

Dubbing in Practice: A Large Scale Study of Human Localization With Insights for Automatic Dubbing

William Brannon*
MIT Media Lab, USA
wbrannon@mit.edu

Yogesh Virkar
AWS AI Labs, USA
yvvirkar@amazon.com

Brian Thompson†
AWS AI Labs, USA
brianjt@amazon.com

Abstract

We investigate how humans perform the task of dubbing video content from one language into another, leveraging a novel corpus of 319.57 hours of video from 54 professionally produced titles. This is the first such large-scale study we are aware of. The results challenge a number of assumptions commonly made in both qualitative literature on human dubbing and machine-learning literature on automatic dubbing, arguing for the importance of vocal naturalness and translation quality over commonly emphasized isometric (character length) and lip-sync constraints, and for a more qualified view of the importance of isochronic (timing) constraints. We also find substantial influence of the source-side audio on human dubs through channels other than the words of the translation, pointing to the need for research on ways to preserve speech characteristics, as well as transfer of semantic properties such as emphasis and emotion, in automatic dubbing systems.

1 Introduction

Considerable attention has been paid to the dubbing of video content from one language into another, both in the literature of several disciplines and in the daily practice of the entertainment industry. One influential line of work, in the fields of film studies and audiovisual translation, studies human dubbing from a qualitative perspective (Chaume, 2012; Zabalbeascoa, 1997, 2008; Freddi and Pavesi 2009), as a profession and semiotic activity. This literature has developed a rich body of theory on the nature of the human dubbing task, and the ways humans approach it, but has little engagement with large-scale data. More recently, machine-learning practitioners have taken up the task of building multimodal systems for automatic dubbing (e.g., Saboo and Baumann, 2019; Federico et al., 2020a; Tam et al., 2022), but lack

deep empirical or theoretical bases for how to organize their work.

What is missing from both literatures, and can help bridge the gap between them, is a large-scale study of human dubbing in practice: a data-driven examination of the way humans actually perform this task. Such an analysis can have benefits for both the qualitative study of human dubbing, by providing empirical evidence of how dubbing teams approach their work, and informing future machine-learning work on automatic dubbing systems. It is exactly this analysis we undertake in this work.

Human dubbing involves a sequence of human contributors each with control over a different aspect of the process (Chiaro, 2008a,b; Matamala, 2010; Chaume, 2012). The first step is an approximately literal translation of the original script, done by a dialogue translator. Next, a dialogue adaptor will modify this translation into a plausible script meeting the requirements for dubbing such as isochrony, lip sync, kinesic synchrony, and so on. Finally, the translated and modified script will go to a production team. Voice actors, with input from a dubbing director or supervisor, have been noted to often have freedom to improvise or make small changes to the dialogue as it is being recorded (Paolinelli and Di Fortunato, 2009; Chiaro, 2008b; Matamala, 2010).

We, however, aim to understand human dubbing by studying not its process, but its product: a large set of actual dubbed dialogues from real TV shows, obtained from Amazon Studios. As compared to qualitative work or interviews with dubbers, this approach has the particular virtue of capturing tacit knowledge brought to bear in the human dubbing process but difficult to write down or explain.

We organize our investigation around one of the most fundamental insights from the qualitative literature, that of human dubbing (and subtitling, which we do not consider here) as “constrained

*Work conducted during an internship at Amazon.

†Corresponding author.

translation’’ (Titford, 1982; Mayoral et al., 1988). A dub, after all, is not just a translation of the original content—indeed, it is not a purely textual product at all. As a translation, it should preserve the meaning of the original; as spoken language, it should sound natural; as an accompaniment to a video track, it should fit with the timing of actors’ mouth movements, body language, and the flow of the story (Chaume, 2020).

Simultaneously satisfying all of these constraints is very difficult, and in general may not be possible. We are accordingly interested in how human dubbers balance the competing interests of semantic fidelity, natural speech, timing constraints, and convincing lip sync. Each can be traded off against the others, with varying effects on the audience’s experience of the resulting product.

We operationalize this broad question as several more specific ones about the human dubbing process:¹

Isochrony Do dubbers respect timing constraints imposed by the video and original audio?

Isometry Do the original and dub texts have approximately the same number of characters?

Speech tempo How much do voice actors vary their speaking rates, possibly compromising speech naturalness, to meet timing constraints?

Lip sync How closely do the voice actors’ words match visible mouth movements of the original actors?

Translation quality How much will dubbers reduce translation accuracy (i.e., adequacy and fluency) to meet other constraints?

Source influence Do source speech traits influence the target in ways not mediated by the words of the dub, indicating semantic transfer?

After exploring each of these questions, we provide insights on several research directions to address weaknesses we uncover in current automatic dubbing approaches.

2 Related Work

2.1 Qualitative

Modern qualitative research on human dubbing began with a seminal monograph by Fodor (1976),

¹We do not consider other constraints or synchronies, like cultural fit with the target audience; though such constraints are important, they are too difficult to examine quantitatively.

himself a translator and writer of dubbed dialogue. He explored many of the constraints and methods which later literature elaborated.

Dubbing (both human and automatic) has subsequently come to be viewed as a type of *constrained translation*, with more constraints than settings like comics, songs, or voice-over video content (Mayoral et al., 1988). Most of the constraints stem from the need for a close match to the original video track.

In particular, dubs have *isochronic* constraints: They should be about the same duration as the source, and should respect perceptible pauses within a speaker turn (Miggiani, 2019). Similarly, dubs benefit from complying with *phonetic synchrony*², also known as *lip sync*: compatibility between the articulatory mouth movements required to produce the dub and the mouth movements, when visible, of the original actors (Fodor, 1976; Miggiani, 2019).

Dubs also need to consider *kinesic synchrony*: the plausibility of the dubbed dialogue in light of visible body movements of the original actors (Chaume, 2012). These three constraints—*isochronic*, *phonetic* and *kinesic*—are true “synchronies” in modern usage as they relate to time. *Kinesic synchrony* is also an example of the broader category of *semiotic* or *iconic* constraints, or constraints “inherent to film language” (Chaume, 2020): the need for coherence between the language of the dub and the visual information of the film (Martí Ferriol, 2010).

Dubs, of course, have non-temporal constraints as well. As cultural products, they should be readily intelligible to a member of the target linguistic and cultural community, with foreign references avoided or used for effect. (As Chaume [2020] puts it, they must comply with “sociocultural constraints.”) As speech, they should sound natural, as though originally recorded in the target language. Dubs which fail to meet this criterion are often considered examples of “dubbese” (Myers, 1973). The peculiarities of “dubbese” have been studied extensively in a wide range of language pairs (see Herbst, 1997; Nencioni, 1976; Pavesi, 1996; Freddi and Pavesi, 2009; and many others), especially as it may be specific to a national or linguistic community of dubbers (Chaume, 2020).

²This is the term used in the literature, but synchrony of visemes (Fisher, 1968) would be a more accurate name, as lip sync bears on externally visible mouth movements.

Turning to content, dubs have the same goal as any translation of preserving the semantic meaning of the source. However, some leeway is allowed; Chaume (2020) provides two examples: (1) In a Spanish-to-English dub, an off-screen omelet may be turned into a pie, as the word for pie better adheres to lip-sync constraints. (2) In a Japanese-to-English dub, non-visible chopsticks might be changed to a fork to adhere to sociocultural constraints. Viewed through this lens, dubbing is a form of non-literal translation called “transcreation” (Zanotti, 2014). However, it is often desirable to keep such changes to a minimum to preserve fidelity to the source film (Martí Ferriol, 2010).

Finally, other qualitative research has examined the social and textual nature of dubbing (Bosseaux, 2018; Chaume, 2020). Scholars have investigated the role of power, ideology, identity, and similar considerations in the production of dubs (Miggiani, 2019; De Marco, 2012; Santamaria, 2016).

2.2 Automatic Dubbing

Several studies have explored the automatic generation of dubs, focusing on a variety of constraints.

One line of work has focused on integrating lip sync constraints into the dub generation process. Taylor et al. (2015) developed a method for automatic dubbing that matches the visemes of the original speech. Saboo and Baumann (2019) integrated lip-sync constraints into an encoder-decoder machine translation architecture. Taking a different approach, Kim et al. (2019) have explored adjusting mouth movements in the original video to match a dubbed audio track.

Other literature has examined “isometric” machine translation: producing a translation for use in automatic dubbing which has a similar length (in characters) to the input. It’s argued that this property is “a proxy for the duration of its spoken realization” (Lakew et al., 2021), and that similarity in character length makes text to speech (TTS)-generated speech sound more natural (Lakew et al., 2022). This approach has garnered interest from the community in the form of a shared task at IWSLT 2022 (Anastasopoulos et al., 2022).

A third line of work has focused on controlling the speaking rate in automatic dubbing systems to achieve *prosodic alignment*, or “synchronizing

the translated transcript with the original utterances” (Federico et al., 2020b). Öktem et al. (2019) focused only on the linguistic content matching between source-target phrases as a way to improve TTS, while Federico et al. (2020a) focused on fluency. Their subsequent work (Federico et al., 2020b; Virkar et al., 2021) further enhanced prosodic alignment by addition of features controlling for TTS speaking rate variation and linguistic content matching. Additionally, they introduced a time-boundary relaxation mechanism that can help to control speaking rate and speech fluency. Virkar et al. (2022) extended the time-boundary relaxation to further relax timing constraints for sentences that are off-screen. Tam et al. (2022) examined integrating pause constraints directly into MT. Finally, in contrast to the pipeline architecture used in most automatic dubbing works, Hu et al. (2021) explored end-to-end dubbing.

2.3 Empirical Studies

In recent years, some studies have attempted to examine human dubbing through a quantitative lens, providing empirical information to inform theoretical debate. One line of work, such as Sánchez-Mompeán (2020a,b), has done detailed studies of prosody in human dubs, generally in a language-specific way. Other recent work has employed laboratory eye-tracking studies (Perego et al., 2016) to gauge audience reaction. Di Giovanni and Romero-Fresco (2019), in particular, found that audiences may not be as sensitive to lip sync as traditionally believed. They report the existence of a “dubbing effect”, in which audiences subconsciously avoid looking at the mouth movements of on-screen actors when dubbed speech fails to be lip synced.

In the ML literature, recent work by Karakanta et al. (2020) concluded that on-screen human dubs have significantly lower translation quality (i.e., translation adequacy and/or fluency) than human off-screen dubs, with the drop in quality attributed to the need to satisfy constraints (e.g., isochrony) not applicable or less applicable to offscreen dubs. They draw this conclusion—on the HEROES corpus (Öktem et al., 2018)—by training a show-specific MT system and showing that it has lower performance (as measured by BLEU against the human dub) for on-screen than off-screen.

Genre	Language	Episodes	Shows	Speakers	Duration (hrs)	Dialogue Lines
ALL	English	674	54	9,215	319.6	234,322
	German	72	13	2,498	43.2	29,210
	Spanish	197	18	7,384	118.7	28,720
Drama	English	264	23	4,737	161.5	115,549
	German	39	7	1,817	29.2	18,892
	Spanish	132	10	5,809	93.7	22,125
Kids	English	320	17	2,086	113.2	82,508
	German	32	5	654	13.7	9,972
	Spanish	60	7	1,483	23.0	6,224
Comedy	English	157	16	2,449	74.1	61,080
	German	23	4	1,023	14.8	13,146
	Spanish	58	6	2,197	30.4	7,942
Suspense	English	52	6	1,002	33.1	23,008
	German	8	2	336	5.7	3,338
	Spanish	19	3	902	12.9	3,985

Table 1: Number of episodes (“Episodes,” e.g., a 45-min video), shows (“Shows”, e.g., a show might have 2 seasons each with 10 episodes), number of speakers as estimated by the number of distinct characters in the given show (“Speakers”), total run time for the show (“Duration”), and the number of distinct dialogue lines (“Dialogue Lines”) for the show. We report statistics for the entire corpus (“All”) as well as four genres (Drama, Kids, Comedy, and Suspense), in each of the 3 languages considered in this work (English, the source language, as well as German and Spanish dubs).

3 Corpus Description & Preprocessing

We begin with a dataset consisting of every TV show produced by Amazon Studios which was available on Prime Video at the end of 2021 for which we were able to locate a hand-curated transcript (for English shows) or dubbing script (for dubbed shows). These scripts are produced as part of the human dubbing process (see § 1 for more details). This dataset contains 674 episodes of 54 shows, constituting 319.57 hours of content from 9,215 distinct speakers. A detailed summary of this dataset is provided in Table 1. Prime Video reports one or more genres per show—to provide more insight into the characteristics of this data, we report statistics for all genres for which we have at least 400 lines of manual on/off annotations (see § 3.5 for more details): Drama, Kids, Comedy, and Suspense. These subsets are used extensively in future sections to check the robustness of our conclusions. Note that these genre subsets have some overlap, due to the fact that some shows have more than one reported genre.

All shows were originally recorded in English; we acquired both audio and video for the English

originals and audio tracks for the professionally produced Spanish and German dubs where available. Much of our analysis relies on a subset of 35.68 hours of content with both Spanish and German dubs. Our dataset also includes final transcripts from both the original and dubbing videos, which contain dialogue lines read by original or voice actors, with each line having a timestamp or “timecode” indicating its relative start time within the episode.

We perform extensive quality filtering prior to analysis. Data amounts for the entire corpus as well as each genre/language subset, at each stage of processing/filtering described below, are provided in Table 2.

3.1 Segmentation and Forced Alignment

The first step of our data preparation pipeline uses script timecodes to segment audio tracks. As scripts do not include end times, each dialogue line is associated with the audio between its start time and the start time of the next line (or the end of the episode for the last line). Lines are roughly, but not exactly, the same as speaker turns: Sometimes one line is only part of a speaker turn, and more

Subset	Orig	Filter	Align	On/Off
ALL	292,252	201,246	42,850	3,617
Drama	156,566	115,159	27,845	3,097
Kids	98,704	72,938	14,351	446
Comedy	82,168	54,525	21,034	2,278
Suspense	30,331	20,789	5,046	608
German	29,210	25,739	22,892	1,926
Spanish	28,720	23,196	19,958	1,691

Table 2: Total number of dialogue lines, for various stages of filtering, for all of the data (“ALL”), genre subsets (Drama, Kids, Comedy, and Suspense), and both target language subsets (German and Spanish). The “Orig” column gives the number of lines before any filtering (see § 3). The “Filter” column gives the number of lines after quality filtering described in sections § 3.1 and § 3.2. The “Align” column gives the number of lines after cross-lingual alignment described in § 3.3. Finally, the “On/Off” column gives the number of lines which have manual on-screen / off-screen annotations (see § 3.5).

rarely one line may include multiple speakers or crosstalk. The timecode-based segmentation process produces 234,322 dialogue lines for English, 29,210 for German, and 28,720 for Spanish.

Next, we use the Montreal Forced Aligner (McAuliffe et al., 2017; MFA) to force align each dialogue line with its corresponding audio, producing a sequence of phones spoken in each word, along with start and end times for each phone. MFA successfully aligns 87.37% of English lines (204,734), 89.35% of German lines (26,099), and 80.81% of Spanish lines (23,209).

Speaker fundamental frequency (i.e., F0 or, less formally, “pitch”) is extracted using pyworld³ and linearly interpolated to fill in missing values, and energy is computed from Mel spectrograms of the speech signals. Both pitch and energy are averaged on a per-phone basis.

3.2 Filtering

There are several ways our data collection, segmentation, and alignment procedures might fail. We extensively filter the English side of the da-

³<https://github.com/JeremyCCHsu/Python-Wrapper-for-World-Vocoder>.

taset to identify and remove erroneous dialogue lines. Specifically, we filter out the following:

Foreign-language Text We identify dialogue lines in the English originals whose text is not in English. We use a language identification model for text⁴ and exclude anything with a low probability of being English, as well as one entire show whose script text appeared not to be in English.

Foreign-language Audio Similarly, we identify lines with non-English audio (from original non-English speech and errors in the corpus), using an audio language identification model trained on the VOXLINGUA107 corpus from the SpeechBrain toolkit (Ravanelli et al., 2021; Valk and Alumae, 2021). We excluded an entire show whose supposedly English audio was actually German, several characters who spoke only in non-English languages, and any lines with low probability of being English.

Multiple Speakers or Overlapping Speech Because overlapped speech is likely to confuse MFA, we ran overlapped speech detection (Bredin et al., 2020; Bredin and Laurent, 2021), and excluded anything with a detected fraction of overlap higher than 30%.

Incorrect Alignments We performed ASR on each line’s audio using an in-house tool and excluded dialogue lines with a) empty ASR output, b) an exact match to the gold text except for an inclusion at the front (these indicate segmentation errors), or c) a Levenshtein distance to the original greater than 80% of the original length.

After filtering, we have 201,246 dialogue lines, from 688 episodes and 52 shows, constituting 355.36 hours of source and target content. Manual inspection with Praat (Boersma and Weenink, 2022) suggests that post-filtering alignment quality is acceptably high. Most words and phones are correctly aligned, with only 12% of words in a hand-audited sample containing any phones with major problems. Errors most frequently occurred on foreign words and at silence boundaries, with word-initial or word-terminal phones incorrectly aligned into a preceding or following silence.

3.3 Cross-lingual Alignment

Given sets of force-aligned and filtered content in each language, we still need to align across

⁴<https://huggingface.co/papluca/xlm-roberta-base-language-detection>.

languages to create a single corpus of parallel (English, dub) examples.

Offset Finding Many of the dubbed shows have episode-initial inserts, such as recaps of previous episodes or intro segments. Our dataset lacks target-side videos, and in lieu of manually identifying these segments from the audio tracks, we rely on cross-correlation of the aligned speech signals. For each (English, dub) pair of a given episode we sample at 100-Hz a binary indicator of whether MFA has aligned a non-silence phone, and find the offset that maximizes the cross-correlation of these two signals. By inspection, these offsets work well and produce closely correlated patterns of silence between English and dubbed content. This process revealed 3 episodes with quality issues, which were dropped from further analysis.

Sentence Alignment Finally, we need to align groups of sentences occurring at approximately the same time in each (English, dub) episode pair. Note that because voice actors do not have to respect the exact distribution of silences in the original audio track, we have a many-to-many alignment problem: Many stretches of speech in English may correspond to many, indeed potentially a different number, of stretches of speech in the dub. Accordingly, we align mostly according to content, using the Vecalign algorithm (Thompson and Koehn, 2019, 2020) on multilingual LASER embeddings (Artetxe et al., 2018) of the English and dubbed lines. We align contiguous stretches of speech. After alignment, we perform two final filtering steps to remove any spurious alignments, dropping sentence pairs where either duration is exactly one frame, or the midpoint times of the source and target speech segments differ by more than 1s.

The final dataset of parallel cross-lingual alignments contains 42,850 aligned dialogue line pairs, from 49 episodes and 11 shows, constituting 35.68 hours of source and target content.

3.4 Gender Annotations

We extract information from the *dramatis personae* lists in the original scripts on characters' genders. The scripts do not list all characters from whom we have speech, and differences in name formatting mean that some characters' gender information is lost. We are able to collect gender

annotations for 23,304 dialogue lines (54.39% of the filtered corpus).

3.5 On-Screen Annotations

We used annotations in the German dubbing scripts to identify on-screen (i.e., when the actor's mouth is visible) and off-screen (the actor's mouth is not visible) speech in the 49 episodes which had English, Spanish, and German versions. Because these are the scripts actually used by dubbing professionals, they are not only human judgments of when characters' mouths are visible, but also directly influenced the actual human dubbing process. Only approximately 9.68% of aligned pairs have on-screen/off-screen annotations.

Because much of our analysis rests on comparing onscreen and offscreen dialogue lines, we would also like to test for systematic differences in what type of content is onscreen or offscreen. In particular, we look for statistically significant differences in the duration of onscreen and offscreen lines, and (to measure the complexity of speech) the average perplexity of the GPT-2 language model (Radford et al., 2019) on each set of lines. Reassuringly, neither is significantly different: an independent-samples t-test fails to reject the null hypothesis that onscreen and offscreen examples have the same source-side mean duration ($p = 0.106$), and bootstrapping the average GPT-2 perplexity fails to reject the null hypothesis that it is the same between onscreen and offscreen ($p = 0.08$). It is possible that dubbing professionals themselves skip adding on/off annotations in cases (like narration) when it would be obvious from the text itself whether it is onscreen or offscreen.

3.6 Data Release Considerations

Unfortunately, content licensing restrictions prevent us from releasing our data. We believe this will be the case for any similar high-quality, large corpus: professionally written, acted, produced, and dubbed shows are proprietary for commercial reasons.

Note that a few prior studies (Pavesi, 2009; Öktem et al., 2018) have released human dubbing datasets; however, these datasets are much smaller than the dataset considered in this work and the legality of these datasets relies on a very permissive interpretation of "fair use", which may not be acceptable at some organizations.

4 Analysis

4.1 Isochrony

Perhaps the most obvious constraint of dubs is isochronic: Dubbed speech should line up in time with the original speech. This constraint is especially binding when the character’s mouth is visible (“onscreen”), but may apply for other reasons even when it is not (“offscreen”): Examples include cuts or transitions in the video, surrounding onscreen speech, and the need to align with actors’ body movements. Much qualitative work has considered isochronic constraints (e.g., Chaume, 2012; Miggiani, 2019; Fodor, 1976), and automatic dubbing work has explored integrating them, usually with a proxy for isochrony such as length in syllables (Saboo and Baumann, 2019; Öktem et al., 2019) or in characters (Federico et al., 2020a; Lakew et al., 2021, 2022; Tam et al., 2022). We are accordingly interested in exploring how much human dubbers respect this constraint.

First, we simply compare the durations of aligned dialogue line pairs. Source duration ought to be a strong predictor of dub duration, which indeed it is: The correlation⁵ between the two is quite high at $r = 0.877$. But duration does not consider the actual start and stop times of lines, and may simply reflect the need to convey the same amount of information in source and target.

As a further check, we look at the *overlap fraction* of speech time: the amount of time in each dialogue line when *both* the original (source language) actor and the dubbing voice actor (target language) are speaking (i.e., the intersection), divided by the amount of time when *either* is speaking (i.e., the union). A value of 1.0 indicates perfect time alignment, while 0.0 indicates the source and target speech occur at entirely different times. The mean overlap fraction in our corpus is 0.658 and the median is 0.731—in 4.3% of lines, overlap is exactly 0, pulling down the mean. Thus, while human dubbed speech mostly co-occurs with source speech, isochronic constraints are also frequently violated by human dubbers.

We observe that on-screen dubs are more isochronic than off-screen, but to a surprisingly small degree. The average offscreen dialogue line has overlap fraction 0.662, vs 0.684 on-screen—an

increase of only 3.3%. Excluding animated shows where characters’ (animated) mouth movements may be less constraining, this gap rises slightly: offscreen overlap of 0.656, vs 0.690 onscreen, for an off-to-on increase of 5.2%. Both differences, while small in magnitude, are statistically significant at the $\alpha = 0.01$ level under independent-samples t-tests (overall: $t = -2.93$, $p = 0.003$; live-action: $t = -5.35$, $p = 8.7 \times 10^{-8}$). For individual genre subsets, we find the offscreen to onscreen gap is significant for Drama ($t = -2.95$, $p = 0.003$) and Comedy ($t = -3.70$, $p = 0.0002$) but not for Kids and Suspense. The increase is not significant at the $\alpha = .01$ level for either language (German: $t = -2.28$, $p = 0.02$, Spanish: $t = -1.57$, $p = 0.12$).⁶

The small gap in on- vs off-screen isochrony may be partially explained by our on/off screen annotations: The dubbing professionals are likely only annotating sections where on- and off-screen dialogues are mixed, and the on-screen constraints may be constraining preceding/successive off-screen lines.

4.2 Isometry

Past work (Lakew et al., 2022; Anastasopoulos et al., 2022) has examined similarity of text length (measured in characters) as a way to constrain translation for automatic dubbing, especially a requirement that the target translation be within $\pm 10\%$ of the source character length. This practice is called “isometric machine translation” (Lakew et al., 2022), and we refer to the length constraint as “isometry.” This literature uses isometry mainly as a proxy for similarity of duration and for isochrony, though it may also help avoid large variations in TTS output rates (Lakew et al., 2022). We aim to test these assumptions: How good a proxy is isometry for isochrony in human dubs, and how much do human dubbers preserve character length?

We examined the text length (measured in characters) of aligned (source, human dub) dialogue line pairs, and especially the percentage change in character length from source to dub. Character lengths on both sides included punctuation

⁵Correlations are Pearson unless otherwise noted.

⁶For significance tests in this work, unless otherwise noted we test for the entire corpus for which the test is valid, as well as for subsets of the valid corpus corresponding to each target language (German and Spanish) and the genres listed in Table 1 (Drama, Kids, Comedy, and Suspense).

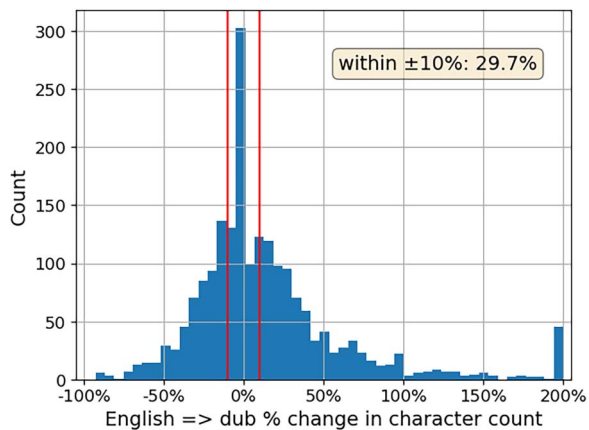


Figure 1: Percentage change in character count from English source to human dub among onscreen lines. The plot is clipped at a 200% increase (and the character count can’t decrease by more than 100%). Vertical red lines indicate a $\pm 10\%$ change.

and spaces (except at the start or end of a dialogue line). To measure how well character length similarity proxies for similarity of duration, we compared the ratio of target to source length to the ratio of target to source duration. We examine here only known onscreen lines, as these are subject to the greatest pressure to be isochronic; results are very similar if using all lines.⁷

We find first that isometry is a weak-to-moderate proxy for isochrony. The ratio of human dub to English character lengths has a correlation of only $r = 0.279$ ($r^2 = 0.078$) with the time overlap fraction of source and target, though it is somewhat more correlated with the ratio of target to source durations ($r = 0.620$, $r^2 = 0.385$).

Our results on character length similarity in human dubs, meanwhile, are summarized in Figure 1. Overall, there are large changes in character length from source to human dub. Most sentence pairs differ in length by more than prior work’s 10% threshold. The absolute percentage change in character length is significantly different from 0 under a one-sample t-test ($t = 20.3$, $p = 3.9e-83$). These changes are significant for both languages and all genre subsets with $p < 1e-38$. Character lengths are more similar for longer sentences (and the distribution of character count change is smoother), but nearly 60% of pairs in which the source sentence is at least 50 characters long differ in length by more than 10%.

⁷We exclude from analysis dialogue lines where either source or target had an aligned duration less than 0.2s; by inspection, most of these lines are segmentation errors.

We observe that human dubs are largely non-isometric in both Spanish and German, with neither language differing from the English source lines by more than 10% in less than 69% of cases. The length differences are, however, distributed differently. German skews toward longer dub lines, with 53% of all lines longer than the matched English lines by $\geq 10\%$, and 16% shorter by $\geq 10\%$; Spanish displays a smaller skew in the opposite direction, with 30% at least 10% longer and 42% at least 10% shorter.

4.3 Speaking Rate

Previous literature has paid considerable attention to the naturalness of human dubbed speech (Sánchez-Mompeán, 2020b). A frequent, though not universal, conclusion is that dubs sound “artificial and contrived” (Chaume, 2020), for reasons ranging from strange intonations to “anglicisms” inspired by the source language (Fresco, 2009). From another angle, the isometric MT literature argues that TTS models, which are less flexible than humans in varying speaking rate, may require isometric input to produce natural sounding isochronic output (Lakew et al., 2022). Because naturalness is a broad topic, and in general may require human evaluation, we focus on examining speaking rates. We’re particularly interested in whether dubbing voice actors are willing to vary their speaking rates, and perhaps compromise naturalness, in order to meet other constraints, like isochrony.

We examined both the dub speaking rate⁸ and the ratio of human dub duration to source duration as functions of the number of words in source and target dialogue lines.⁹ As the dub-to-source ratio of length in words (“word length ratio”) increases, in other words, what happens to dub speaking rate and the duration ratio?

Perhaps counterintuitively, it seems that the duration ratio is much more closely related to relative length of content than the dub speaking rate. Simple linear regression of each outcome variable on the word length ratio indicates a correlation of 0.523 between word length ratio and duration

⁸In this work, we calculate speaking rate as the average number of words spoken per second in each dialogue line, including pause time, following Laver (1994).

⁹As in § 4.2, we excluded lines with either source or target duration less than 0.2s. Results are robust to thresholds as low as 0.06s. We also exclude 4 dialogue lines, which appear to be alignment errors, in which the target-to-source duration ratio was more than 20.

ratio ($r^2 = 0.273$), while human dub speaking rate has a correlation of only 0.163 with duration ratio ($r^2 = 0.027$).

As an additional check, we examined the variance of speaking rate (at the dialogue-line level) in source and human dub. If the dubbing voice actor is varying speaking rate to meet timing constraints, we would expect more variability in the dubbed speech than the source speech. We do not, however, observe this: The standard deviation of dubbing voice actor speaking rate is lower for both Spanish (1.25 w/s, vs 1.47 w/s on the source side) and German (1.26 w/s, vs 1.46 w/s on the source side). In both cases we can reject the null hypothesis that the standard deviation of speaking rate is higher for dub than source via a percentile-bootstrap test (Spanish: $p < 10^{-10}$; German: $p < 10^{-10}$). Likewise, we can reject the null hypothesis for all genre subsets considered with $p < 10^{-10}$.

When forced to pick one or the other, human dubbers appear more willing to break timing constraints than vary speaking rate.

4.4 Lip Sync

Both qualitative (Chaume, 2012; Fodor, 1976; Miggiani, 2019) and technical work (Taylor et al., 2015; Hu et al., 2021; Kim et al., 2019) has considered “lip sync” constraints in human and automatic dubbing, respectively. The idea is that dubbed audio should match the (visible) mouth movements of the original actors. Failing to do so may be jarring to the audience and reduce the quality of the dub. Some recent empirical studies, however, have found that this constraint may not be as binding as previously assumed (Perego et al., 2016). Accordingly, we ask here whether human dubbers produce speech which matches the mouth movements of the original actors.

Rather than relying on the video tracks, we use the notion of a “viseme”, or visual phoneme (Fisher, 1968), to capture alignment between source and human dub mouth movements. Phones in the same viseme are produced with similar articulatory movements of the lips and tongue, and look visually similar. We use viseme tables¹⁰ to map each MFA-aligned phone to its corresponding viseme. The glottal stop and four guttural

German sounds made without moving the lips are dropped. To measure cooccurrence, we sample the viseme active on both source and dub sides and compute the viseme-viseme cooccurrence matrix. We normalize the matrix so that the observed frequency of each viseme pair is a fraction of the frequency expected if source and target visemes were independent, but with the observed marginal distributions.

Over all the data (onscreen, offscreen, and unannotated), the average within-viseme cooccurrence rate as a fraction of the rate under independence is 1.575, with an average across-viseme rate of 0.981. For onscreen the average within-viseme cooccurrence rate is 1.613, while for offscreen it is 1.463. We believe both on- and off-screen rates are above 1.0 due to the presence of names and cognates, where the phones may be (nearly) the same for the source and dub. The onscreen / offscreen differences are statistically significant at the $\alpha = 0.05$ level under a percentile-bootstrap test ($p = 0.017$) for the entire corpus, as well as for both languages and all genre subsets.

Moreover, we see similar patterns by language: The amount of excess cooccurrence is 40.8% for German (1.638 to 1.898)¹¹ and 37.8% for Spanish (1.439 to 1.605). But though the effect is significant, it is not large in absolute terms: even in onscreen speech, only about 12.4% of speech time has the same viseme on the source and target sides. This suggests that human dubbers do sometimes lip sync their output, but it is a fairly soft constraint.

Note that this analysis is sensitive to small errors in the exact start and stop times of aligned phones, and to inconsistencies across languages in the phone boundaries used to train aligners. Our results are thus likely to be a lower bound on how closely human dubbers observe lip sync constraints.

4.5 Translation Quality

As previously discussed (see § 1), the human dubbing process is complicated, with the translation modified throughout the process to satisfy isochrony, lip sync, and other constraints. Thus an obvious question to ask is how faithful the resulting translation actually is to the source material.¹²

¹¹The increase in excess cooccurrence is $(0.898 - 0.638) / 0.638 = 40.8\%$.

¹²We use the term *translation quality* here to refer to translation adequacy and fluency. It does not refer to the

¹⁰<https://docs.aws.amazon.com/polly/latest/dg/ref-phoneme-tables-shell.html>.

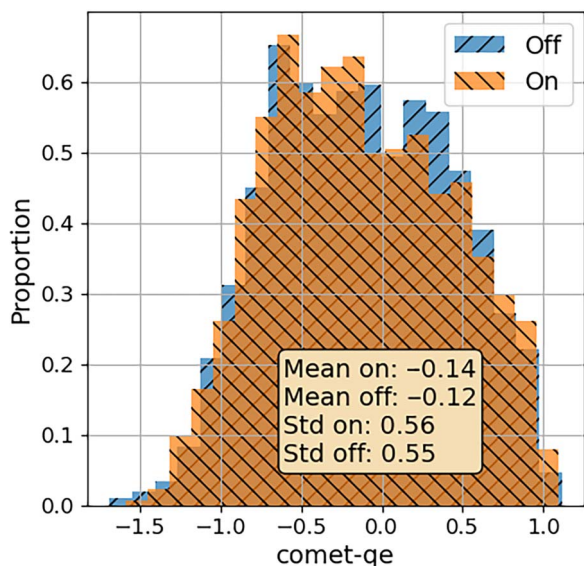


Figure 2: Comet-qe scores, including means and standard deviations, broken out by onscreen/offscreen status. Note that these scores represent only data points with onscreen/offscreen annotations. No meaningful onscreen/offscreen differences are observed in comet-qe or prism-src (not shown for space) scores.

Reducing translation quality may, after all, make it easier to satisfy other constraints: For example, changing the meaning in the target language may better fit the original English mouth movements than a more correct translation.

To address this question, we rely again on the onscreen/offscreen annotations. Isochronic constraints should be more binding onscreen, and we’ve shown above that the (inherently onscreen) lip sync constraints are at least partly followed. If translation quality is sacrificed to meet these other goals, we should see lower-quality translations onscreen than offscreen.

In contrast to Karakanta et al. (2020), we use a more straightforward approach of simply measuring the quality of the human dubs using automatic MT metrics. Since we do not have access to the original, pre-adaptation, human translation, we rely on reference-free metrics. In particular, we measure performance for each (source, human dub) pair with comet-qe (Rei et al., 2020) and prism-src (Thompson and Post, 2020a,b). Despite the lack of references, both have been shown to have better correlation with human judgements of MT quality than BLEU (Papineni et al.,

overall quality of the human dubbing output, which may intentionally decrease translation adequacy and/or fluency to meet other constraints.

2002), which requires a reference, in many cases (Thompson and Post, 2020a; Freitag et al., 2021).

The results are summarized in Figure 2. We find no substantial differences between onscreen and offscreen speech for either metric, with the on/off difference in means being less than 1/10th of a standard deviation for both comet and prism. Neither average comet scores ($t = 0.936$, $p = 0.349$) nor average prism scores ($t = 1.51$, $p = 0.131$) were significantly different onscreen from offscreen under a two-sided independent-samples t-test. Likewise, for both metrics, we did not find statistically significant differences for either language or any genre at the $\alpha = .05$ level. Though not depicted in Figure 2, the results are similar when broken out by language: Neither Spanish nor German dubs show any meaningful worsening of translation quality when onscreen. Human dubbers do not, in other words, appear to be sacrificing translation quality to hit other constraints.

4.6 Non-Text Transfer

Finally, we explore whether the human dub audio depends on the source audio in ways not mediated by the text of the dub translation.

We first look at source influence on three aspects of dubbing speech actor (target language) audio: speaking rate, pitch, and energy. For pitch and energy, we compute both the mean and the standard deviation per dialogue line, relying on higher standard deviation, and thus greater range, of pitch and energy as a crude indicator of emotion (Frick, 1985). (We drop from analysis of standard deviations any line with only one phone on the source or target side.) We also use the gender annotations extracted in § 3.4 to control for the effect of gender on the dubbing voice actor’s vocal pitch. As in § 4.2, we exclude dialogue lines where either source or target has aligned duration less than 0.2s.

Overall, we find that source audio properties explain a substantial fraction of target variance. Source speaking rate correlates with target speaking rate ($r = 0.439$, $r^2 = 0.193$), and the correlation is stronger the longer the dialogue lines. For lines with source and target both at least 1s long, the correlation is $r = 0.584$ ($r^2 = 0.342$). Line-level mean pitch is even more strongly related, with $r = 0.792$, $r^2 = 0.628$, though standard deviation of pitch is less so ($r = 0.429$, $r^2 =$

Property	C(Speaker)	+Source	Δ
Spk rate, 0.2s+	0.116	0.239	+0.122
Spk rate, 1s+	0.191	0.390	+0.198
Pitch mean	0.675	0.733	+0.057
Pitch std.	0.284	0.316	+0.032
Energy mean	0.184	0.268	+0.084
Energy std.	0.210	0.275	+0.065

Table 3: r^2 values for linear models predicting various properties of target audio (dubbing voice actor) from source audio (original actor). The first column reflects models containing only indicator or dummy variables for the speaker, while models in the second column add the line-level property for the source side. All increases in explained variance are significant at the $\alpha = 10^{-6}$ level by F-test. This finding also holds for both language and all genre subsets (not shown).

0.184). Both the mean ($r = 0.381, r^2 = 0.145$) and standard deviation ($r = 0.366, r^2 = 0.134$) of energy also display some linear relationship between source and target, though even more weakly.

By fitting sets of linear models predicting dubbing voice actor speaking rate, mean pitch and standard deviation of pitch, first as a function only of indicator / dummy variables for speakers, and second adding in the line-level property on the source side, we show this relationship is not simply a speaker-level effect: see Table 3. While speaker identity is generally a good predictor of target audio characteristics, dialogue line-level variables also increase predictive power. This line-level information is more useful for speaking rate than pitch, but its increase in predictive power is significant for both. Additionally, we find the gender of the source character is only a weak predictor of line-level mean pitch, with an indicator variable for male having only about $r^2 = 0.058$ in predicting the dub-side mean pitch.

Altogether, these results suggest that there is quite a bit of both speaker-level and line-level influence for future machine learning work to consider.

4.6.1 Semantic Transfer

Finally, as a more stringent check, we also conduct word alignment via FastAlign (Dyer et al., 2013) between each English dialogue line and its human dub. The alignment process produces (source,

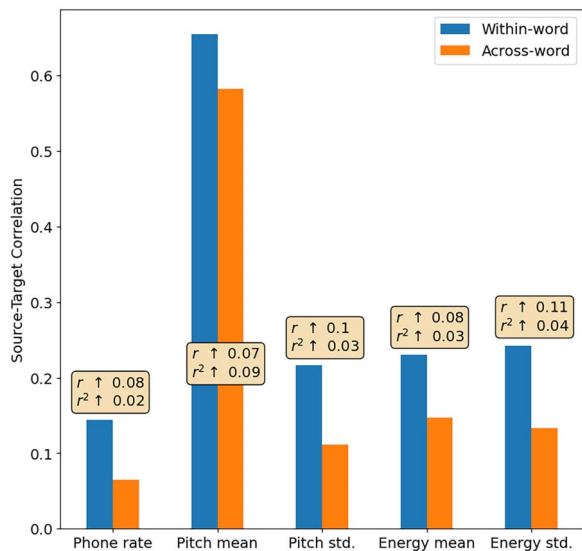


Figure 3: Pearson correlations of various audio properties between source and target (dubbing voice actors) within aligned word pairs (“Within-word”) and within unaligned word pairs (“Across-word”) in the same dialogue line. All properties show greater correlation within aligned word pairs than across them.

target) pairs of semantically similar words. If dubbing voice actors are imitating properties of the source speech in their own speech, we might expect to find that speaking rate, pitch and energy at the word level are more closely correlated within aligned word pairs than in other word pairs within the same dialogue line. We look in particular at the number of phones per second as well as the word-level mean and standard deviation of both pitch and energy.

All of these properties are, in fact, more closely correlated within aligned word pairs than in other word pairs, as shown in Figure 3. The amount of increase from across-pair to within-pair ranges from 0.08 to 0.11, with all six increases significant at the $\alpha = 10^{-6}$ level by a test based on Fisher’s z-transform for correlation coefficients. The increases are also significant for each language and genre subset at the $\alpha = 10^{-6}$ level.

As an even more stringent check, we first normalize the word-level variables, subtracting their line-level means and dividing by their line-level standard deviations. Doing so guards against the possibility that patterns at the line level, such as the amount of attention human dubbers pay to different types of line, influence these results. This analysis, shown in Table 4, confirms the findings of the unnormalized version. As expected, little to no relationship is visible between unaligned

Property	Within	Across	Δ
Phone Rate	0.117	-0.009	+0.125
Pitch Mean	0.112	-0.009	+0.120
Pitch Std	0.090	-0.007	+0.097
Energy Mean	0.092	-0.007	+0.099
Energy Std	0.108	-0.009	+0.116

Table 4: Pearson correlations of various audio properties between source and target (dubbing voice actor) within aligned word pairs (“within”) and within unaligned word pairs (“across”) in the same dialogue line. Word-level variables have first been normalized at the line level before being correlated, subtracting their line-level means and dividing by their line-level standard deviations.

pairs of words, while aligned pairs are weakly, but positively, correlated across several metrics. All of these differences are also significant at the $\alpha = 10^{-6}$ level by the same test as above, using Fisher’s z-transform. The increases are also significant for each language and genre subset at the $\alpha = 10^{-6}$ level.

These patterns clearly indicate that human dubbers are imitating properties of the source audio at a very granular (and *semantic*) level. We interpret these results as evidence of emotion and/or emphasis transfer from source to target.

5 Insights for Automatic Dubbing

Our analysis of the human dubbing process points to several directions that should (and perhaps should not) be pursued in automatic dubbing, which we summarize below.

Translation quality and **speech naturalness** appear to be paramount. The input to the dubbing process mostly consist of dialogue with challenging issues for automatic translation systems, like ambiguous speaker gender, ambiguous addressee gender and number, and formality between characters. Speaker gender and number issues are especially critical since the audience can often both hear and see the speakers and addressees. We note a stark lack of literature on automatic translation of dialogues, compared to common domains in literature like news. Likewise, naturalness for TTS systems is challenging enough under normal circumstances, but TV shows often include yelling, crying, whispering, and so forth, making the problem even harder. While research

does exist in this space, we suspect there is much room for improvement.

We find strong evidence for several levels of **non-textual transfer** of source audio properties into human dubs: speaker characteristics, dialogue line-level effects, and emotion/emphasis transfer when considering *semantic* alignments at the word level. This points to a glaring issue with pipeline approaches employed by the vast majority of automatic dubbing literature: Without a mechanism to encode emotion/emphasis, individual vocal profiles and other traits of the source speech, we expect them to be nearly impossible to replicate in synthetic target speech.

The high rates of **isochrony** that we observe in human dubs support the need for continued research on isochronic MT, especially given the observed unwillingness of human dubbers to vary their speaking rate, which shows that automatic dubbing systems should not simply vary speaking rates to achieve isochronic constraints. However, our findings do not support the use of **isometric MT**. Our work refutes the claim that isometry is a good proxy for isochrony, and it also casts doubt on the claim that isochrony is more necessary with TTS than with human voice actors because TTS is less able to vary speaking rates (i.e., we find that human dubbers are not varying speaking rate to meet isochronic constraints, and thus automatic dubbing systems should likely not either). The authors suspect that directly optimizing isochrony (as opposed to isometry) is likely a better approach for automatic dubbing.

Finally, the low rates of **lip sync** that we observe (and the very small if still statistically significant difference between on- and off-screen rates) in human dubs suggest that research on automatic lip sync can be marginally useful, at best, for automatic dubbing. Efforts like Kim et al. (2019), however, which edit mouth movements in the video, may be an exception.

6 Future Work

This work focused on two language pairs: English-German and English-Spanish. In future work, we hope to analyze more distant language pairs (e.g., English-Chinese or English-Arabic), as well as non-English source material.

Our analysis has shown that isometry is a poor proxy for isochrony in human dubs, yet several

prior works have claimed that isometric MT benefits automatic dubbing. In future work, we hope to perform analysis to understand this discrepancy.

The scope of this work necessitated automatic metrics. However, in future work, we hope to verify some of these findings (e.g., translation quality on- vs off-screen) using human annotators.

Finally, the aggregate analysis in this work is necessary to provide high-level insights for automatic dubbing. However, it likely also hides interesting variations across different individual translators, adaptors, dubbers, dubbing studios, and so on. We hope to better explore these dimensions in future work.

7 Conclusion

We present the first large-scale quantitative study of how humans perform the task of dubbing video content from one language into another. Our results challenge a number of popular assumptions in both qualitative and machine learning literature: Human dubbers display less respect for isochrony and especially lip sync than is suggested by qualitative literature, while being surprisingly unwilling to vary speaking rates or sacrifice translation quality to hit other constraints. Our analysis provides insights on research directions to address weaknesses in current automatic dubbing approaches.

Acknowledgments

This effort would not have been possible without Joel Chengottusseriyil, Robert Enyedi, Bradley Gordon, Natawut Monaikul, Prashanth Rajagopal, and Shuai Tang, who prepared the initial data used in this work. We also thank Marcello Federico, Prashant Mathur, Surafel Lakew, Reza Madad, and the anonymous TACL reviewers for helpful discussions and feedback.

References

Antonios Anastasopoulos, Loïc Barrault, Luisa Bentivogli, Marcello Federico, Ondřej Bojar, Roldano Cattoni, Anna Currey, Georgiana Dinu, Kevin Duh, Maha Elbayad, Clara Emmanuel, Yannick Estève, Marcello Federico, Christian Federmann, Souhir Gahbiche, Hongyu Gong, Roman Grundkiewicz, Barry Haddow, Benjamin Hsu, Dávid Javorský, Věra Kloudová, Surafel Lakew, Xutai Ma,

Prashant Mathur, Paul McNamee, Kenton Murray, Maria Nădejde, Satoshi Nakamura, Matteo Negri, Jan Niehues, Xing Niu, John Ortega, Juan Pino, Elizabeth Salesky, Jiatong Shi, Matthias Sperber, Sebastian Stüker, Katsuhito Sudoh, Marco Turchi, Yogesh Virkar, Alexander Waibel, Changhan Wang, and Shinji Watanabe. 2022. Findings of the IWSLT 2022 Evaluation Campaign. In *Proceedings of the 19th International Conference on Spoken Language Translation (IWSLT 2022)*, pages 98–157, Dublin, Ireland (in-person and online). Association for Computational Linguistics. <https://doi.org/10.18653/v1/2022.iwslt-1.10>

Mikel Artetxe, Gorka Labaka, and Eneko Agirre. 2018. A robust self-learning method for fully unsupervised cross-lingual mappings of word embeddings. In *Proceedings of the 56th Annual Meeting of the Association for Computational Linguistics (Volume 1: Long Papers)*, pages 789–798, Melbourne, Australia. Association for Computational Linguistics. <https://doi.org/10.18653/v1/P18-1073>

Paul Boersma and David Weenink. 2022. Praat: Doing phonetics by computer [Computer program]. Version 6.2.14, retrieved July 6, 2022 from <https://www.praat.org>, <https://www.fon.hum.uva.nl/praat/>

Charlotte Bosseaux. 2018. Dubbing, Luis Pérez-González, editor, *The Routledge Handbook of Audiovisual Translation Studies*, Routledge,

Hervé Bredin and Antoine Laurent. 2021. End-to-end speaker segmentation for overlap-aware resegmentation. In *Interspeech 2021*, pages 3111–3115. ISCA. <https://doi.org/10.21437/Interspeech.2021-560>

Hervé Bredin, Ruiqing Yin, Juan Manuel Coria, Gregory Gelly, Pavel Korshunov, Marvin Lavechin, Diego Fustes, Hadrien Titeux, Wassim Bouaziz, and Marie-Philippe Gill. 2020. Pyannote.Audio: Neural building blocks for speaker diarization. In *ICASSP 2020 - 2020 IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP)*, pages 7124–7128. <https://doi.org/10.1109/ICASSP40776.2020.9052974>

- Frederic Chaume. 2012. *Audiovisual Translation: Dubbing*, 1st edition. Translation Practices Explained, St. Jerome Pub, Manchester, UK.
- Frederic Chaume. 2020. Dubbing. In Łukasz Bogucki and Mikołaj Deckert, editors, *The Palgrave Handbook of Audiovisual Translation and Media Accessibility*, pages 103–132. Springer International Publishing, Cham. https://doi.org/10.1007/978-3-030-42105-2_6
- Delia Chiaro. 2008a. Issues in audiovisual translation. In *The Routledge Companion to Translation Studies*, 1st edition, page 25. Routledge.
- Delia Chiaro. 2008b. Issues of quality in screen translation: Problems and solutions. In Delia Chiaro, Christine Heiss, and Chiara Bucaria, editors, *Benjamins Translation Library*, volume 78, pages 241–256. John Benjamins Publishing Company, Amsterdam. <https://doi.org/10.1075/btl.78.24chi>
- Marcella De Marco. 2012. *Audiovisual Translation through a Gender Lens*, volume 37 of *Approaches to Translation Studies*. Rodopi, Amsterdam. <https://doi.org/10.1163/9789401207881>
- Elena Di Giovanni and Pablo Romero-Fresco. 2019. Chapter 6. Are we all together across languages? An eye tracking study of original and dubbed films. In Irene Ranzato and Serenella Zanotti, editors, *Benjamins Translation Library*, volume 148, pages 126–144. John Benjamins Publishing Company, Amsterdam. <https://doi.org/10.1075/btl.148.06di>
- Chris Dyer, Victor Chahuneau, and Noah A. Smith. 2013. A simple, fast, and effective reparameterization of IBM Model 2. In *Proceedings of the 2013 Conference of the North American Chapter of the Association for Computational Linguistics: Human Language Technologies*, pages 644–648, Atlanta, Georgia. Association for Computational Linguistics.
- Marcello Federico, Robert Enyedi, Roberto Barra-Chicote, Ritwik Giri, Umut Isik, Arvinth Krishnaswamy, and Hassan Sawaf. 2020a. From speech-to-speech translation to automatic dubbing. In *Proceedings of the 17th International Conference on Spoken Language Translation*, pages 257–264, Online. Association for Computational Linguistics. <https://doi.org/10.18653/v1/2020.iwslt-1.31>
- Marcello Federico, Yogesh Virkar, Robert Enyedi, and Roberto Barra-Chicote. 2020b. Evaluating and optimizing prosodic alignment for automatic dubbing. In *Interspeech 2020*, pages 1481–1485. ISCA. <https://doi.org/10.21437/Interspeech.2020-2983>
- Cletus G. Fisher. 1968. Confusions among visually perceived consonants. *Journal of Speech and Hearing Research*, 11(4):796–804. <https://doi.org/10.1044/jshr.1104.796>, PubMed: 5719234
- István Fodor. 1976. *Film Dubbing: Phonetic, Semiotic, Esthetic and Psychological Aspects*, 1st edition. Buske, Hamburg.
- Maria Freddi and Maria Pavesi, editors. 2009. *Analysing Audiovisual Dialogue: Linguistic and Translational Insights*. CLUEB, Bologna.
- Markus Freitag, Ricardo Rei, Nitika Mathur, Chi-kiu Lo, Craig Stewart, George Foster, Alon Lavie, and Ondřej Bojar. 2021. Results of the WMT21 Metrics Shared Task: Evaluating metrics with expert-based human evaluations on TED and news domain. In *Proceedings of the Sixth Conference on Machine Translation*, pages 733–774, Online. Association for Computational Linguistics.
- Pablo Romero Fresco. 2009. Naturalness in the Spanish dubbing language: A case of not-so-close friends. *Meta*, 54(1):49–72. <https://doi.org/10.7202/029793ar>
- Robert W. Frick. 1985. Communicating emotion: The role of prosodic features. *Psychological Bulletin*, 97(3):412–429. <https://doi.org/10.1037/0033-2909.97.3.412>
- Thomas Herbst. 1997. Dubbing and the dubbed text—style and cohesion. In Anna Trosborg, editor, *Benjamins Translation Library*, volume 26, page 291. John Benjamins Publishing Company, Amsterdam. <https://doi.org/10.1075/btl.26.21her>
- Chenxu Hu, Qiao Tian, Tingle Li, Wang Yuping, Yuxuan Wang, and Hang Zhao. 2021. Neural Dubber: Dubbing for videos according to scripts. In *Advances in Neural*

- Information Processing Systems*, volume 34, pages 16582–16595. Curran Associates, Inc.
- Alina Karakanta, Supratik Bhattacharya, Shravan Nayak, Timo Baumann, Matteo Negri, and Marco Turchi. 2020. The two shades of dubbing in neural machine translation. In *Proceedings of the 28th International Conference on Computational Linguistics*, pages 4327–4333, Barcelona, Spain (Online). International Committee on Computational Linguistics. <https://doi.org/10.18653/v1/2020.coling-main.382>
- Hyeonwoo Kim, Mohamed Elgharib, Michael Zollhöfer, Hans-Peter Seidel, Thabo Beeler, Christian Richardt, and Christian Theobalt. 2019. Neural style-preserving visual dubbing. *ACM Transactions on Graphics*, 38(6):1–13. <https://doi.org/10.1145/3355089.3356500>
- Surafel M. Lakew, Marcello Federico, Yue Wang, Cuong Hoang, Yogesh Virkar, Roberto Barra-Chicote, and Robert Enyedi. 2021. Machine translation verbosity control for automatic dubbing. In *ICASSP 2021 - 2021 IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP)*, pages 7538–7542, Toronto, ON, Canada. IEEE. <https://doi.org/10.1109/ICASSP39728.2021.9414411>
- Surafel M. Lakew, Yogesh Virkar, Prashant Mathur, and Marcello Federico. 2022. ISO-METRIC MT: Neural machine translation for automatic dubbing. In *ICASSP 2022 - 2022 IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP)*, pages 6242–6246, Singapore, Singapore. IEEE. <https://doi.org/10.1109/ICASSP43922.2022.9747023>
- John Laver. 1994. *Principles of Phonetics*. Cambridge Textbooks in Linguistics. Cambridge University Press, Cambridge.
- José Luis Martí Ferriol. 2010. *Cine independiente y traducción*, Prosopopeya (Tirant lo Blanch): Manuales, Tirant lo Blanch, Valencia.
- Anna Matamala. 2010. Translations for dubbing as dynamic texts: Strategies in film synchronisation. *Babel. Revue internationale de la traduction / International Journal of Translation*, 56(2):101–118. <https://doi.org/10.1075/babel.56.2.01mat>
- Roberto Mayoral, Dorothy Kelly, and Natividad Gallardo. 1988. Concept of constrained translation. Non-linguistic perspectives of translation. *Meta*, 33(3):356–367. <https://doi.org/10.7202/003608ar>
- Michael McAuliffe, Michaela Socolof, Sarah Mihuc, Michael Wagner, and Morgan Sonderegger. 2017. Montreal forced aligner: Trainable text-speech alignment using Kaldi. In *Interspeech 2017*, pages 498–502. ISCA. <https://doi.org/10.21437/Interspeech.2017-1386>
- Giselle Spiteri Miggiani. 2019. *Dialogue Writing for Dubbing: An Insider’s Perspective*, 1st edition. Palgrave Macmillan, Basingstoke, Hampshire. <https://doi.org/10.1007/978-3-030-04966-9>
- Lora Myers. 1973. The art of dubbing. *Filmmakers’ Newsletter*, 6(6):56–58.
- Giovanni Nencioni. 1976. Parlato-parlato, parlato-scritto, parlato-recitato. *Strumenti critici*, 29:49.
- Alp Öktem, Mireia Farrús, and Antonio Bonafonte. 2018. Bilingual prosodic dataset compilation for spoken language translation. In *IberSPEECH 2018*, pages 20–24. ISCA. <https://doi.org/10.21437/IberSPEECH.2018-5>
- Alp Öktem, Mireia Farrús, and Antonio Bonafonte. 2019. Prosodic phrase alignment for machine dubbing. In *Interspeech 2019*, pages 4215–4219. ISCA. <https://doi.org/10.21437/Interspeech.2019-1621>
- Mario Paolinelli and Eleonora Di Fortunato. 2009. *Tradurre per il doppiaggio: la trasposizione linguistica dell’audiovisivo: teoria e pratica di un’arte imperfetta*, 4th edition. Hoepli, Milano.
- Kishore Papineni, Salim Roukos, Todd Ward, and Wei-Jing Zhu. 2002. BLEU: A method for automatic evaluation of machine translation. In *Proceedings of the 40th Annual Meeting of the Association for Computational Linguistics*, pages 311–318, Philadelphia, Pennsylvania, USA. Association for Computational Linguistics. <https://doi.org/10.3115/1073083.1073135>

- Maria Pavesi. 1996. L'allocuzione nel doppiaggio dall'inglese all'italiano. In *Traduzione multimediale per il cinema, la televisione e la scena: atti del Convegno internazionale: Forlì, 26-28 ottobre 1995 = Multimediale Übersetzung für film ... = Multimedia translation for film ... / a cura di Christine Heiss, Rosa Maria Bollettieri Bosinelli*, pages 117–130. CLUEB.
- Maria Pavesi. 2009. Chapter 14: Dubbing English into Italian: A closer look at the translation of spoken language. In Jorge Díaz Cintas, editor, *New Trends in Audiovisual Translation*, pages 197–209. Multilingual Matters. <https://doi.org/10.21832/9781847691552-016>
- Elisa Perego, David Orrego-Carmona, and Sara Bottiroli. 2016. An empirical take on the dubbing vs. subtitling debate: An eye movement study. *Lingue e Linguaggi*, 19:255–274.
- Alec Radford, Jeffrey Wu, Rewon Child, David Luan, Dario Amodei, and Ilya Sutskever. 2019. Language models are unsupervised multitask learners.
- Mirco Ravanelli, Titouan Parcollet, Peter Plantinga, Aku Rouhe, Samuele Cornell, Loren Lugosch, Cem Subakan, Nauman Dawalatabad, Abdelwahab Heba, Jianyuan Zhong, Ju-Chieh Chou, Sung-Lin Yeh, Szu-Wei Fu, Chien-Feng Liao, Elena Rastorgueva, François Grondin, William Aris, Hwidong Na, Yan Gao, Renato De Mori, and Yoshua Bengio. 2021. Speech-Brain: A general-purpose speech toolkit. *arXiv preprint arXiv:2106.04624*.
- Ricardo Rei, Craig Stewart, Ana C. Farinha, and Alon Lavie. 2020. COMET: A neural framework for MT Evaluation. In *Proceedings of the 2020 Conference on Empirical Methods in Natural Language Processing (EMNLP)*, pages 2685–2702, Online. Association for Computational Linguistics.
- Ashutosh Saboo and Timo Baumann. 2019. Integration of dubbing constraints into machine translation. In *Proceedings of the Fourth Conference on Machine Translation (Volume 1: Research Papers)*, pages 94–101, Florence, Italy. Association for Computational Linguistics. <https://doi.org/10.18653/v1/W19-5210>
- Sofía Sánchez-Mompeán. 2020a. *Dubbing and Prosody at the Interface*, pages 19–88. Springer International Publishing, Cham.
- Sofía Sánchez-Mompeán. 2020b. *The Prosody of Dubbed Speech: Beyond the Character's Words*. Springer International Publishing. <https://doi.org/10.1007/978-3-030-35521-0>
- Laura Santamaria. 2016. Filmmaking: Cultural referents, terminology and identity, Lucía Molina and Laura Santamaria, editors, *Traducción, Interpretación y Estudios Interculturales*, Editorial Comares, Granada.
- Derek Tam, Surafel M. Lakew, Yogesh Virkar, Prashant Mathur, and Marcello Federico. 2022. Isochrony-aware neural machine translation for automatic dubbing. In *Interspeech 2022*. <https://doi.org/10.21437/Interspeech.2022-11136>
- Sarah Taylor, Barry-John Theobald, and Iain Matthews. 2015. A mouth full of words: Visually consistent acoustic redubbing. In *2015 IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP)*, pages 4904–4908. South Brisbane, Queensland, Australia. IEEE. <https://doi.org/10.1109/ICASSP.2015.7178903>
- Brian Thompson and Philipp Koehn. 2019. Vecalign: Improved sentence alignment in linear time and space. In *Proceedings of the 2019 Conference on Empirical Methods in Natural Language Processing and the 9th International Joint Conference on Natural Language Processing (EMNLP-IJCNLP)*, pages 1342–1348, Hong Kong, China. Association for Computational Linguistics. <https://doi.org/10.18653/v1/D19-1136>
- Brian Thompson and Philipp Koehn. 2020. Exploiting sentence order in document alignment. In *Proceedings of the 2020 Conference on Empirical Methods in Natural Language Processing (EMNLP)*, pages 5997–6007, Online. Association for Computational Linguistics. <https://doi.org/10.18653/v1/2020.emnlp-main.483>
- Brian Thompson and Matt Post. 2020a. Automatic machine translation evaluation in many languages via zero-shot paraphrasing. In *Proceedings of the 2020 Conference on Empirical*

- Methods in Natural Language Processing (EMNLP)*, pages 90–121, Online. Association for Computational Linguistics. <https://doi.org/10.18653/v1/2020.emnlp-main.8>
- Brian Thompson and Matt Post. 2020b. Paraphrase generation as zero-shot multilingual translation: Disentangling semantic similarity from lexical and syntactic diversity. In *Proceedings of the Fifth Conference on Machine Translation*, pages 561–570, Online. Association for Computational Linguistics.
- Christopher Titford. 1982. Sub-titling-Constrained Translation. *Lebende Sprachen*, 27(3):113–116.
- Jorgen Valk and Tanel Alumae. 2021. VOXLINGUA107: A dataset for spoken language recognition. In *2021 IEEE Spoken Language Technology Workshop (SLT)*, pages 652–658, Shenzhen, China. IEEE. <https://doi.org/10.1109/SLT48900.2021.9383459>
- Yogesh Virkar, Marcello Federico, Robert Enyedi, and Roberto Barra-Chicote. 2021. Improvements to prosodic alignment for automatic dubbing. In *ICASSP 2021 - 2021 IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP)*, pages 7543–7574, Toronto, ON, Canada. IEEE. <https://doi.org/10.1109/ICASSP39728.2021.9414966>
- Yogesh Virkar, Marcello Federico, Robert Enyedi, and Roberto Barra-Chicote. 2022. Prosodic alignment for off-screen automatic dubbing. <https://doi.org/10.21437/Interspeech.2022-11089>
- Patrick Zabalbeascoa. 1997. Dubbing and the nonverbal dimension of translation, Fernando Poyatos, editor, *Benjamins Translation Library*, volume 17, page 327. John Benjamins Publishing Company, Amsterdam. <https://doi.org/10.1075/btl.17.26zab>
- Patrick Zabalbeascoa. 2008. The nature of the audiovisual text and its parameters. In Jorge Díaz-Cintas, editor, *Benjamins Translation Library*, volume 77, pages 21–37. John Benjamins Publishing Company, Amsterdam. <https://doi.org/10.1075/btl.77.05zab>
- Serenella Zanotti. 2014. Translation and transcreation in the dubbing process: A genetic approach. *Cultus*, 7:109–134.