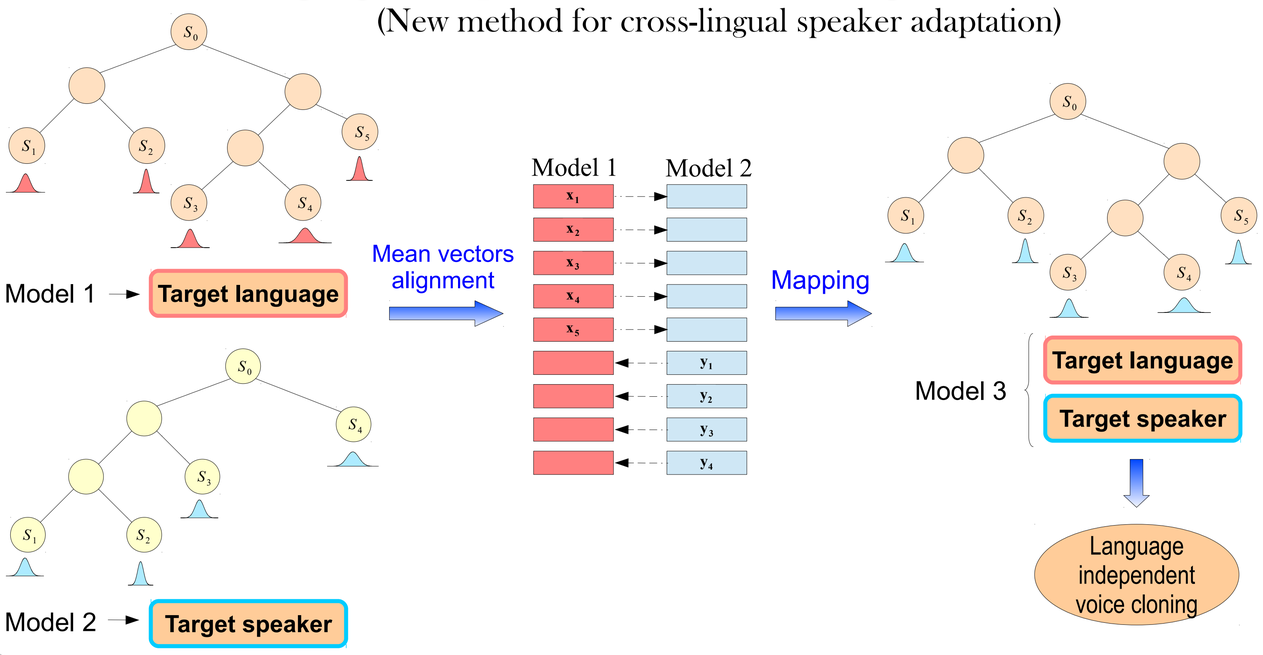
IMPROVEMENTS IN HMM-BASED AND UNIT-SELECTION SPEECH SYNTHESIS TECHNIQUES

Carmen Magariños Iglesias and Eduardo Rodríguez Banga (Advisor)
Multimedia Technology Group (GTM), University of Vigo

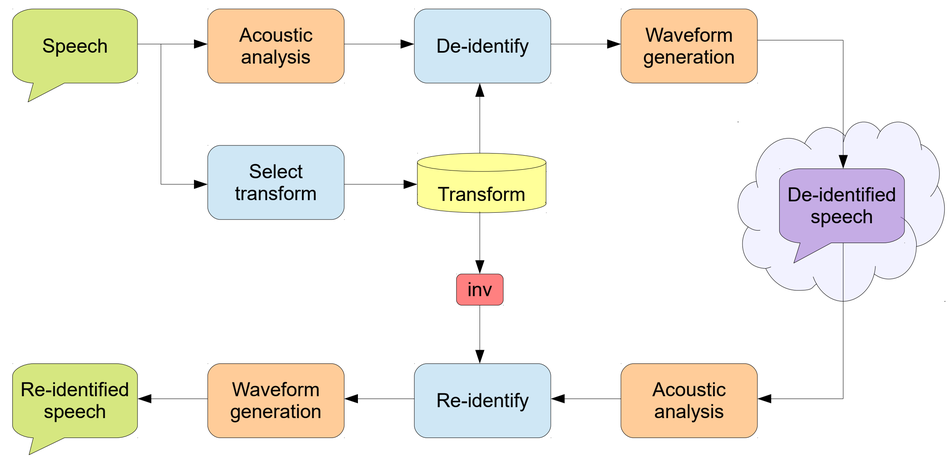
Motivation of the work

Language-independent acoustic cloning of HTS[1] voices [1]

(New method for cross-lingual speaker adaptation)



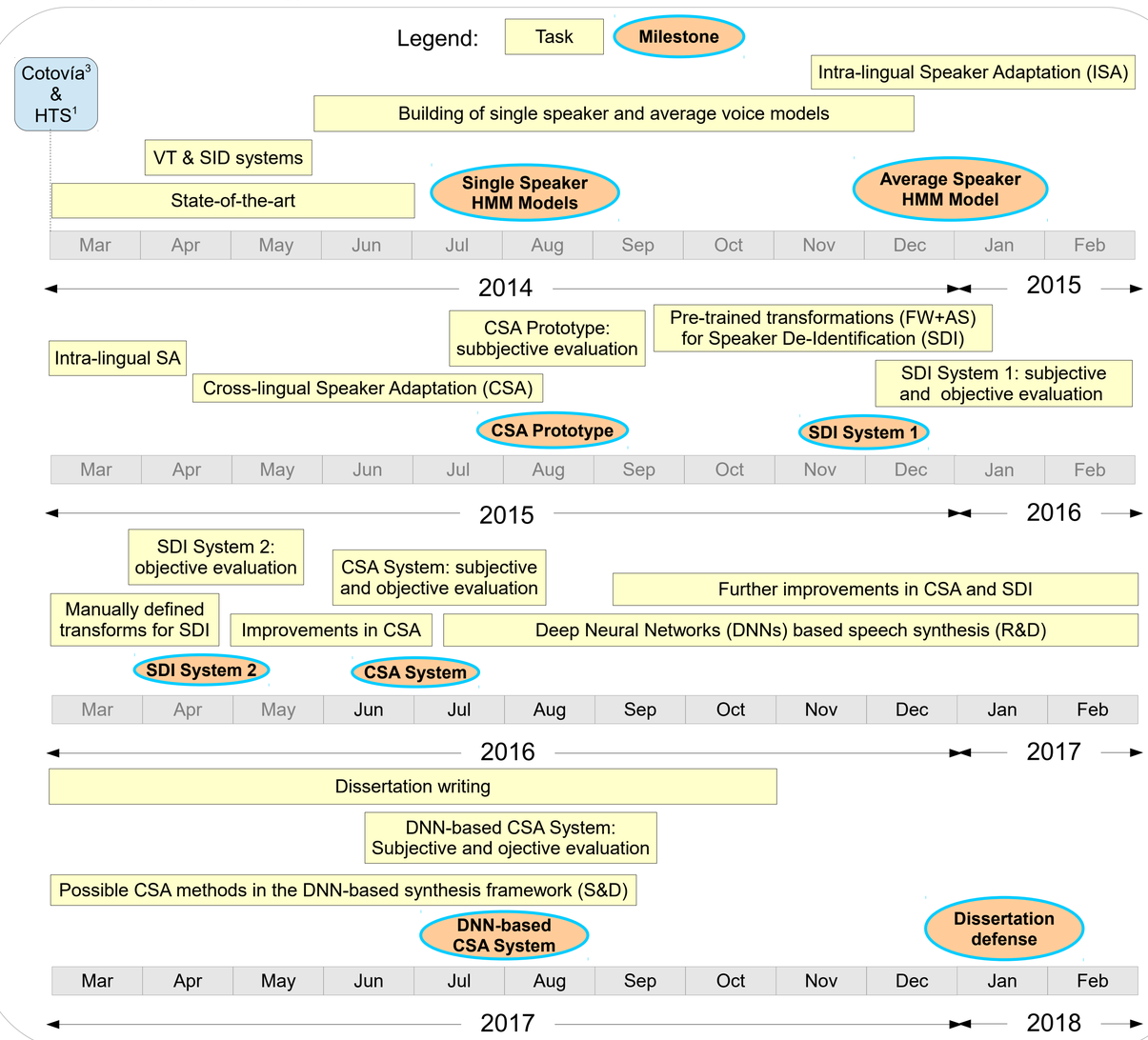
Speaker de/re-identification using voice transformation functions



Thesis objectives

- Analysis of **state-of-the-art techniques** for speech synthesis [2], covering speaker adaptation (SA) methods.
- Propose **improvements** to existing techniques, including more **efficient systems** for specific applications with memory or computational load restrictions.
- Apply **intra-lingual speaker adaptation techniques** [3] to increase the flexibility of the speech synthesis systems (larger number of speakers, speaking styles and emotions).
- Study, development and implementation of **cross-lingual speaker adaptation techniques** [4] with the aim of obtaining multilingual speakers (speech-to-speech translation).
- Analysis of different voice transformation (VT) techniques [5] and application in the field of **speaker de-identification**.
- Use of speech synthesis techniques in related applications, such as the **robustness evaluation** of Speaker Identification (SID) Systems. [6]

Research Plan



Next Year Planning

- **Cross-lingual speaker adaptation**
  - Improvements in the adaptation method (final version of the cross-lingual adaptation system).
  - Subjective and objective evaluations (MOS tests and speaker identification system).
- **Speaker de-identification**
  - Further improvements and evaluation.
- **DNN-based speech synthesis**
  - Exploration of DNN-based speech synthesis approaches.
  - Research on possible cross-lingual adaptation methods for DNN-based synthesis.
- **Conference/Journal publications**
  - Coming journal submission: improved cross-lingual adaptation system.

Acknowledgements

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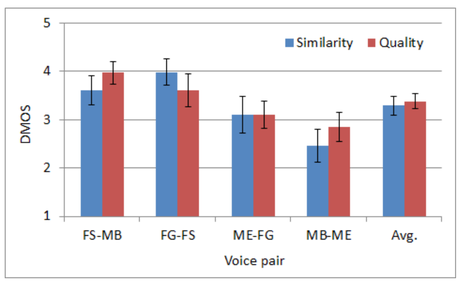
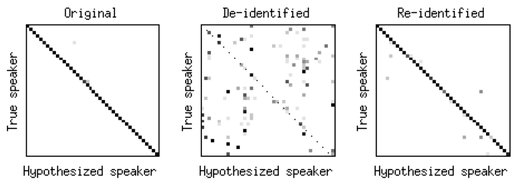
Results & Discussions

- **Improvements in HMM-based speech synthesis**
  - Ahocoder [7] integration: higher quality than previous vocoder.
- **Intra-lingual speaker adaptation**
  - Average voice model (AVM) for Spanish using Albayzín database.
  - Inclusion of the Galician language in the "Zure TTS" platform[2] [8].
- **New method for cross-lingual speaker adaptation** [1]
  - Language-independent acoustic cloning of HTS[1] voices.
  - Adaptation method based on INCA algorithm [9].
  - Examples at http://goo.gl/FwemL4.
- **Speaker de-identification using voice transformation functions**
  - Pre-trained transformations based on the FW+AS technique [10] (SDI System 1).
  - Manually defined transformations using piecewise linear approximation of FW functions (SDI System 2) [11].
- **Subjective evaluations**
  - Perceptual listening tests.
  - Differential mean opinion score (DMOS).
- **Objective evaluations**
  - Speaker identification system as objective measure.
  - State-of-the-art i-vector approach combined with dot-scoring.
- **Conference/Journal publications**
  - eNTERFACE 2014 [8], Interspeech 2015 [11], ICASSP 2016 [1], SPLINE 2016 (accepted) [12], IEEE Signal Processing Letters (submitted).

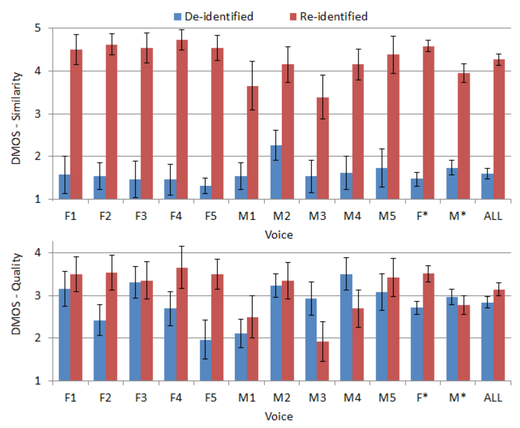


Proposed **cross-lingual adaptation** method: DMOS results.



Identification accuracy on original speech	99.2%
De-identification accuracy	89.5%
Re-identification accuracy	94.2%

Proposed **SDI System 1**: confusion matrices and results in terms of accuracy for original, de-identified and re-identified speech.



Proposed **SDI System 1**: DMOS results for similarity and quality.

Transformation	FW	FW+F0	FW+F0+AS
Trans1	82.5%	98.6%	96.9%
Trans2	53.9%	87.2%	88.1%
Trans3	30.6%	64.2%	68.3%
Trans4	4.4%	28.0%	36.7%

Proposed **SDI System 2**: speaker de-identification results in terms of accuracy for the different transformations.

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[1] http://hts.sp.nitech.ac.jp/, [2] http://aholab.ehu.eus/zuretts/, [3] http://sourceforge.net/projects/cotovia/