MRI vocal tract data during CV production

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Objectives
The main purpose of the presented algorithm is to enlarge MRI speech corpus by synthesising data. In this work, we use a method [2] that captures the dynamics of speech during CV production by using non-rigid image transformations. This information is adapted and then applied to a target speaker in order to synthesise its CV production data using only his silence frame. We evaluated the performance of the proposed method using image cross-correlation in which we compared the original images of the target speaker pronouncing the same CVs with synthesised images.

Materials and Methods
Recordings of two French subjects (one male, one female) in the supine position were performed in a 3T MRI (Prisma, Siemens, Erlangen, Germany) with a 20-channel head and neck antenna. An echo-dispersed, Cartesian TI-weighted echo gradient sequence VIBE (Volumetric Interpolated Breath-hold Examination) was used for the 3D recordings while for the 2D real time recordings we used radial RF-spoiled FLASH sequence.

12 CV syllables (combination of C=[/l, /p, /s, /t]/ and V=[/i, /a, /u, /i]) were used. The chosen planes were the midsagittal (M) its left (L) and right (R) adjacent planes. Data in each plane were acquired in different acquisition on the same session.

A non-rigid image transformation method [1] was used, based on the displacement field between the images. To measure the image similarity, histogram matching between the images is applied and then the mean square error of the pixels intensity is computed. To validate the results, cross-correlation between the synthesised and the original images, normalized by the autocorrelation of the original images, was used.

Algorithm Description
The input of the algorithm is a silence frame of both train and target speaker and the rMRI data of the target CV. An image transformation is computed from each CV frame to the next one, creating a set of transformations that describe the dynamics of the CV production. Another image transformation is computed from the silence frame of train speaker to the silence frame of the target speaker and is used to adapt the set of transformations computed previously to the target speaker. The adapted set of transformations is applied to the silence frame of the target speaker to synthesise his/her CV pseudo rMRI data. Synthesised images are compared with the original ones using image cross-correlation. Results show good agreement between the synthesised and the original images.

Results
In Fig. 1 we can see chosen images during /pi/ in both synthesised and the corresponding original form. Synthesised images have average match of 94.37% ± 0.96% with the original ones over the set of syllables studied, using normalized image cross-correlation. By visually inspecting them, we can see that some different appears at the back part of the tongue, which is a little flatter. Additionally, lip protrusion is weaker on the upper lip and some artifacts appear sometimes at the level of the epiglottis. Apart from this, images look quite similar in terms of vocal tract shape with an exception is a few cases were a small artifact may appear mainly at the region of the tongue due to the existence of a similar artifact in the corresponding training images. However, there are also cases that synthesised images had less artifacts and were smoother compared to the original ones.

Discussion and conclusion
Some differences can be observed in the images mainly due to different articulation and vocal tract shapes. Additionally, the algorithm uses the same phoneme duration as the train speaker since it does not take into account any known information about the speaking style of the test speaker. Further research could include learning this registration more globally, or by using DNN learning techniques or other ways to implement the duration model.

References