasa Core: ML-based dialogue management to predict appropriate actions from an intent trigger.

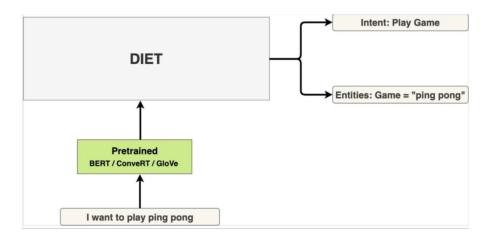


During intent predictions and entity extractions, Rasa NLU executes a pipeline based on question tokenization, tokens featurizing and output prediction.

We tested a Sparse (BoW) and Dense (Word embedding) vector on a DIET Classifier which uses an inner CRF Entities Extractor.

During the analysis, we saw better performances on the chosen Sparse vector usage.

NLP Pipeline



Vectors	F1	Recall	Precision
Sparse vectors	0.92 (±0.05)	0.90 (±0.03)	0.90 (±0.06)
Dense vectors	$0.89(\pm 0.04)$	0.91 (±0.03)	$0.90(\pm 0.05)$

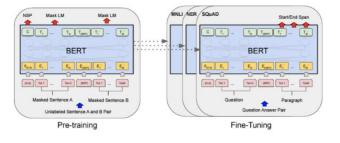
Alysia Tasks

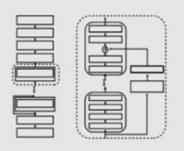
- To-Do List Management
- Weather System
- Jokes System

- Sending Emails
- Time Service
- WikipediaSearch

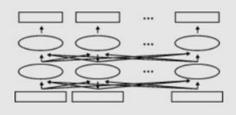
Wikipedia Search System uses the QAModel provided by NeMo NLP based on a BERT-based configuration and trained on the SQuAD dataset.



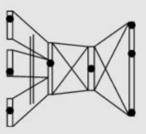




Acoustic models

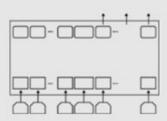


Decoders

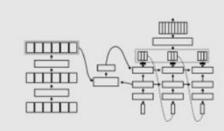


Language models

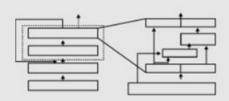




BERT models

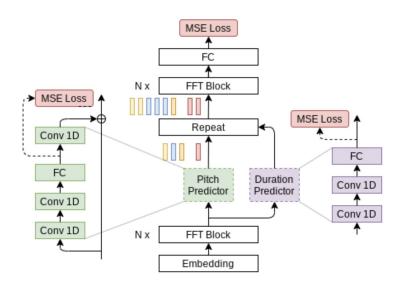


Speech Synthesis models



Voice encoder models

Spectrogram Generator



FastPitch

It's the chosen model for the spectrogram generator.

FastPitch is a fully-parallel text-to-speech model based on FastSpeech with higher real-time factor than Tacotron2 for the mel-spectrogram synthesis of an utterance.

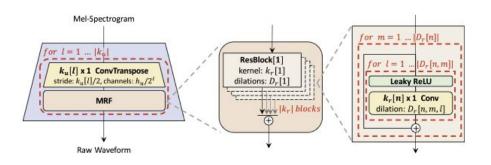
It processes letters and predicts pitches and durations of mel-spectrogram elements using a parallel approach.

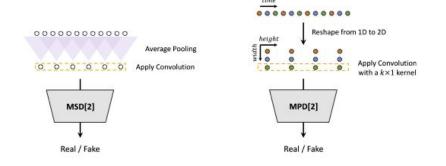
HiFiGAN

GAN-based model for audio generation from a mel-spectrogram.

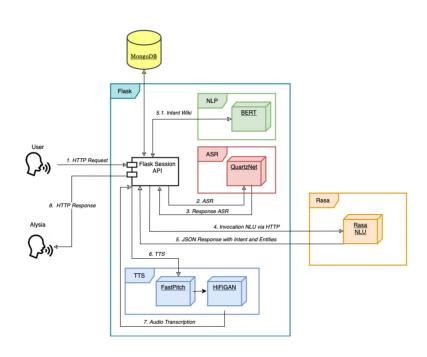
It is composed by a generator and two discriminators: multi-scale and multi-period discrimnators.

HiFiGAN defines one of the state-of-art speech synthesis model with optimal performances in synthesis speed suitable for real-time applications like Alysia

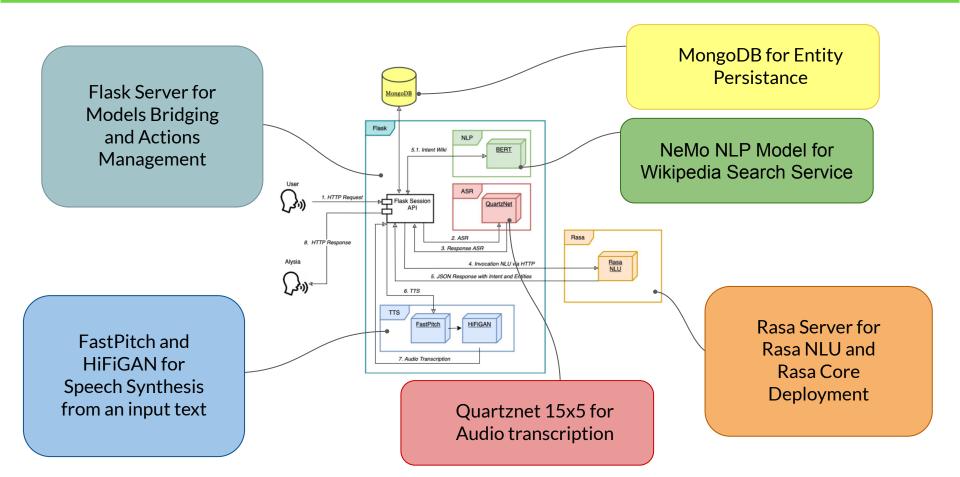




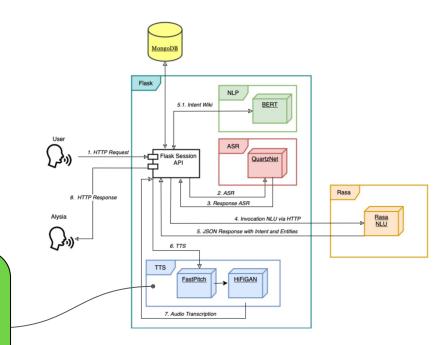
Alysia Deployment Diagram



Alysia Deployment Diagram



Alysia Deployment Diagram



Important Note:

For Performances problems, we replace it with the Speech Synthesis Model from Web Speech API

Conclusions

 NVIDIA NeMo is adaptive, easy to use and with high abstraction for many applications.



- It guarantee state-of-art models in pre-trained form available in its collection libraries.
- It can integrates external services like MongoDB or other databases for entity persistence and intent predictions or entity extractions like Rasa and Google Dialogflow.
- NeMo framework obtains the high performances in TPU or GPU-based running environments but it works on CPU-only host too.



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