Grounded Sequence to Sequence Transduction

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Abstract—Speech recognition and machine translation have made major progress over the past decades, providing practical systems to map one language sequence to another. Although multiple modalities such as sound and video are becoming increasingly available, the state-of-the-art systems are inherently unimodal, in the sense that they take a single modality — either speech or text — as input. Evidence from human learning suggests that additional modalities can provide disambiguating signals crucial for many language tasks. Here, we describe the How2 dataset, a large, open-domain collection of videos with transcriptions and their translations. We then show how this single dataset can be used to develop systems for a variety of language tasks and present a number of models meant as starting points. Across tasks, we find that building multi-modal architectures that perform better than their unimodal counterpart remains a challenge. This leaves plenty of room for the exploration of more advanced solutions that fully exploit the multi-modal nature of the How2 dataset, and the general direction of multimodal learning with other datasets as well.

Index Terms—Multimodal machine learning, grounding, speech recognition, machine translation, summarization, representation learning

I. INTRODUCTION

Multimodal machine learning covers topics at the intersection of natural language processing, speech recognition, and computer vision [1]. Research in this area is motivated by recent advances in representation learning and the reported benefits of multi-sensory inputs: e.g. visual and tactile interaction increases infant sensitivity to colour differences over purely visual inputs [2], and psycholinguistic studies show the benefits of multiple modalities in concept representation [3]. Significant progress has been made in the last decade on major problems, including image captioning [4], visual question answering [5], image–sentence retrieval [6], and video captioning [7]. A common aspect of these problems is that they typically involve bi-modal learning, e.g. images and sentences in image captioning, due to the nature of the freely available datasets.

In recent years, there has been a collective effort in multilingual and multimodal representation learning, and models of visually grounded speech. In multimodal machine translation, researchers have focused on methods for integrating visual information into sequence-to-sequence models [8]–[10], and in multilingual image–sentence retrieval, it has been shown that cross-lingual sentence–sentence objectives improve retrieval performance [11], and that these findings extend to working with multiple languages [12]. In multimodal speech recognition, the image modality has been used to adapt the acoustic model [13], the language model [14] and, more recently, end-to-end systems [15], [16]. In spite of these recent successes, researchers have worked with bi- or multilingual datasets [17] that are much smaller than the datasets typically used for machine translation and speech recognition research.

This paper introduces the large-scale tri-modal How2 dataset, which consists of 2,000 hours of instructional videos with audio signals and two types of English text: closed captions of the speech and a self-written summary of the video, and crowdsourced Portuguese translations of a subset of the human annotated transcripts (Section II). The How2 dataset affords a wide variety of bi-, tri- and multi-modal experiments; here, we focus on multimodal speech recognition (Section III), multimodal machine translation (Section IV), abstractive video summarization (Section V), and multiview learning from speech, video, and multi-lingual transcripts (Section VI). The main findings from these experiments is that learning multimodal representations almost always results in better task-specific performance, and that there are numerous opportunities for future research on effective feature integration in multimodal learning.

II. THE HOW2 DATASET

In the How2 dataset, we collect 79,114 English instructional videos from YouTube with English subtitles. The dataset consists of a total of 2,000 hours of video. Videos have an average length of 90 seconds [18] and manual Portuguese translations. This collection of videos and translations constitutes a large-scale resource for testing a substantial part of multimodal language processing methods in a real-world scenario.

An alignment process is needed to use the audio, the English subtitles, the Portuguese translations, and the video modality together. To this end, we first re-segment the English subtitles into sentences using NLTK [19]. Then, we force-align the speech signal at the word level with an HMM-GMM pre-trained on the Wall Street Journal dataset. Finally, using the

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† The tools to download and construct the corpus are freely available at https://github.com/srvk/how2-dataset.
timings provided by the word alignment, we create video clips aligned to the initial segmented sentences. This process splits a video into a sequence of clips, aligned with the speech signal and the segmented sentences. Table I presents summary statistics of the 2000h set and 300h subset: the val and test sets can be used for early-stopping, model selection and evaluation; the held set is reserved for future evaluations or challenges. The total set (i.e. 2000h) contains around 22.5M words. The tokenized training set of 300h subset contains around 3.8M (43K unique) and 3.6M (60K unique) words for English and Portuguese respectively. Videos are broken down into clips, as described above, with an average length of 5.8 seconds, or 20 words of spoken language.

We collected Portuguese translations using the Figure Eight crowdsourcing platform, where we could reliably find Portuguese speaking crowdworkers. In order to speed-up the annotation process, we framed the translation task as a post-editing task. We first selected the best online machine translation service among three state-of-the-art services based on Figure Eight’s workers preferences. Then, we used the translations generated by this system as a proxy and paid the crowd workers to post-edit the translation. We attempted to ensure that the workers were in fact post-editing the proxies by replacing content words of the proxy with a random Portuguese word. If the substituted word remains in the post-edit, we removed the worker from the pool and re-collected the post-edit. Each of the 200 workers used in this project have a limit of post-editing 5,000 sentences. None of them reached the post-edit. Each of the 200 workers used in this project have a limit of post-editing 5,000 sentences. None of them reached the post-edit.

We estimated the quality of this process by comparing the performance of a translation model trained on either the post-editied translations or the machine-generated proxy-translations. The model trained on the proxy-translations performed 1 BLEU point worse on predicting the post-edited translations than the model trained on the post-edited translations, which suggests that our data collection method indeed resulted in different human-edited translation data.

To estimate the topic diversity in How2 dataset, we ran a Latent Dirichlet Allocation (LDA) [20] over the English subtitles. Then, we defined 22 clusters by analyzing empirical distances between videos and centroids. Finally, we applied a topic label to each cluster by analyzing the top words.

### Table I: Statistics of How2 dataset.

<table>
<thead>
<tr>
<th></th>
<th>Videos</th>
<th>Hours</th>
<th>Clips/Sentences</th>
</tr>
</thead>
<tbody>
<tr>
<td>300h</td>
<td>13,168</td>
<td>298.2</td>
<td>184,949</td>
</tr>
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<td></td>
<td>150</td>
<td>3.2</td>
<td>2,022</td>
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<td>175</td>
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<tr>
<td></td>
<td>169</td>
<td>3.0</td>
<td>2,021</td>
</tr>
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<td>2000h</td>
<td>73,993</td>
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<td>-</td>
</tr>
<tr>
<td></td>
<td>2,965</td>
<td>71.3</td>
<td>-</td>
</tr>
<tr>
<td></td>
<td>2,156</td>
<td>51.7</td>
<td>-</td>
</tr>
</tbody>
</table>

![Fig. 1: An example from How2 dataset where visual semantics can be helpful when transcribing *ukulele.*](image)

**III. MULTIMODAL SPEECH RECOGNITION**

Figure 1 shows an illustrative example from How2 where a purely monomodal ASR is prone to transcribe the utterance *ukulele* to an homophonetic equivalent *eucalyptus*. Earlier work in ASR suggests that a correlated auxiliary modality can be helpful within the context of instructional videos where videos consistently provide visual cues related to the speech semantics [13]–[15], [26]. This section discusses multimodal extensions to automatic speech recognition (ASR) with vision as supporting modality. We mainly explore two different multimodal interactions: first, we apply the visual adaptive training framework [13]–[15] to S2S ASR systems; second, we propose end-to-end multimodal grounding methods inspired by previous work in image captioning [27] and multimodal neural machine translation [28], [29].

### A. Features

In what follows, we detail the features that we extract for each modality.

1) **Speech Features**: For speech, we extract 40-dimensional filter bank features from 16kHz raw speech signal using a time window of 25ms and an overlap of 10ms. 3-dimensional pitch features are then concatenated to form the final 43-dimensional feature vectors. The speech features of a given video are further normalized using the mean and variance statistics from that specific video.

2) **Action Features (video-level)**: We extract action-level video features from a 3D ResNeXt-101 [21] pretrained on the Kinetics action recognition dataset [22] which comprises 400 different actions.

3) **Object Features (frame-level)**: A ResNet-152 [23] trained on ImageNet [24] which consists of 1000 categories ranging from animals, flowers to devices and foods and so on.

4) **Scene Features (frame-level)**: A ResNet-50 trained on Places365 [25] for scene recognition with 365 categories including, but not limited to: garden, valley, studio, theater and office.

5) **Object-level**: A ResNet-152 [23] trained on ImageNet [24] which consists of 1000 categories, ranging from animals and flowers to devices and foods.

![A ResNet-152](image)

**How2** dataset (Section II). For textual features, we first lowercase and remove punctuation from the English transcripts and then train...
a SentencePiece model [30] to construct a subword vocabulary of 5000 tokens. For speech, we use the 43-dimensional features (Section II) as they are. Finally for the visual modality, we explore two more pre-trained CNNs in addition to the action features described in Section II: a ResNet-152 [23] trained on ImageNet [24] for object recognition and a ResNet-50 trained on Places365 [25] for scene recognition. For all types of features, we obtain an average-pooled (avgpool) representation from the corresponding CNN. For object and scene-level features, we also experiment with class probabilities (prob) which are 1000-dimensional and 365-dimensional, respectively.

We explore two methods to obtain a single feature vector for each clip of a given video: (1) a per-clip representation by averaging frame level feature vectors of the clip and (2), a per-video representation by averaging frame level feature vectors across the whole video. We train all models with three randomly initialized instances using nmtpy [31]. For each instance, the best model is obtained by early-stopping on validation set word error rate (WER).

B. Baseline Model

All multimodal ASR systems in this section extend the well-known recurrent, attentive sequence-to-sequence model [32]. In the following, \(X = \{x_0, \ldots , x_{T-1}\}\) represents an input sequence of \(T\) speech features and \(f\) is the corresponding visual feature for that utterance. All recurrent, attention and embedding layers in the network are 320-dimensional.

The speech encoder is composed of 6 bidirectional LSTM layers [33], each followed by a tanh projection layer. The middle two LSTM layers apply a temporal subsampling [34] by skipping every other input, reducing the input sequence length \(T\) to \(T/4\). The decoder implements the so-called Conditional GRU architecture [35] where an attention mechanism [32] is wrapped between two GRU [36] layers. At timestep \(t=0\), the hidden state of the first GRU is initialized with the mean-pooled speech encoder state. The second GRU receives the output of the attention layer.

C. Multimodal ASR Systems

1) Visual Adaptive Training (VAT): This method fine-tunes a pre-trained ASR model using the visual modality. VAT adds a new linear layer to the model to project the visual feature vector \(f\) into the speech feature space. The projected utterance-specific shift vector is then added to the speech features and the network is jointly optimized until convergence:

\[ x_t = x_t + (W_v f + b_v) \quad t \in \{0, \ldots , T-1\} \quad (1) \]

2) Tied Initialization of Recurrent Blocks: Initializing the encoder and the decoder is an approach previously explored in multimodal machine translation [28], [29]. In order to prime the speech encoder with visual context, two non-linear layers are employed to learn an initial hidden state \(h_0^\kappa\) and an initial cell state \(c_0^\kappa\) for all the 6 LSTM layers in the encoder:

\[ h_0^\kappa = \tanh (W_h f + b_h) \quad k \in \{1, \ldots , 6\} \quad (2) \]

\[ c_0^\kappa = \tanh (W_c f + b_c) \quad k \in \{1, \ldots , 6\} \quad (3) \]

The same idea can also be applied to the first GRU in the decoder so that its initial hidden state is visually primed:

\[ h_0' = \tanh (W_d f + b_d) \quad (4) \]

Finally we explore a third variant where we fuse the two approaches by sharing the projection parameters in equations 2 and 4. In the following, these three variants will be referred to as \(\text{edinit}\), \(\text{dinit}\) and \(\text{edinit}\) respectively.

3) Visual Beginning-of-Sentence: Neural decoders receive a special beginning of sentence vector as input at timestep \(t=0\) in order to begin decoding. This vector can be either constant or learned during training, the latter being the approach taken in this work. The disadvantage of both methods is the fact that during inference, the decoder always receives the same embedding at \(t=0\) regardless of the input modality. Here we propose to modulate the decoder by using a visually-informed embedding for a given example \(i\):

\[ y_0^i = W_v f^i + b_v \quad (5) \]

D. Experimental Results

In what follows, we report single best, mean and ensembled WER across the three training runs of each model.

1) Visual Adaptive Training: In Table II, we clearly see that avgpool features consistently outperform class probability features. Similarly, a per-video representation seems to give a slight boost compared to per-clip granularity. Overall, avgpool features reduces the WER by up to 1.4% depending on the feature type and granularity. The contrastive restart continues
training the baseline ASR model without visual adaptation, and shows that the improvements are not a side-effect of training the model for additional epochs. But interestingly, once the learned adaptation layer is removed from the network so that the model falls back to the vanilla speech features $x_t$, the model still obtains around 18% WER. This seems to indicate that the effect of adaptation is indirect in the sense that it leads to a more robust ASR without necessarily relying on the visual modality.

2) **End-to-End Variants:** We observe that tied initialization ($edinit$) reduces the WER by 0.8% and 0.5% in terms of single best and mean scores, respectively (Table III). With ensembling, the $edinit$ variant reaches the best WER (15.0%) among all the models explored. The $visual$-bos method performs on par with the $edinit$. Action features give slightly better performance for both.

Returning to example in Figure 1, we checked how successful the systems are when transcribing the word **ukulele**. We observe that $edinit$ systems with action and object features could transcribe it once (out of ten occurrences in the test set) while the baseline system could not. However, this should be taken with a grain of salt, as the **ukulele** occurs only three times in the training set.

### E. Discussion

In this section, we first explored visual adaptive training for S2S ASR models and then experimented with novel multimodal extensions to S2S ASR. Our experiments showed that the method is effective for the S2S paradigm too, reaching up to 1.4% absolute WER improvement with action-level features. However, we also discovered that the adaptive system still preserves its performance even when the adaptation layer is removed during inference. We leave the analysis of this phenomenon to future work. Although end-to-end models perform better than the baseline, the difference is smaller compared to adaptive training. But when ensembling is used, the end-to-end models obtain the best WER among all models. With regard to visual representations, we show that average pooled CNN features perform better than class probabilities and the action-level features are slightly better than others.

### IV. Region-Specific Machine Translation

This section discusses another multimodal sequence to sequence task – Multimodal machine translation (MMT). MMT is a research field that aims to enrich textual context with additional modalities (images, videos, audio) for machine translation (MT). The assumption is that context provided by these modalities can help ground the meaning of the text and, as a consequence, generate more adequate translations. This becomes more critical when translating content that is naturally multimodal, such as picture posts on social media, audio descriptions or subtitles. MMT is especially useful when dealing with ambiguous or out-of-vocabulary words, e.g. translating **hat** into German (there is a distinction between summer **hat** and winter **hat Mütze**). Even a human translator would need to see the image to decide which word to use.

Existing work on image-based MMT [37]–[39], especially neural network approaches, often incorporates images as context either as a single, global vector representation of the whole image, or by attending to grid-based representations of different local subregions of the image. We argue that such models do not exploit images effectively for MT. A global image representation provides only a summary of the image and is expected to apply equally to the whole text, but MT operates at the word level. For attention-based models, there is a mismatch between the visual unit (equally divided grid-like image subregions) and the textual unit (a word) because the subregions may not correspond to a word or cover multiple words. This makes it hard to learn the correspondence between the textual and visual units during decoding due to a lack of visual consistency, especially when trained on small datasets; any assumed learned correspondences are also hard to interpret since the subregions are not well defined.

Our work in this section involves new referential grounding approaches to MT where the correspondences between the visual units (object regions) and textual units (source words) are better defined, and can then be used more effectively for translation (Figure 3). By **object region** we mean the depiction of the entity instance from the image as single, coherent unit. The object instance can be a concrete entity, amorphous ‘stuff’ (**sky, cloud**), or a scene (**beach, forest**). The main motivation of using objects as a visual unit is that it may potentially result in better and more interpretable grounding. As a motivational example, Figure 4 shows a case where the ambiguous word **player** can be translated correctly into a gender-marked language (female player) if its correspondence to the correct region in the image is identified.

Our main contributions in this section are:

1) An **implicit referential grounding** MT approach where the model jointly learns how to ground the source language in the object-level image representations, and to translate, while exploring training regimes with and without providing the correspondence as supervision;

2) An **explicit referential grounding** MT approach where object-level grounding is performed at the source side, independent of the translation model, and is subsequently used to guide MT, where we vary the ways...
The referential grounding models are dependent on image region annotations and their mapping to the text. We consider bounding box localisations of an object as “region”, for which we have region annotations derived from Flickr30K. In the dataset, each entity mention (noun phrase) in Flickr30K descriptions is annotated with a bounding box of the instance(s) depicted. Any entity without a bounding box is labeled as non-visual. Each entity mention is also assigned at least one of eight high-level categories (person, clothing, bodyparts, animals, vehicles, instruments, scene and others).

B. Model

1) Implicit grounding: We propose two new attention mechanisms for MMT, where grounding happens on the source language and where the process may be supervised by examples of aligned word-image region pairs.

   a) Base model: As a baseline, we experiment with the standard visual attention approach by Caglayan et al. [29] and its extension to hierarchical fusion by Libovický and Helcl [9].

   The visual representation for each object region, \( \varphi \), is a 2,048-dimensional vector generated as a non-linear transform of the last convolutional layer of a 152-layer ResNet [23] as a 14×14×1024 feature map.

   b) Source co-attention: Our first proposed object-level attention model learns to align source words to object regions and to translate them jointly.

   Let \( V = v_1, \cdots, v_m \) be the \( m \) oracle or detected object-level regions that have been cropped from the image. The visual representation for each object region, \( \varphi(v_i) \), is a 2,048-dimensional vector generated as a non-linear transform of the penultimate (pool5) layer of a 152-layer ResNet CNN.

   Given these representations, we adapt the co-attention mechanism of Lu et al. [43] to ground the source words where the model jointly learns to align these words to the image regions, and to translate them. This is done by first obtaining the affinity matrix \( A \):

   \[
   A = \tanh \left( H^\top W_a V \right)
   \]

   where \( H \in \mathcal{R}^{n \times d} \) are the encoder hidden states and \( V \in \mathcal{R}^{m \times t} \) are the object-level image representations and \( W_a \) is the bilinear parameter matrix. The image and encoder attention maps are obtained as:

   \[
   C_a = \tanh \left( W_{c_a} H + (W_{cv} V)A^\top \right)
   \]

   \[
   a^a = \text{softmax}(w_{ca}^c C_a)
   \]

   where \( a^a \) computes the source affinity. Similarly, visual affinity \( a^v \) is computed as:

   \[
   C_v = \tanh \left( W_{cv} V + (W_{cv} H)A \right)
   \]

   \[
   a^v = \text{softmax}(w_{cv}^c C_v)
   \]

   Hierarchical attention [9] is added on top of co-attention such that, at decoding time, the model jointly attends to the source context vector computed using the standard attention and the sum of the source affinity attention and the visual affinity attention from Eq 7 and Eq 8.
where grounding loss as:

\[ L_{\text{grounding}} = -\frac{1}{B} \sum_{b=1}^{B} \log(\Pr(j | \mathbf{a}^v)) \]  

(9)

where \( B \) is the number of phrases per batch and \( \mathbf{a}^v \) is from Eq 8. The loss is only active if the ground truth has an alignment; otherwise, it is set to zero.

In Figure 5 we show an example of attention weights learned for image regions (indicated by letters A-D on the grids) for a source sentence with both the unsupervised and supervised versions of the source co-attention mechanism. The supervised version clearly learns to assign the attention weights to the correct regions for each given content source word.

2) Explicit grounding: While attention is a well-established approach, for a dataset as small as ours (30K training instances), the models do not observe enough instances of similar visual representations with the same textual context for attention to be effective. The exception is supervised attention, as shown in the previous section, but it requires region annotations and their alignments to source words for training.

Here we introduce a different approach: regions and their correspondences (alignments) to words in the source sentence are identified beforehand, and then fed to the model as a way of further specifying the source words.

Previous work has explored word-level information in neural MT as morphological features [45] and as topics [46]. In both cases, every word was specified with a vector containing the additional information (e.g. POS tags). We follow a similar approach; however, our setting is more complex in that we do not have an image region associated to every word in the sentence. We experiment with different strategies for words that do not align to a region in the image, including function words, as we discuss below. As for the content of the external vector, we experiment with two types of additional information: (i) object categories, and (ii) CCA projections.

a) Object categories: The idea is to specify a word with the category of the object in the image it aligns to. We focus on nouns, which are more commonly depicted in images. Instead of using pool5 features, for visual representations we rely on the category of the objects in the image for which an alignment exists. Figure 6 shows a motivational example, where the pool5 visual representation for the two woman regions would be very different despite belonging to the same semantic category. To make the representation more flexible, instead of the category label itself, we use pre-trained word embeddings for the word representing the category. By doing so, visual representations for woman and girl would be closer than those for woman and dog, for example. We refer to this representation as \( \mathbf{E}_{\text{obj}} \).

b) CCA projections: Since the specification involves relating words to image representations, we evaluate the utility of projecting the image representation such that it is highly correlated with the word representations by using canonical correlation analysis (CCA) [47]. Formally, given paired matrices \( \mathbf{V} \) and \( \mathbf{E} \), where each row of \( \mathbf{V} \) is a visual region and its corresponding word represented by its embedding \( \mathbf{E} \), we generate a linear projection using CCA. We then use these projections to obtain transformed representations of \( \mathbf{V} \) as \( \mathbf{V}_{\text{cca}} \) and use them as visual features. \( \mathbf{V} \) can contain either category embeddings or pool5 representations.

For both object categories and CCA projections, for unaligned words we specify them with an empty vector or with the vector containing pre-trained word embeddings of the word itself. We experiment with specifying every single word in the phrase for multi-word alignments, or specifying the head nouns only. We explore two methods for specifying visual information for words: concatenation and projection.

3) Concatenation: The source word embedding is specified with region-grounded information via concatenation:

\[ \hat{\phi}(s_i) = [\phi(s_i); \phi(r)] \]  

(10)

where, \( \phi(s_i) \) is the source word embedding and \( \phi(r) \) is the object-level region information (category label embedding or CCA projection). These are the initial representations of the words for the encoder bidirectional recurrent units.

4) Projection: Alternatively, we learn a linear projection \( W \) over the region-grounded information:

\[ \hat{\phi}(s_i) = \phi(s_i) + W \phi(r) \]  

(11)

C. Experimental results

We build attention-based sequence-to-sequence models [32] with bidirectional recurrent neural networks with gated re-
current units [48] as the encoder and decoder. We use the \texttt{nmtpytorch} tool [31] with the following settings: early stop by Meteor \cite{denkowski-moses:2014:EACLshort}, selection of best model according to Meteor, beam size = 6, batch size = 64, Adam as optimizer, word embedding dimensionality = 256, and no sub-word units (they do not improve performance in our case).

For category embeddings and CCA representations we use \texttt{fasttext} 300-dimensional pre-trained word embeddings \cite{bojanowski2017enriching}. In the results reported for explicit alignments we specify only head nouns for which an alignment exists to a region in the image, and use the pre-trained embeddings of the words themselves for the remaining words.

1) MMT results: Table IV summarises the results for the following models, using BLEU \cite{papineni2002bleu} and Meteor \cite{denkowski-moses:2014:N14-1}, where the latter is the official metric used for this task (following the WMT16 shared tasks):

- \textbf{Text-only}: NMT baseline without visual information.
- \textbf{SubrAttention}: Visual attention over image subregions at decoding time (Section IV-B1a) with hierarchical fusion.
- \textbf{CoAttention}: Co-attention over image regions (\texttt{pool5} features for objects) and source words (Section IV-B1b).
- \textbf{SupCoAttention}: Supervised co-attention over (\texttt{pool5} image region features for objects) and source words (Section IV-B1c).
- \textbf{ExplicitProj}: Projection of category embedding $E_{\text{obj}}$ (Section IV-B2b).
- \textbf{ExplicitConc}: Concatenation of category embedding $E_{\text{obj}}$ and learned word embeddings (Section IV-B3).
- \textbf{ExplicitCCA}: Concatenation of $V_\text{cca}$ (\texttt{pool5} object features) and learned word embeddings (Section IV-B3).

The results in Table IV show that the proposed multimodal models outperform text-only counterparts as well as the standard multimodal approach SubrAttention for EN-CS and EN-FR. As it has been shown in the WMT shared tasks on MMT \cite{karpathy2014visual,wu2016visual}, automatic metrics often fail to capture nuances in translation quality such as the ones we expect the visual modality to help with, which – according to human perception – lead to better translations. This may be particularly the case for EN-DE, where rich morphology and compounding may result in better translations, even though these do not match the reference sentences.

### Table IV: Comparison of models using oracle object annotations and alignments, according to METEOR. Results are average of three runs with different seeds. The first row indicates the best system for EN-DE, the only pair tested on this test set at WMT16 [37].

<table>
<thead>
<tr>
<th>Systems</th>
<th>EN-CS</th>
<th>EN-DE</th>
<th>EN-FR</th>
</tr>
</thead>
<tbody>
<tr>
<td>Best WMT16</td>
<td>28.90</td>
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<td>74.09</td>
</tr>
<tr>
<td>Text-only</td>
<td>28.84</td>
<td>55.45</td>
<td>73.31</td>
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<td>CoAttention</td>
<td>30.37</td>
<td>57.15</td>
<td>\textbf{75.85}</td>
</tr>
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<td>SupCoAttention</td>
<td>30.34</td>
<td>56.48</td>
<td>75.10</td>
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<tr>
<td>ExplicitProj</td>
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<td>57.05</td>
<td>75.02</td>
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<tr>
<td>ExplicitConc</td>
<td>\textbf{30.61}</td>
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<td>ExplicitCCA</td>
<td>30.52</td>
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</tbody>
</table>

2) Lexical ambiguity evaluation: To deal with the weaknesses of the automatic metrics above, we also evaluate systems using Lexical Translation Accuracy (LTA) \cite{terlevich-moses:2013:WMT} following the methodology used at the WMT16 shared task on MMT \cite{wu2016visual}. LTA measures how accurately a system translates a subset of ambiguous words found in the Multi30K corpus. A word is said to be ambiguous in the source language if it has multiple translations (as given in the Multi30K training corpus) with different meanings. A lexical translation is considered correct if it matches exactly the (lemmatised) word aligned to it on the reference test set. The test set of 1,000 sentences contains 1,708 such words for EN-DE, 1,298 for EN-FR, and 249 for EN-CS. Table V shows that all multimodal models are better than their text-only counterpart.

### Table V: Comparison of models using oracle object annotations and alignments, according to LTA.

<table>
<thead>
<tr>
<th>Model</th>
<th>EN-CS</th>
<th>EN-DE</th>
<th>EN-FR</th>
</tr>
</thead>
<tbody>
<tr>
<td>Text-only</td>
<td>10.44</td>
<td>37.00</td>
<td>53.62</td>
</tr>
<tr>
<td>SubrAttention</td>
<td>10.84</td>
<td>37.82</td>
<td>53.62</td>
</tr>
<tr>
<td>CoAttention</td>
<td>12.45</td>
<td>38.06</td>
<td>\textbf{55.16}</td>
</tr>
<tr>
<td>SupCoAttention</td>
<td>13.25</td>
<td>37.47</td>
<td>\textbf{55.16}</td>
</tr>
<tr>
<td>ExplicitProj</td>
<td>13.65</td>
<td>\textbf{38.41}</td>
<td>54.08</td>
</tr>
<tr>
<td>ExplicitConc</td>
<td>12.85</td>
<td>38.06</td>
<td>53.78</td>
</tr>
<tr>
<td>ExplicitCCA</td>
<td>\textbf{14.06}</td>
<td>38.17</td>
<td>54.08</td>
</tr>
</tbody>
</table>

3) Oracle versus predicted regions: Thus far we showed results where the oracle bounding boxes and object-word alignments are used. In the implicit grounding models this is not a major issue given that the alignments are only needed at training time. For the explicit grounding models, however, this information is also needed at test-time. Therefore, we also investigate using predicted objects and object-word alignments \cite{chen2017iterative}.

The results indicate that there are no significant differences in performance.

D. Discussion

We proposed referential grounding approaches for MMT that use clearly defined correspondences between a source word and an object in the image to guide translation. We showed that MMT models using such groundings at object-level can better exploit image information, leading to better performance, especially when translating challenging cases such as ambiguous words.

V. Summarization

All videos in the \texttt{How2} dataset are accompanied by a manually written summary that should attract the attention of viewers and increase the chance of the video being found in a keyword search. The goal of the summarization task on this dataset is to generate this type of video summary. An example video summary is shown in Figure 7.

A. Characteristics of the summaries

In order to get a reliable estimate of the summarization quality, we use a different split than for ASR and MT.

\footnotetext[2]{We use the \texttt{w2v-max} and \texttt{union} model described in their paper.}
today we are going to show you how to make spanish omelet. i'm going to dice a little bit of peppers here. i'm not going to use a lot, i'm going to use very very little. a little bit more then this maybe. you can use red peppers if you like to get a little bit color in your omelet. some people do and some people don't. t is the way they make there spanish omelets that is what she says. i loved it, it actually tasted really good. you are going to take the onion also and dice it really small. you do n't want big chunks of onion in there cause it is just pops out of the omelet. so we are going to dice the up also very very small. so we have small pieces of onions and peppers ready to go.

The standard splits contain enough text for sentence-level evaluation; however, there is only one summary per video. We use 73,993 videos for training, 2,965 for validation and 2,156 for testing. The average length of transcripts is 291 words and the average length of summaries is 33 words.

B. Baseline methods

a) Language input: For text-based input, we use the transcripts of the videos. We leverage the speech modality by using the outputs from a pre-trained speech recognizer trained with other data, as inputs to a text summarization model. We use state-of-the-art models for distant-microphone conversational speech recognition, ASpIRE [54] and EESEN [55], [56]. The word error rate of these models on the How2 test data is 35.4%. This high error mostly stems from normalization issues in the data. For example, recognizing and labeling “20” as “twenty” etc. We accept these as-is for this task. Also, note that this is the WER on the larger 2000-hour corpus rather than 300-hour subcorpus.

b) Visual input: We represent videos by a sequence of 2048-dimensional action feature vectors (see Section II).

C. Models

We study various summarization models. First, we use a Sequence-to-Sequence (S2S) model [57] consisting of an encoder RNN to encode (text or video features) with the attention mechanism [32] and a decoder RNN to generate summaries. Our second model is a Pointer-Generator (PG) model [58], [59] that has shown strong performance for abstractive summarization [60], [61]. As our third model, we use hierarchical attention approach [9] originally proposed for multimodal machine translation to combine textual and visual modalities to generate text. This model first computes the context vector independently for each of the input modalities (text and video). In the next step, the context vectors are treated as states of another encoder, and a new vector is computed. When using a sequence of action features instead of a single averaged vector, the RNN layer helps capture context. In Figure 8, we present the building block of our models.

D. Evaluation

To evaluate the generated summaries we use ROUGE-L [62], a standard metric for abstractive summarization that measures the longest common sequence between the reference and the generated summary. Additionally, we introduce a new metric Content F1 that fits the template-like structure of the summaries observed in our dataset.

a) Content F1: We compute the F1 score of the content words in the summaries based over a monolingual alignment obtained using METEOR toolkit [63]. Then, we remove function words and task-specific stop words that appear in most of the summaries from the reference and the hypothesis. These stop words (how, learn, tips, free, etc.) are very frequent in the reference summaries making it easy for the decoder to predict these and thus increase the ROUGE score. We treat the remaining words from the reference and the hypothesis as two bags of words and compute the F1 score over the alignment. Note that the score ignores the fluency of output in line with recently proposed metrics such as HighRes [64].

b) Human Evaluation: In addition to automatic evaluation, we also evaluate system summaries by eliciting human judgments. Following the abstractive summarization human annotation work of Grusky et al. [65], we ask our annotators to label the generated output on a scale of 1–5 on metrics.
of informativeness, relevance, coherence, and fluency. We perform this on randomly sampled 500 videos from the test set. We evaluate three models: two unimodal (text-only, 5a; video-only 6-7) and one multimodal (text-and-video, 8). Three workers annotated each video on Amazon Mechanical Turk.

E. Output Examples from Different Models

Table VII shows the example outputs from our different text-only and text-and-video models. The text-only model produces a fluent output which is close to the reference. The action features with the RNN model, which sees no text in the input, produces an in-domain (“fly tying”’ and “fishing”) abstractive summary that involves more details like “equipment” which is missing from the text-based models but is relevant. The next neighbor model is related to “knot tying” but not related to “fishing”. The scores for each of these models reflect their respective properties. Observing other outputs of the model, we noticed that although predictions were usually fluent and thus getting high ROUGE scores, there is a large room for improvement by predicting all details from the ground truth summary, like the subtle selling point phrases, or by using the visual features in a different adaptation model.

In Table VIII, we report human evaluation scores of the best text-only, video-only, and multimodal models. We observe that text-only summaries dominate on relevance but multimodal models are the most informative, coherent and fluent, indicating that these models can fuse complementary information from multiple modalities to generate relevant summaries. The example presented in Table VII shows how the generated summaries vary with different models and features.

Our parallel work [66], [67] demonstrates the use of our summarization models trained in this work for a transfer learning-based summarization task on the Charades dataset [68], which has audio, video, and text (summary, caption, and question-answer pairs) modalities just like the How2 dataset. Pre-training and transfer learning with the How2 dataset led to significant improvements in unimodal and multimodal adaptation tasks on the Charades dataset.

VI. CORRELATION-BASED UNSUPERVISED LEARNING

All machine learning involves learning representations on top of the input features [69]. In deep learning, representation is learned implicitly, as a result of finding a local minimum of a loss function. In contrast to this implicit representation learning stand several explicit representation learning paradigms [70]–[72]. How to best exploit multiple views is an open problem, especially when there is a latent alignment between views, such as between an image and its spoken caption [73]. We treat the How2 dataset as a 4-way parallel corpus, and explore an advanced, correlation-based representation learning objective.

A. Deep Generalized Canonical Correlation Analysis

It has been shown that the availability of a second view in addition to a primary input can help with any task. For instance, the video stream of a speaker’s face, in addition to the audio recording, helps perform speech recognition [74]. This qualitative result is still true when the second view is reconstructed from the primary input by a trained predictor. However, in some cases, it may be difficult to learn such a predictor, as in the speech recognition example above. Instead of reconstructing the secondary view, it is simpler to learn a representation for each view that is maximally reconstructive of the representations learned for the other views [72], [75]. This intuition was first formalized as Canonical Correlation Analysis (CCA) [47], extended to pairs of views [76] and arbitrary feature extractors [77]. We use the formulation of [78], [79] which we describe next.

For each view \( j \in \{1..J\} \), all \( N \) points of the dataset are stored in a matrix \( X_j \in \mathbb{R}^{d_j \times N} \), where \( d_j \) is the dimensionality of the feature vector. We denote \( f_j : \mathbb{R}^{d_j} \rightarrow \mathbb{R}^{h_j} \) the \( j \)-th learned feature extractor — in our case a neural net — and \( U_j \in \mathbb{R}^{h_j \times k} \) a linear transformation matrix. The \( \{f_j\}_j \) and \( \{U_j\}_j \) are trained jointly to reconstruct an unknown shared representation, under constraints, resulting in the following problem:

\[
\text{minimize } \sum_{j=1}^{J} \| G - U_j^T f_j(X_j) \|_2^2 \text{ subject to } GG^T = I_k,
\]

with respect to parameters \( \{G, \{f_j, U_j\}_j\} \). Here, \( G \in \mathbb{R}^{k \times N} \) can be viewed as the learned representation for the dataset, and \( k \) is the dimensionality of said representation. The constraint on \( G \) prevents trivial solutions. Note that each learned feature extractor \( (f_j, U_j) \) tries to reconstruct \( G \) from \( X_j \). We refer to this method as deep generalized CCA (DGCCA). Deep CCA [77] is equivalent to the case \( J = 2 \).

B. Experiments and Results

Within the framework of DGCCA, we use the How2 dataset as a 4-way parallel corpus: video, speech, transcription in English, translation in Portuguese. Each data point in that corpus corresponds to one utterance. For the text

<table>
<thead>
<tr>
<th>Method</th>
<th>ROUGE-L</th>
<th>Content F1</th>
</tr>
</thead>
<tbody>
<tr>
<td>Naive baselines</td>
<td></td>
<td></td>
</tr>
<tr>
<td>1 Language Model sampling</td>
<td>27.5</td>
<td>8.3</td>
</tr>
<tr>
<td>2a Rule-based Extractive summary</td>
<td>16.4</td>
<td>18.8</td>
</tr>
<tr>
<td>2b Next-neighbor Summary</td>
<td>31.8</td>
<td>17.9</td>
</tr>
<tr>
<td>Text-only models</td>
<td></td>
<td></td>
</tr>
<tr>
<td>3 S2S on 2a only</td>
<td>46.4</td>
<td>36.0</td>
</tr>
<tr>
<td>4 S2S on 200 tokens of Transcript</td>
<td>40.3</td>
<td>27.5</td>
</tr>
<tr>
<td>5a S2S on Transcript</td>
<td>53.9</td>
<td>47.4</td>
</tr>
<tr>
<td>5b PG on Transcript</td>
<td>50.2</td>
<td>42.0</td>
</tr>
<tr>
<td>5c S2S on ASR</td>
<td>46.1</td>
<td>34.7</td>
</tr>
<tr>
<td>Video-only models</td>
<td></td>
<td></td>
</tr>
<tr>
<td>6 AF only</td>
<td>38.5</td>
<td>24.8</td>
</tr>
<tr>
<td>7 RNN over AF</td>
<td>46.3</td>
<td>34.9</td>
</tr>
<tr>
<td>Multimodal models</td>
<td></td>
<td></td>
</tr>
<tr>
<td>8 Transcript + AF w/ Hier. Attn</td>
<td>54.9</td>
<td>48.9</td>
</tr>
<tr>
<td>9 ASR + AF w/ Hier. Attn.</td>
<td>46.3</td>
<td>34.7</td>
</tr>
</tbody>
</table>
and speech modalities, we use encoder-decoder sequence-
to-sequence models trained on the How2 dataset to extract the \( \{ x_j \} \) features. We average either the encoder-side embeddings or the sequence of context vectors to obtain a single vector for each sequence, following [80]. For the video modality, we first break up the videos into keyframes, then average the outputs of a ResNet [23] over the time window corresponding to a given utterance. We thus obtain a single vector representing the video modality for each utterance.

1) Retrieval experiments: We start with an intrinsic evaluation of our learned representations. We use a retrieval task to probe the reliability of the learned embedding space. Given a source point \( u \), we return the 10 closest points within a reference set \( \{ u_i \} \). The source and reference points come from different views of the dev and test sets of the How2 dataset. This allows us to score the retrieval based on whether the correct point is within the 10 closest points, and we report this as Recall@10. Picking the 10 closest points at random results in a Recall@10 of 0.5% for the dev set and 0.4% for the test set. Using our DGCCA model, retrieving the 10 closest points involves projecting the source point and the reference set into the shared space, computing pairwise distances (we use mean-centered cosine distance) and taking the 10 closest points.

To validate the approach, we compare linear and deep CCA on pairs of modalities. Linear CCA corresponds to \( f_j \) being set to the identity mapping for all \( j \). We report retrieval results in Table IX. With the exception of speech-to-text retrieval, deep CCA performs systematically better than linear CCA.

We train models on 3 and 4 modalities, and report retrieval scores in Tables X and XI. In both cases, \( k = 160 \). When adding modalities, we note that retrieval scores decrease, since the model needs to accommodate additional views. Some retrieval scores are higher than others; most likely, the model trades off higher scores for easier pairs of views (e.g. Portuguese text and English text) against lower scores for harder pairs of views (e.g. video and speech). This could be compensated by adding weights \( \{ w_j \} \) for each reconstruction loss, or by tuning the architectures of the \( \{ f_j \} \) separately.

Overall, retrieval scores between language modalities are high, ranging from 71.0% to 98.4%. There are several reasons which could explain the lower scores involving the video modality. First, it is not quite clear how much temporal coherence the video modality has in the How2 dataset. For instance, objects mentioned by the speaker might appear much later in the video, very briefly, or not at all. Further, ResNet features might not be able to adequately represent the domain of the How2 dataset. We experimented with representations from action networks [21] trained on an action dataset [22], and obtained similar results. Most likely, given the noisy input features, our models lack either the expressive power or a sufficient amount of training data to capture the correspondence between the language modalities and the video [73], [81].

2) Scoring top-1 retrieval results: Given our high retrieval scores between language modalities, we attempt to measure their performance with conventional ASR, MT and ST metrics — WER and BLEU scores. For each data point in the test set, we retrieve the closest point from a reference set, and
TABLE XI: Recall@10 for retrieving column modality given source row modality, for a DGCCA model trained on 4 views. Results from the bottom left triangle can be compared to those in Table IX.

<table>
<thead>
<tr>
<th></th>
<th>Text (pt)</th>
<th>Text (en)</th>
<th>Speech</th>
<th>Video</th>
</tr>
</thead>
<tbody>
<tr>
<td>Text (pt)</td>
<td>-</td>
<td>98.8</td>
<td>73.5</td>
<td>2.1</td>
</tr>
<tr>
<td>dev</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>test</td>
<td>98.3</td>
<td>-</td>
<td>71.0</td>
<td>1.1</td>
</tr>
<tr>
<td>Text (en)</td>
<td>98.8</td>
<td>-</td>
<td>88.2</td>
<td>1.4</td>
</tr>
<tr>
<td>dev</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>test</td>
<td>98.4</td>
<td>-</td>
<td>85.4</td>
<td>0.9</td>
</tr>
<tr>
<td>Speech</td>
<td>73.0</td>
<td>88.1</td>
<td>-</td>
<td>1.1</td>
</tr>
<tr>
<td>dev</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>test</td>
<td>70.7</td>
<td>85.4</td>
<td>-</td>
<td>1.0</td>
</tr>
<tr>
<td>Video</td>
<td>2.1</td>
<td>1.1</td>
<td>1.0</td>
<td>-</td>
</tr>
<tr>
<td>dev</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>test</td>
<td>1.1</td>
<td>1.1</td>
<td>0.9</td>
<td>-</td>
</tr>
</tbody>
</table>

use it as the output hypothesis of either an MT, ASR or ST model, which can then be scored with the relevant metric. If using the test set as a reference set, given the high retrieval scores, the WER or BLEU scores would be almost perfect. We thus report two more challenging settings in Table XII: the reference set can be either the train set, or the union of the train set and the test set. As compared to the baseline sequence-to-sequence neural model, our models perform reasonably well, and are consistent with our retrieval scores: MT works best, then ASR, then ST. When the reference set is the train set, the scores drop considerably, also because the train set does not necessarily contain adequate sentences. To quantify this, we pick, for each target sentence from the test set, the closest sentence from the train set in terms of edit distance, which yields a BLEU of 10.6 and a WER of 63.0%.

TABLE XII: Scoring top-1 retrieval result from DGCCA models with ASR, MT and ST metrics. Models used (from left to right) were trained using speech and text (en); text (en) and text (pt); speech, text (en), text (pt) and video. Source sentences for the retrieval are from the test set.

<table>
<thead>
<tr>
<th>Reference Set</th>
<th>WER</th>
<th>BLEU (MT)</th>
<th>BLEU (ST)</th>
</tr>
</thead>
<tbody>
<tr>
<td>train</td>
<td>134%</td>
<td>5.2</td>
<td>0.2</td>
</tr>
<tr>
<td>train + test</td>
<td>27.4%</td>
<td>80.7</td>
<td>19.8</td>
</tr>
<tr>
<td>Baseline S2S</td>
<td>24.3%</td>
<td>57.3</td>
<td>27.9</td>
</tr>
</tbody>
</table>

C. Discussion

We framed the How2 dataset as a multiview representation learning problem, and probed the quality of the learned representations using intrinsic evaluations. While our results show it is possible to learn high-quality representations on the language modalities, the video modality remains a major challenge, possibly calling for specialized architectures or transfer learning. Further integrating the learned representations into supervised tasks is left for future work.

VII. CONCLUSION

This paper describes (1) the How2 dataset, a collection of large-scale open-domain user-generated instructional videos, and (2) a detailed study of different multi-modal learning experiments on this dataset or other proxy datasets like Multi30K for MT. This corpus brings together English audio, English transcripts, Portuguese transcripts, videos, and summaries, along with meta-data such as topic of the video. This makes the How2 dataset a good resource for research at the intersection of vision, language and speech. By releasing this dataset, we hope to enable research on multi-lingual, multi-modal, highly correlated and well-aligned parallel modalities. We presented numerous uni-, multi- and cross-modal tasks such as speech recognition, machine translation, summarization, and multi-view representation learning. With this study, we hope to shed light on the current state of vision, language and speech grounding and to help researchers with designing new tasks in this space.

APPENDIX

A. The How2 Dataset

Figures 9 show the LDA topic distribution and segment length analysis of the 300h subset of the How2 dataset.

B. Region-Specific Multimodal Machine Translation

Table XIII shows qualitative examples for results presented in Section IV.
C. Correlation-based Multiview Learning

1) Feature Extraction: We use baseline ASR and MT models from Sections III and IV. For each input sequence, the encoder produces a corresponding sequence of feature vectors \( h_1, \ldots, h_T \). We use \( \frac{1}{T} \sum_{t=1}^{T} h_t \) to represent that input sequence. The decoder with attention produces a sequence of context vectors \( c_1, \ldots, c_S \), and we use \( \frac{1}{S} \sum_{i=1}^{S} c_i \) to represent the target sequence. Since we use word-based ASR and MT systems, each \( c_j \) and \( h_i \) roughly represents a word in context. For the video modality, we first break up the videos into keyframes, then use a ResNet [23] to map each keyframe to a multi-class posterior, based on the 1000 ImageNet classes. For each speech utterance, we then compute the average of the posteriors corresponding to the time window of the speech utterance. The averaging process is meant to capture the most persistent predictions and reduce the variability due to noise. We thus obtain a single vector representing the video modality for each utterance. As a result, for text, speech and video, the context vectors \( c_j \) features are 320-, 800- and 1000-dimensional, respectively.

2) Models and Training: The features described above are kept fixed, while we use feed-forward neural networks with 2 hidden layers and tanh non-linearities for the \( \{ f_j \} \). The first layer has the same dimensionality as the input, and the second layer the same size as \( k \). To avoid under-defining the objective, \( k \) should be no larger than the smallest of the \( \{ h_j \} \).

We set \( k \) to half the smallest of the \( \{ h_j \} \) involved, as a heuristic to retain most of the informative components and discard uninformative ones. For numerical stability, we add the identity matrix scaled by \( 10^{-16} \) to all the view-specific covariance matrices. We use stochastic gradient descent with batch size 5500 and Adam optimizer with default parameters. The analytical expression of the gradient was taken from [78]. In the experiments involving video, we use a weight decay of \( 10^{-5} \). After each full pass over the training set, we measure retrieval scores between all possible pairs of different views on the dev set, using the highest of these scores to measure the performance of our model. We use this score to do early stopping.

ACKNOWLEDGEMENTS

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### TABLE XIII: Qualitative examples comparing text-only NMT and multimodal models. We show the source (SRC), text-only MT (NMT) and a multimodal model (MMT). In both cases we also show the back translation into English for clarity. Underlined words represent translation errors, while bold face words, the correct (or better) version.

<table>
<thead>
<tr>
<th>EN-FR</th>
<th>SRC: A man on a tag line going into the water.</th>
<th>NMT: Un homme sur une ligne de métro en train de marcher dans l’eau.</th>
<th>MMT: Un homme sur une ligne de sable allant dans l’eau.</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>(A man on the metro line walking to the water.)</td>
<td>(A man on the sand line going into the water.)</td>
<td></td>
</tr>
<tr>
<td>EN-DE</td>
<td>SRC: A large group of people of various ages and genders sit outside together.</td>
<td>NMT: Un grand nombre de personnes de différents âges et des accessoires sont assis ensemble.</td>
<td>MMT: Un grand nombre de personnes de différentes âges et d’autres sont assis ensemble.</td>
</tr>
<tr>
<td></td>
<td>(A large number of people of different ages and accessories sit together.)</td>
<td>(A large number of people of different ages and others sit together.)</td>
<td></td>
</tr>
<tr>
<td></td>
<td>SRC: A fox terrier leaps after a ball.</td>
<td>NMT: Ein metzger springt nach einem ball.</td>
<td>MMT: Ein terrier springt nach einem ball.</td>
</tr>
<tr>
<td></td>
<td>(A butcher jumps for a ball.)</td>
<td>(A fox terrier jumps for a ball.)</td>
<td></td>
</tr>
</tbody>
</table>
REFERENCES


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