Automated assessment of second language comprehensibility: Review, training, validation, and generalization studies

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Abstract

Whereas many scholars have emphasized the relative importance of comprehensibility as an ecologically valid goal for L2 speech training, testing, and development, eliciting listeners’ judgments is time-consuming. Following calls for research on more efficient L2 speech rating methods in applied linguistics, and growing attention toward using machine learning on spontaneous unscripted speech in speech engineering, the current study examined the possibility of establishing quick and reliable automated comprehensibility assessments. Orchestrating a set of phonological (maximum posterior probabilities and gaps between L1 and L2 speech), prosodic (pitch and intensity variation), and temporal measures (articulation rate, pause frequency), the regression model significantly predicted how naïve listeners intuitively judged low, mid, high, and nativelike comprehensibility among 100 L1 and L2 speakers’ picture descriptions. The strength of the correlation (r = .823 for machine vs. human ratings) was comparable to naïve listeners’ interrater agreement (r = .760 for humans vs. humans). The findings were successfully replicated when the model was applied to a new dataset of 45 L1 and L2 speakers (r = .827) and tested under a more freely constructed interview task condition (r = .809).

Introduction

Adult second language (L2) speech is generally foreign-accented due to a range of factors, such as the influence of first language (L1) phonetic systems (Flege & Bohn, 2021), perceptual-cognitive aptitude (Saito, 2017), and identity (Sung, 2016). Thus, many scholars have emphasized the importance of setting realistic goals for adult L2 learners, prioritizing understanding over nativelikeness (Munro & Derwing, 1995). There is ample evidence that many L2 speakers are perceived as sufficiently comprehensible regardless of foreign accentedness (Isaacs & Trofimovich, 2012); and L2 speakers can continue to enhance aspects of speech affecting comprehensibility as long
as they use the target language, receive feedback, and strive to improve with a view of successful communication (Saito & Akiyama, 2017). Although teachers play an important role in providing immediate feedback, helping students understand their own comprehensibility, and promoting autonomous L2 speech learning, such a resource (provision of feedback) is limited in foreign language classrooms (Muñoz, 2014). In the field of speech engineering, there is a growing amount of research attention toward the development of more robust automatic recognition of not only controlled but also spontaneous L2 speech, and the application of the technology to L2 speech training and evaluation (Fu et al., 2020). Interfacing perspectives from education and speech sciences, we took a first step toward training, validating, and generalizing the automatic assessment of comprehensibility in the context of 190 L1 and L2 speakers of English.

Second Language Comprehensibility

Given that technology allows people to interact worldwide regardless of physical constraints using videoconferencing tools and social networking, attaining adequate L2 speech proficiency is considered a key skill in academic, business, and social settings. On the one hand, some L2 users strive to attain nativelike proficiency (Scales et al., 2006) and traditional teaching syllabi highlight native speakers as an ideal instructional model (Foote et al., 2011). On the other hand, research has convincingly shown that few postpubertal L2 learners can attain nativelike phonological accuracy and fluency as their L2 system builds on and thus inevitably interacts with their already-developed first language system (Flege & Bohn, 2021). When it comes to English, attaining nativelike phonological proficiency is arguably unnecessary as most interactions take place between L2 users (Pennycook, 2017). Thus, a number of scholars have pointed out the importance of setting more realistic goals for postpubertal L2 speech learning, such as the enhancement of comprehensibility rather than the attainment of nativelike proficiency (Munro & Derwing, 1995).

From a methodological point of view, simulating behaviors in real-life conversation, L2 comprehensibility is operationalized as listeners’ intuitive judgments of spontaneous L2 speech on a 9-point scale (1 = difficult to understand, 9 = easy to understand). Comprehensibility is thought to represent the amount of listener effort necessary to understand the speakers’ message despite the degree of foreign accentedness; and the process but not product of listeners’ understanding (for the further discussion on “intelligibility” rather than “comprehensibility” as a barometer of actual understanding and its terminological and methodological issues, see Levis, 2018). According to Derwing and Munro’s seminal work, accented L2 speech can be perceived to be comprehensible as certain phonological errors do not hinder listeners’ understanding (e.g., Munro & Derwing, 1995). Since Munro and Derwing (1995), there have been a number of follow-up studies showing that listeners’ understanding is negatively influenced by certain (but not all) phonological errors, such as the mispronunciation of segmentals with high functional load (Suzukida & Saito, 2019), melodic inaccuracies (Suzuki & Kormos, 2020; for a meta-analysis of the phonological correlates of L2 comprehensibility, see Saito, 2021). Additionally, certain listeners likely assign higher and thus more lenient comprehensibility ratings when they have more familiarity with foreign accents (Kennedy & Trofimovich, 2008), linguistic training (Saito et al., 2017), pedagogical experience (Isaacs & Thomson, 2020), and a greater level of awareness of the importance of L2 comprehensibility (rather than accentedness) (Saito et al., 2019).
From a theoretical perspective, it is important to note that comprehensibility can serve as an index of adult L2 speech development. As stated in the interaction account of L2 acquisition (Mackey, 2012), language learning takes place precisely when L2 speakers actively participate in conversational interactions and end up in communication breakdowns due to linguistic errors. L2 speakers work together with interlocutors when comprehensibility is not sufficient by relying on a range of negotiation-for-meaning behaviors, such as clarification requests, and comprehension and confirmation checks. This whole sequence is hypothesized to help L2 speakers become more comprehensible, functional, and proficient users of the target language (for an empirical evidence, see Saito & Akiyama, 2017).\textsuperscript{1} According to longitudinal (Derwing & Munro, 2013) and cross-sectional investigations (Saito, 2015a), L2 learners tend to show quick improvements in comprehensibility and accentedness within the first few years of immersion. Whereas attaining nativelike L2 speech proficiency may be limited to certain individuals with earlier ages of acquisition (Saito, 2015b), linguistically similar L1 backgrounds (e.g., Bongaers et al., 1997), and/or special language aptitude (e.g., Hu et al., 2013 for phonemic coding; Kachlicka et al., 2019 for perceptual acuity), many L2 learners can continue to enhance the comprehensibility aspects of their speech over time as long as they use a target language on a regular basis. Similar learning patterns have been observed in various foreign language classroom settings (e.g., Nagle, 2018).

Due to the significant amount of practical and theoretical relevance, many scholars have promoted comprehensibility as an ecologically valid, more realistic target for adult L2 speech assessment not only in high-stakes testing settings but also for practitioners (Isaacs et al., 2017). However, eliciting listeners’ comprehensibility judgments is a time-consuming task (e.g., about 1 hour for rating 70 samples in Derwing & Munro, 2013). To this end, where some researchers use shorter stimuli to avoid listener fatigue (e.g., 4.5–10.5 seconds in Derwing & Munro, 1997), others recommend collecting L2 rating

\textsuperscript{1}An interaction account of L2 acquisition would indicate that language development occurs when communication breaks down and interlocutors are required to negotiate for meaning (Mackey, 2012). In this article and elsewhere (e.g., Saito, 2021), therefore, it has been argued that comprehensibility is an index of L2 development. Negotiation-for-meaning episodes likely take place when L2 speakers’ comprehensibility is low. Through more interaction and negotiation-for-meaning opportunities, learners with low L2 oral proficiency are pushed to work on the comprehensibility of their speech, even though they still sound foreign-accented. In this regard, it is reasonable to say L2 speech acquisition takes place on the continuum of comprehensibility as learners’ speech becomes increasingly easier to understand despite their detectable foreign accentedness. To provide empirical support, Derwing and Munro’s (2013) longitudinal dataset showed that the comprehensibility of L2 speakers continued to improve as a function of increased immersion for an extensive period (7 years) while their speech remained foreign-accented throughout (cf. Saito, 2015a for cross-sectional evidence; Saito & Akiyama, 2017 for a long-term training study). A reviewer pointed out that what matters is intelligibility rather than comprehensibility. Research has indeed shown that speech can be low in comprehensibility yet still intelligible (i.e., even if difficult to understand, speech can be accurately understood). Technically, it should only be speech that is low in intelligibility (which likely entails low comprehensibility) that would initiate a degree of negotiation of meaning. However, we would like to argue that negotiation for meaning can be triggered by comprehensibility and intelligibility alike. In the case of low-level comprehensible but still intelligible speech, interlocutors likely initiate a range of particular interactional moves such as clarification requests and repetition, both of which are termed as “negotiation for meaning” strategies (Mackey, 2012). To our knowledge, few studies have ever explored precisely when negotiation for meaning occurs in accordance with different levels of interlocutors’ understanding, i.e., high versus low comprehensibility and high versus low intelligibility. This should be considered as an intriguing direction for future research.
data using online platforms to reduce the burden on researchers (e.g., Nagle & Rehman, 2021 for Amazon Mechanical Turk). Therefore, due to the time-consuming nature of the assessment procedure, a growing amount of attention has been given to the idea of using automated L2 comprehensibility scoring (O’Brien et al., 2018). To this end, the current study took an exploratory approach toward examining this topic.

**Automatic Assessment of Second Language Speech**

To date, research has provided a range of pedagogic techniques that teachers can use (Pennington, 2021) and online learning tools that learners can access (Loewen et al., 2019). To make the most of such opportunities, provision of online feedback plays a critical role for many reasons. First and foremost, learners need to understand the current state of their L2 proficiency, compare it with their target, and make an effort to fill in the gaps accordingly (Lyster & Saito, 2010). Without such adequate self- and teacher-assessments, learners may either overestimate or underestimate their own L2 proficiency and struggle to decide the best course of action, slowing down acquisitional processes (Trofimovich et al., 2016).

Although feedback is integral to successful L2 speech learning, the provision of feedback is a resource-intensive task, especially in foreign language classrooms. Not only are students’ L2 practice opportunities limited to several hours of weekly classroom instruction (which is typically devoid of conversation activities; Nishino & Watanabe, 2008) but teachers also lack ample time to record, listen to, and provide feedback on each student’s speech, given that they often teach many students at one time (Muñoz, 2014). Thus, many researchers and practitioners alike are interested in the introduction of automated assessment as a pedagogical tool for optimal L2 speech education (O’Brien et al., 2018).

As summarized in Table 1, a range of studies have been conducted to examine the extent to which automated scoring can simulate human listeners’ assessments of L2 oral proficiency. In terms of method, scholars have primarily drawn on correlational analyses. First, L2 speech stimuli were collected using a range of speaking tasks (controlled, spontaneous). Then, the stimuli were evaluated by trained and untrained human listeners for various aspects of L2 oral proficiency rubrics (e.g., pronunciation accuracy, overall speaking proficiency, perceived fluency and comprehensibility). Finally, the stimuli were submitted to a range of automated speech measures (phonological and fluency analyses). To examine the potential of automated L2 speech assessment, researchers explored the strength of the associations between human L2 oral proficiency ratings and automated speech scores. With respect to automated speech measures, there is much methodological variation in the primary studies. Although each study adopted different measures, they could be roughly categorized into three subgroups with a view of methodological syntheses—that is, fluency, melodic, and phonological measures. Relevant citations in the text that follows derive from Table 1 (i.e., a summary of automated L2 speech assessment studies).

**Fluency Measures**

This subcategory refers to a set of outcome measures that tap into the temporal characteristics of speech. As stated in Tavakoli and Skehan’s (2005) model, such fluency features can be categorized as speed (e.g., speech and articulation rate), breakdowns (e.g., filled and unfilled pause frequency), and repairs (e.g., repetition...
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<td></td>
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<td>3,000 sentences</td>
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<td>Kang &amp; Johnson (2018)</td>
<td>Speakers • 120 L2 English speakers (intermediate to advanced) Materials • 120 monologues</td>
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<td>Fu et al. (2020)</td>
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<td>Speaking materials</td>
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<td>14 L2 English speakers (spontaneous)</td>
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<td>Shen et al. (2021)*</td>
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<td>Speakers</td>
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<td>Materials</td>
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<td>Expert judgments of fluency (9-point scale)</td>
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<td>100 picture descriptions tasks</td>
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<td>DNN acoustic model scores</td>
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<td>Machine vs. humans</td>
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<td>r = .8–.9 (spontaneous)</td>
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*Features Experiment 2.
*Serves as a pilot study for the current study (see the "Method" section).
and self-correction frequency). Whereas some studies covered all three of these aspects of fluency (e.g., Cucchiarini et al., 2000 for speed, breakdown, and repair), other studies focused on one dimension (e.g., Neumeyer at al., 2000 for speed). Within the same subgroup (e.g., breakdown), the measures were operationalized differently (e.g., Cucchiarini et al., 2000 for the frequency of silent pauses [defined as > 20ms] per sample vs. Kang & Johnson, 2018 for the frequency of silent pauses [defined as > 100ms]).

**Melodic Measures**

This subcategory refers to a set of outcome measures related to the varied use of pitch and intensity to mark stress and intonation. Some scholars have adopted different types of pitch measures, such as tone choices and relative pitch (Kang & Johnson, 2018). Others have also adopted both pitch and amplitude measures, such as pitch and intensity variation (e.g., van Santen et al., 2009). It is important to acknowledge that we do not yet know how to conduct the automated analyses of word and sentence stress accuracy. In the field of L2 speech, such analyses require linguistically trained coders to examine how each instance of word and sentence stress has been marked (with higher pitch, longer duration, and/or greater intensity) in a contextually appropriate manner (e.g., Isaacs & Trofimovich, 2012 for the manual analyses of word stress and intonation accuracy).

**Phonological Measures**

Thanks to advancements in automatic speech recognition (ASR) technology, one notable change in this line of work concerns the development, sophistication, and application of machine algorithms to calculate phonetic accuracy scores (i.e., qualitative measures; Zechner & Evanini, 2020). In ASR, speech classes are generally defined in a more sophisticated way compared to linguistically driven phonemes. Speech samples are first segmented into small frames (e.g., approximately 20 ms) and converted into spectrum (i.e., different frequencies) using a Fast Fourier Transform. Because acoustic realization of phoneme /x/ depends on its phonemic environment, /x/ is further divided into a full set of /a/–/x/+ /b/. This means acoustically realized phoneme /x/ produced after /a/ and before /b/. When the number of linguistically driven phonemes is $N$, the full set of /a/–/x/+ /b/ has $N^2$ variants. All of them are considered as the speech classes related to /x/.

In the classical ASR framework based on Hidden Markov Models (HMM), a number of spectrograms were collected from thousands of L1 speakers to build an acoustic model for each speech class. Because the spectrogram was used as a fundamental speech representation, it inevitably conveyed a range of extralinguistic factors, such as speaker, age, gender, and microphone. These factors in turn clouded the accuracy of ASR.

Recently, a more sophisticated framework for ASR has been used based on Deep Neural Networks (DNN). Various frameworks have been proposed with DNN and one of them uses posteriorgram (rather than spectrogram) as a fundamental speech representation. A posteriorgram is defined as a temporal sequence of posterior probability distributions of classes, demonstrating the probability that an input observation belongs to a particular class. Unlike spectrogram the extralinguistic factors are well suppressed. Whereas the spectrograms of “hello” generated by a male speaker and a female speaker can be substantially different, the posteriorgrams of the same speech samples can be indistinguishable. If we use the linguistically driven phonemes as the
speech classes, a posteriorgram can be viewed as a probabilistic version of its phonemic transcript (i.e., how similar the speech signals are to each phonemic class). Unlike the spectrogram (which provides an acoustic representation of speech), a posteriorgram can serve as a linguistic or phonetic representation of speech.\(^2\)

Once ASR models are established in a target language, they are applied to analyze the acoustic profiling of L2 speech data. Looking at the posteriorgram distance between L1 and L2 speech data, for example, some studies have explored the overall acoustic similarity of L2 realizations of phonemes to their L1 equivalents. For a more detailed account of posteriorgram-based L1 versus L2 phoneme categorizations, see Shen et al. (2021) and Supporting Information-B. To further increase the accuracy of ASR, another intriguing idea concerns evaluating the quality of L2 speech using L1 and L2 corpus data. For example, the gap among Japanese speakers of English in Fu et al. (2020) was assessed using the outcomes from ASR trained on L1 Japanese data and those from L2 English data. L2 English speech with low word-error rate in both L1 Japanese and L2 English ASR (the gap is nil or small), could be considered highly proficient. In contrast, if word error rate was low in L1 Japanese ASR but high in L2 English ASR (the gap is large), such samples could be considered less proficient.

Overall Findings

Once again, we would like to remind the readers that the primary studies operationalized both L2 speech ratings and automated measures in a substantially different fashion. Thus, the intention of the methodological synthesis is to provide overall patterns. All in all, while human listeners’ agreement is generally strong (\(r = .7-.9\)), the relationship between human and automated scoring is comparable (\(r = .6-.8\)). Earlier studies demonstrated strong correlations between human and machine evaluations of L2 speech elicited from controlled speech tasks (e.g., word and sentence reading). More recent studies have explored the replicability of the findings, focusing on more spontaneous speech samples (e.g., monologues, interviews, picture descriptions). Thus far, scholars have convincingly shown that L2 speech proficiency scores were strongly associated with the results of automated analyses of temporal features in L2 speech (e.g., Ginther et al., 2010).

Motivation for the Current Study

Throughout the past 15 years, many scholars have extensively examined what contributes to naïve listeners’ perceptions of foreign-accented yet sufficiently comprehensible speech (Munro & Derwing, 1995). Such intuitive judgments of comprehensibility, intelligibility, and communicative competence (rather than nativelikeness) are believed to play a key role in communicative success in today’s globalized society in which most communication in English takes place between L2 speakers (Pennycook, 2017). From a theoretical standpoint, comprehensibility serves as a crucial index of adult L2 speech development as learners’ oral proficiency continues to become comprehensible rather than nativelike as a function of increased practice and immersion experience (Derwing & Munro, 2013; Saito, 2015a, 2015b).

\(^2\)Whereas scholars initially used HMM for more precise and reliable ASR, DMM has become increasingly dominant in more recent literature. For detailed accounts of HMM and DNN models, see Supporting Information A.
The quick, reliable, and automatic assessment of L2 comprehensibility has been strongly called for among practitioners (to help students achieve comprehensible L2 speech via feedback; Trofimovich et al., 2016) and researchers (to assess the different stages of adult L2 speech learning; Isaacs et al., 2017).

To advance the agendas of L2 comprehensibility and automatic assessment research, the current study serves as a first attempt to explore the extent to which machine-based assessments can simulate naïve listeners’ comprehensibility ratings of 100 L1 and L2 speakers’ semispontaneous speech (Study 1). Subsequently, we further delved into the validity, replicability, and generalizability of the regression model when it was applied to new speakers (Study 2); and tested under a more freely constructed interview task condition (Study 3).

Whereas some prior work has demonstrated the potential of automated L2 speech assessment (e.g., \( r = .6-.7 \) for automated vs. human assessments in Table 1), they have exclusively focused on controlled speech where transcripts are available. The current investigation looks at the potential of automated assessment in spontaneous speech. Although L2 speech research has shown that listeners attend to segmental, melodic, and temporal information while judging the quality of foreign-accented speech, most of the existing studies have adopted either phonological, melodic, or fluency measures. Based on the methodological synthesis presented in the preceding text (cf. Table 1), the current study adopted all the measures (phonological, melodic, and fluency) within the same model. As revied earlier, these measures included (a) speed and breakdown analyses for the fluency measures (e.g., Cucchiarini et al, 2000 for articulation rate and pause ratio); (b) variation analyses for the melodic measures (e.g., van Santen et al., 2009 for pitch and intensity variation); and (c) posterior-based analyses for the phonological measures (e.g., Shen et al., 2021 for posterior probabilities and gaps).

**Research Questions**

The following three research questions and predictions were formulated:

1. To what degree can automatic comprehensibility scoring simulate native listeners’ intuitive judgments of various levels of comprehensibility (Study 1)?
2. Can automated fluency scoring predict different levels of comprehensibility when applied to new datasets (Study 2)?
3. Can automated fluency scoring predict different levels of comprehensibility when applied to new task contexts (Study 3)?

**Predictions**

As shown in the previous studies (e.g., \( r = .7-.8 \); Saito, 2021 for a meta-analysis), about 50–60% of the variance in L2 comprehensibility can be explained by a range of manual phonological measures (linguistically trained coders’ analyses of segmental and melodic accuracy and temporal fluency). Thus, we expected to find relatively large correlation coefficients \( (r = .7-.8) \) between the automated phonological (maximum posterior probabilities, posterior gaps to natives), melodic (pitch and intensity variation), and fluency measures (articulation rate, pause ratio) and native listeners’ L2 comprehensibility judgments (R1). It was also predicted that such findings can be replicated under new speaker (R2) and task conditions (R3).
Study 1: Model Training Phase

Speakers

As a part of the investigators’ larger project, the team established a speech dataset of 1,000+ Japanese learners with a wide range of proficiency and experience levels both in Japan (i.e., beginner-to-intermediate learners with relatively limited opportunities to use L2 in English-as-a-Foreign-Language settings) and in Canada (i.e., intermediate-to-advanced learners who use their L2 on a daily basis in English-as-a-Second-Language settings). Their speech was elicited using a range of speaking tasks. The dataset was derived from a series of L2 speech projects that the team has been working on over the past 10 years; parts of the cross-sectional and longitudinal analyses have been reported elsewhere (Saito, 2015a, 2015b for experienced Japanese speakers; Saito & Hanzawa, 2016 for inexperienced Japanese speakers).

Given the length of time required to collect comprehensibility judgments (e.g., more than one hour including explanation, training, and rating for 50 30-sec samples), to avoid listener fatigue, only samples of a single speaking task (picture description) from 100 speakers were used. Four subgroups of speakers were carefully selected to represent various levels of comprehensibility. While all of them started learning L2 English in Japan from Grade 7 (13–14 years of age), they differed substantially in terms of the length and timing of their immersion experience.

In conjunction with both cross-sectional and longitudinal evidence of a significant relationship between increased Length of Residence (LOR) and enhanced comprehensibility (e.g., Derwing & Munro, 2013; Saito, 2015a), participants’ LOR was taken into consideration to create three subgroups (i.e., inexperienced, moderately experienced, and highly experienced Japanese speakers). Of course, LOR may not always reflect the amount of input that L2 learners have received. For example, the frequency of L2 use may widely vary among L2 speakers even if they stay in an L2-speaking country for the same period, with some choosing to use only L1 rather than L2 (for relevant discussion, see Flege & Bohn, 2021). To use LOR as an index of L2 experience for Moderately and Highly Experienced Japanese Speakers, efforts were made to recruit only those who reported English (rather than Japanese) as their main language of communication at work and/or home. As shown in our precursor projects (e.g., Saito, 2015a), LOR served as a significant predictor of L2 comprehensibility. In essence, the current dataset comprised Japanese speakers with substantially different levels of comprehensibility.

- **Inexperienced Japanese Speakers of English** ($n = 10$): This group represents low-level L2 comprehensibility. All the participants were first-year university students in Tokyo ($M$ age = 20.4 years; $Range$ = 18–21 years). While all of them were registered for three hours of English classes a week at the time of the project, they reported little conversational use of the language outside the classroom. None of them had any immersion experience abroad.

- **Moderately Experienced Japanese Speakers of English** ($n = 40$): This group represents mid-level L2 comprehensibility. The participants were residents in Canada ($M$ age = 34.7 years; $Range$ = 22–48 years). Their lengths of immersion varied widely ($M$ = 1.4 years; $Range$ = 0.1 to 5 years). They were considered late bilinguals as they had moved to major cities in Canada (Vancouver, Montreal, and Calgary) after puberty ($M$ age = 28.3 years; $Range$ = 19–40 years).

- **Highly Experienced Japanese Speakers of English** ($n = 40$): This group represents high-level L2 comprehensibility. All of them were late bilinguals with their ages of arrival being after puberty ($M$ age = 27.1 years; $Range$ = 21–36 years) and were
long-term residents with at least six years of immersion experience in Canada (M = 11.3 years; Range = 6–18 years). In addition, they reported extensive and regular use of L2 English as their primary language of communication in various settings. Following the standard in SLA research (see DeKeyser, 2013), they were considered highly experienced attainers who had reached the upper range of L2 speech proficiency.

- **L1 Baseline (n = 10):** This group serves as a L1 speaker baseline. A total of 10 L1 speakers of English were recruited in Vancouver, Canada (M-age = 27.5 years; Range = 18–37 years). They reported English as their L1 from birth onward with both of their parents being L1 English speakers.

**Speaking Task**

All the participants completed a timed picture description task. Building on a similar task procedure (Munro & Mann, 2005), participants described seven different pictures with five seconds of planning time for each photo. Three key words were provided per photo to help low proficiency speakers to produce sufficient lengths of spontaneous speech without too much dysfluency due to insufficient vocabulary knowledge. The first four pictures were used as practice for participants to get used to the task procedure (describing a photo in English with minimal planning), and the last three pictures (A, B, and C) were used for the final analyses.

The pictures depicted (a) a table left out in driveway (key words: rain, table, and driveway), (b) three men playing rock music with guitars (keywords: three guys, guitar, and rock music), and (c) a long road under a cloudy blue sky (keywords: blue sky, road, and cloud). These key words were selected to elicit segmental, melodic, and syllabic structures that are especially difficult for Japanese learners of English. For example, Japanese speakers are likely to neutralize the English /r/-/l/ contrast (“rain, rock, brew,” vs. “lane, lock, blue, cloud”) and substitute borrowed words (i.e., Katakana) by inserting epenthesis vowels between consecutive consonants (/dərəwə/ for “drive,” /θəri/ for “three,” /səkɑi/ for “sky”) and after word-final consonants (/teɪbɔl/ for “table,” /myʊzɪkə/ for “music”).

All the samples were recorded at individual meetings that took place at a community center, a university lab, and participants’ residences (prior to the COVID-19 pandemic) using digital Roland-05 audio recorders at 44.1 kHz sampling rate with 16-bit quantization.

**Comprehensibility Judgments**

**Listeners**

A total of 10 L1 listeners of General American English were recruited online (M-age = 24.8 years). They were all born in the United States and raised by monolingual parents. While all of them held a BA and/or MA degree, none of them majored in linguistics. Using a 6-point scale (1 = not at all, 6 = very much), they all reported that they were strongly familiar with foreign accented English speech (M = 5.8, Range = 4–6) but their familiarity with Japanese-accented English varied (M = 3.6, Range = 1–6). In line with Isaacs and Thomson’s (2020) categorization, these listeners can be considered naïve rather than expert. Following Munro and Derwing (1995), L2 comprehensibility was operationalized as naïve listeners’ intuitive judgments of spontaneous speech.
Procedure

Due to the pandemic, all of them completed the rating sessions individually with an investigator using a video-conferencing tool. All the rating sessions were conducted through the Gorilla platform for online research (Anwyl-Irvine et al., 2020). During the data collection, the investigator engaged with the participants to provide training, monitor their rating behaviors, respond to any enquiries, and help solve any technological problems. First, the listeners familiarized themselves with the listening materials (three picture descriptions by 100 L1 and L2 speakers) and the online rating procedure using Gorilla. Then, each listener received instruction on what characterized comprehensibility using the following definition:

- This term refers to how much effort it takes to understand what someone is saying. If you can understand (what the picture story is about) with ease, then a speaker is highly comprehensible. However, if you struggle and must listen very carefully, or in fact cannot understand what is being said at all, then a speaker has low comprehensibility.

All the samples were played in a randomized order. Upon hearing each sample once, they made an intuitive judgment using a 9-point scale (1 = difficult to understand, 9 = easy to understand). No repeat button was provided.

First, the listeners practiced the rating procedure using three samples (not included in the main dataset). By listening to the speech dataset that the research team established (beyond the current study), the author carefully identified the three practice samples that represented the three subgroups of Japanese speakers (inexperienced, moderately experienced, highly experienced). After the practice session was completed, the listeners proceeded to judging the comprehensibility of the 100 samples. Each session took around 2 hours with a 5-minute intermission a halfway through. In accordance with the recommended quality control measures for online L2 rating data collection (Nagle & Rehman, 2021), the platform was designed so that the listeners had to listen to the full-length of each sample (30 sec) before they rated the comprehensibility. The entire session was carefully monitored by the investigator using a video-conferencing tool.

Automated Fluency Measures

Following Tavakoli and Skehan’s (2005) model of fluency, and the automated fluency measures adopted in Cucchiarini et al. (2000), the two major temporal dimensions of L2 speech were measured: speed (speech rate), and breakdown (pause ratio). The repair aspect of fluency (the ratio of self-correction and repetition) was not taken into consideration as recent literature has suggested it is a trait of first language fluency (Duran-Karaoz & Tavakoli, 2020) or/and cognitive individual differences (Zuniga & Simard, 2019).

Using the “To TextGrid (silences)” function in Praat (Boersma & Weenink, 2019), pauses were detected as silences longer than 250 ms. The silence threshold (i.e., maximum intensity) was set to −20 dB as some speech samples included background noise. Next, the number of syllables was calculated based on de Jong and Wempe’s (2009) Praat script. Nuclei were detected when the following phonetic conditions were met: (a) peak intensity was 2 dB above the median intensity, and (b) it was preceded and followed by 4+ dB of dips in intensity. The Praat script did not
automatically identify and remove any filled pauses and repetitions (both of which were included in phonation time). All the temporal information was used to calculate the following fluency measures:

1. **Articulation rate:** This was calculated by dividing the number of syllables by the phonation time (sample duration minus all pauses).
2. **Pause ratio:** This was calculated by dividing the number of unfilled pauses by sample duration.

**Automated Phonological Measures**

To capture the multilayered nature of comprehensibility, it is important to include not only speech features related to the quantity of phonation (fluency) but also those related to the phonological quality of pronunciation (accuracy). As reviewed earlier, a range of previous studies have adopted the posteriorgram-based analyses (DNN-HMM; Chen et al., 2018; Zechner et al., 2009). As preparation for the current investigation, the preliminary project was conducted, wherein a range of analysis methods for posteriorgram-based data were proposed, piloted, and refined with a view of optimal L2 speech assessments (Shen et al., 2021). The two posteriorgram-based analyses were adopted to assess phonological quality: (a) maximum posterior probabilities and (b) posterior gaps to natives. They were found to greatly boost the predictive power of the automatic assessment of perceived fluency ($r = .8-.9$ for machine vs. humans).

First, DNN models were trained with the *Wall Street Journal* corpus, which featured 37,416 utterances spoken by 123 L1 speakers of General American English and corresponding scripts. Because the current dataset (100 picture descriptions) included noticeable noise, it was necessary to adjust the sound clarity of the corpus. To this end, four levels of babble noises and machine noises (computer-synthesized distortions) were added to the *Wall Street Journal* corpus (signal-to-noise ratio = 10, 30, 40, and 50 [dB]). As a result, noise-robust English DNN-HMM acoustic models were trained based on the *Wall Street Journal* corpus (Povey et al., 2011). Once the models were trained for ASR, any input sample can be converted to its posteriorgram.

The number of phonemes used in posteriorgrams is generally large ($n = 2,000–3,000$) (for further discussion on the concepts of posterior-based phonemes, see Supporting information B). As proposed in Kashiwagi et al. (2016), the Bhattacharyya distance between two states was calculated directly from their state posterior probabilities through Bayes’s theorem; the original dimension of the posteriorgrams ($n = 2,000$) was reduced to 50. From the posteriorgram of each utterance after pause removal, the two quality measures were calculated:

3. **Averaged Maximum Posterior Probabilities:** When L1 posteriorgrams were visualized with context-independent phonemes, a posterior vector at each time appeared to be a one-hot vector. This means that the phoneme intended had a probability close to 1.0 and the others had almost 0.0. Here, from a given posteriorgram, the maximum posterior probability was calculated at each time, and then averaged over time. For example, at any point in time, there are always probability scores for 50 phoneme states. Subsequently, the maximum posterior probability score was identified out of the 50 scores, and its corresponding phoneme was
assumed to indicate a speaker’s intended phoneme. The higher the average was, the more distinct the pronunciation of the utterance.

(4) *Averaged Posterior Gaps to Natives:* For each speaker, their averaged posterior vector was calculated. Because 10 native speakers were included in the main dataset, a gap from one learner to each native speaker was calculated to generate 10 gap scores in total. The distance was calculated via the Bhattacharyya metric. The averaged gap scores were thought to quantify the proximity of nativelikeness based on the distribution of perceived phonemes. Figure 1 visualizes the averaged posterior vectors and the posterior gaps. The former characterizes quality of pronunciation, represented as location in the feature space, and the latter characterizes distances to the 10 native speakers.

**Automated Melodic Measures**

As reviewed earlier, the previous literature has developed and adopted the automatic analyses of the varied use of pitch and amplitude in L2 speech (e.g., van Santen et al., 2009). To assess the melodic variation in participants’ speech production, the software openSMILE (“open-source Speech & Music Interpretation by Large-space Extraction”) was used (Eyben et al., 2010). Given that listeners rely on both pitch and intensity contour to perceive English melody (e.g., Lieberman, 1960), two directly relevant measures were adopted in the current study, that is, interquartile range values for F0 envelope and loudness:

(5) **Pitch variation:** After extracting fundamental frequencies (pitch frequencies) of given utterances and sorting the frequency values, the difference between 75th and 25th percentiles was calculated. The value indicated the magnitude of pitch variation or dynamics in the utterances. Monotonous speech (typical of beginner-to-intermediate L2 speakers) could be generally characterized as smaller pitch variations.

(6) **Intensity variation:** Similar to pitch variation, loudness was calculated as sequential data, which was defined as normalized intensity raised to a power of 0.3. After sorting the intensity values, the difference between 75th and 25th percentiles was calculated. The lack of intensity variation (and/or pitch variation) could help identify certain L2 English speakers who fail to distinguish between stressed and unstressed vowels.

![Figure 1. Conceptual summary of averaged posterior and posterior gap.](https://doi.org/10.1017/S0272263122000080) Published online by Cambridge University Press
Results

Listeners’ Judgments of Comprehensibility

The first objective of the analyses was to examine how 10 L1 listeners perceived the comprehensibility of 90 Japanese speakers (Inexperienced, Moderately Experienced, and Highly Experienced) and 10 L1 speakers. As stated earlier, these groups were designed to represent various levels of comprehensibility among inexperienced/experienced L2 speakers and L1 speakers. In terms of interrater reliability, the listeners demonstrated medium-to-strong agreement ($r = .760$). The strength of the correlations widely varied from $r = .569$ between Listeners 3 and 7 to $r = .832$ between Listeners 6 and 8 (see Table 2). As in many L2 comprehensibility studies (e.g., Munro & Derwing, 1995), their rating scores were averaged across to derive one comprehensibility score for each speaker as an index of the listeners’ overall ratings. According to the results of a Kolmogorov–Smirnov test, the averaged comprehensibility scores did not significantly differ from normal distribution ($D = .086$, $p = .096$). While all the participants in the L1 baseline group received 9 points, the L2 speakers’ comprehensibility scores demonstrated a great deal of individual variation ($M = 5.01$, $SD = 1.81$, $Range = 1.3–8.5$).

Automatic Assessments of Comprehensibility

The second objective of this study was to examine the extent to which L1 listeners’ comprehensibility judgments can be tied to a set of automated phonological, melodic, and fluency measures (for descriptive statistics, see Table 3). The results of the normality test (Kolmogorov–Smirnov) demonstrated that whereas most of their fluency, phonological, and melodic scores were comparable to normal distribution ($p > .05$), their posterior gap scores were significantly different from normal distribution ($D = .202$, $p < .001$). Because a severe positive skewness was observed, the posterior gap scores were submitted to inverse transformation, resulting in the scores becoming normally distributed ($D = .080$, $p = .515$). As such, the distance measure linearly represents the degree of proximity (i.e., nativelikeness). Given that both the dependent (comprehensibility ratings) and predictor variables (fluency, phonological, and melodic measures) followed normal distribution, only linear models were considered in the subsequent statistical analyses.

To examine how the automated assessments predicted different levels of comprehensibility (treated as ordinal values), a multiple regression model was constructed.

Table 2. Interclass correlations among 10 listeners’ comprehensibility judgments (Study 1)

<table>
<thead>
<tr>
<th>Listener 1</th>
<th>Listener 2</th>
<th>Listener 3</th>
<th>Listener 4</th>
<th>Listener 5</th>
<th>Listener 6</th>
<th>Listener 7</th>
<th>Listener 8</th>
<th>Listener 9</th>
<th>Listener 10</th>
</tr>
</thead>
<tbody>
<tr>
<td>Listener 1</td>
<td>.868</td>
<td>.694</td>
<td>.730</td>
<td>.712</td>
<td>.854</td>
<td>.767</td>
<td>.851</td>
<td>.755</td>
<td>.797</td>
</tr>
<tr>
<td>Listener 2</td>
<td>.696</td>
<td>.803</td>
<td>.758</td>
<td>.843</td>
<td>.778</td>
<td>.842</td>
<td>.795</td>
<td>.776</td>
<td></td>
</tr>
<tr>
<td>Listener 3</td>
<td>.614</td>
<td>.754</td>
<td>.716</td>
<td>.569</td>
<td>.658</td>
<td>.687</td>
<td>.764</td>
<td></td>
<td></td>
</tr>
<tr>
<td>Listener 4</td>
<td>.726</td>
<td>.804</td>
<td>.756</td>
<td>.802</td>
<td>.739</td>
<td>.752</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Listener 5</td>
<td>.799</td>
<td>.686</td>
<td>.741</td>
<td>.797</td>
<td>.803</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Listener 6</td>
<td>.784</td>
<td>.832</td>
<td>.748</td>
<td>.818</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Listener 7</td>
<td></td>
<td>.784</td>
<td>.744</td>
<td>.701</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Listener 8</td>
<td></td>
<td></td>
<td>.756</td>
<td>.807</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Listener 9</td>
<td></td>
<td></td>
<td></td>
<td>.773</td>
<td></td>
<td></td>
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</tr>
</tbody>
</table>
using the `lm` function in the R statistical environment (The R Foundation, n.d.). The model comprised listeners’ averaged comprehensibility scores as dependent variables relative to the six automated measures (articulation rate, pause ratio, maximum posterior probabilities, inversed posterior gaps, pitch, and intensity variability). As for model selection, “a full model” was chosen to maximize the predictive power of the model (with all six predictors entered in the equation). In the current analyses, we did not choose stepwise models (backward or forward selection based on the results of F tests) due to the following criticisms. First, the multiple comparisons conflated the occurrences of type 1 and 2 errors. Second, stepwise models are prone to overfitting the data and underestimating the degrees of freedom (for guidelines for the use of multiple regression analyses in applied linguistics research, see Larson-Hall, 2010).

As summarized in Table 4, the following model was tested: Listeners’ averaged comprehensibility scores = Intercepts + articulation rate + pause ratio + maximum posterior probabilities + inversed posterior gaps + pitch + intensity variability. The full model significantly explained 67.7% of the variance in the listeners’ comprehensibility judgments, $F(6, 93) = 32.507, p < .001$, without any clear evidence of multicollinearity problems (Variance Inflation Factor [VIF] = 1.016–1.456). In terms of the standardized $\beta$ values, perceived comprehensibility was mainly predicted by segmental quality ($\beta = .607$ for posterior gaps, $\beta = .267$ for max posterior probabilities), secondarily by temporal quality ($\beta = -.193$ for pause ratio, $\beta = .123$ for articulation rate), and finally by melodic quality ($\beta = .113$ for intensity variation, $\beta = -.093$ for pitch variation).

The predicted comprehensibility scores were calculated based on the regression model’s coefficients in Table 4. For each predictor variable, raw predictor values

<table>
<thead>
<tr>
<th>Measures</th>
<th>Interpretations</th>
<th>M</th>
<th>SD</th>
<th>95% CI</th>
</tr>
</thead>
<tbody>
<tr>
<td>A. Temporal quantity</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Articulation rate</td>
<td>syllables per second</td>
<td>3.618</td>
<td>0.510</td>
<td>3.279–3.5343</td>
</tr>
<tr>
<td>Pause ratio</td>
<td>%</td>
<td>0.406</td>
<td>0.107</td>
<td>0.385–0.428</td>
</tr>
<tr>
<td>B. Phonological quality</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Maximum posterior probabilities</td>
<td>Probabilities between 0 (silence) and 1 (only one particular phoneme identified)</td>
<td>0.805</td>
<td>0.030</td>
<td>0.799–0.811</td>
</tr>
<tr>
<td>Posterior gaps to natives</td>
<td>Distance to native speaker data between 0 (no distance) and 1 (heavily foreign accented)</td>
<td>0.066</td>
<td>0.023</td>
<td>0.061–0.070</td>
</tr>
<tr>
<td>C. Melodic quality</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Pitch variability</td>
<td>Degree of variation (25th vs. 75th percentile) in fundamental frequencies in Hz</td>
<td>69.218</td>
<td>35.550</td>
<td>62.164–76.271</td>
</tr>
<tr>
<td>Intensity variability</td>
<td>Degree of variation (25th vs. 75th percentile) in normalized intensity between 0 (flat amplitude) and 1 (varied amplitude)</td>
<td>0.716</td>
<td>0.217</td>
<td>0.673–0.759</td>
</tr>
</tbody>
</table>
(i.e., articulation rate, pause ratio, maximum posterior probabilities, inversed posterior gaps, pitch variability, and intensity variability) were multiplied by unstandardized $B$ (i.e., 0.524, –3.920, 19.427, 0.262, –0.006, and 1.137). Then, constant (i.e., –15.371) was added to total scores (i.e., the sum of all explanatory variables). Because one L1 speaking participant yielded 9.02, this was adjusted to the upper

*Table 4. Results of multiple regression analysis using automated measures as predictors of listeners’ comprehensibility scores*

<table>
<thead>
<tr>
<th>Predicted variables</th>
<th>Predictor variables</th>
<th>Unstandardized $B$</th>
<th>Standardized $\beta$</th>
<th>$F$</th>
<th>$p$</th>
<th>VIF</th>
</tr>
</thead>
<tbody>
<tr>
<td>Comprehensibility scores</td>
<td>Constant</td>
<td>–15.371</td>
<td>–4.096</td>
<td>&lt; .001</td>
<td></td>
<td></td>
</tr>
<tr>
<td></td>
<td>Articulation rate</td>
<td>0.524</td>
<td>.123</td>
<td>1.784</td>
<td>.078</td>
<td>1.359</td>
</tr>
<tr>
<td></td>
<td>Pause ratio</td>
<td>–3.920</td>
<td>–.193</td>
<td>–2.967</td>
<td>.004</td>
<td>1.213</td>
</tr>
<tr>
<td></td>
<td>Max posterior probabilities</td>
<td>19.427</td>
<td>.267</td>
<td>4.493</td>
<td>&lt; .001</td>
<td>1.016</td>
</tr>
<tr>
<td></td>
<td>Inversed posterior gaps</td>
<td>0.262</td>
<td>.607</td>
<td>8.535</td>
<td>&lt; .001</td>
<td>1.456</td>
</tr>
<tr>
<td></td>
<td>Pitch variation</td>
<td>–0.006</td>
<td>–.093</td>
<td>–1.324</td>
<td>.189</td>
<td>1.417</td>
</tr>
<tr>
<td></td>
<td>Intensity variation</td>
<td>1.137</td>
<td>.113</td>
<td>1.656</td>
<td>.101</td>
<td>1.345</td>
</tr>
</tbody>
</table>

*Figure 2. The relationship between human comprehensibility scores and predicted comprehensibility scores ($r = .823$).*
The Pearson correlation coefficients between predicted and human comprehensibility scores were relatively strong, $r = .823$, $p < .001$ (see Figure 2). The figure here was higher than the 10 listeners’ averaged interrater agreement ($r = .760$ among Listeners 1–10), and comparable to that of Listeners 6 and 8 who demonstrated the strongest agreement ($r = .832$ for Listeners 6 vs. 8). According to the results of an independent $t$-test, the predicted and human comprehensibility scores did not significantly differ, $t = .002$, $p = .998$, $d = 0.01$.  

**Study 2: Model Validation Phase**

Overall, Study 1 demonstrated (a) that the machine scoring can successfully simulate human listeners’ comprehensibility judgments ($r = .823$ for machine vs. humans; $r = .760$ for humans vs. humans); and (b) that such automated comprehensibility assessments can be primarily determined by phonological accuracy ($\beta = .607$ for inversed posterior gaps; $\beta = .267$ for max posterior probabilities) and secondarily by fluency features ($\beta = -.193$ for pause frequency; $\beta = .123$ for articulation rate). The results here are in line with the existing literature (the relation weights of phonological accuracy over fluency in L2 comprehensibility; see Isaacs & Trofimovich, 2012). Interestingly, none of the melodic measures (pitch and intensity variation) turned out to be significant predictors. This could be due to several reasons. First, the timed picture description task used in the current study may not have elicited sufficiently long sentences, limiting speakers from demonstrating varied and accurate use of melodic cues. Secondly, the relationship between pitch height measures (including pitch range) and L2 comprehensibility may be minimal ($\beta = -.10$ in Kang et al., 2010). Thirdly, listeners orchestrate a range of different melodic cues (beyond pitch and intensity) to perceive, identify, and encode the targetlikeness of lexical and sentence stress in English (Lieberman, 1960). Study 2 was designed to test the extent to which the automated assessments can predict different levels of L2 comprehensibility when applying the regression formula in Table 4 to a new L2 speech dataset ($N = 45$ timed picture descriptions).

**Method**

**Speakers**

A total of 40 Japanese learners of English were recruited from Japan and Canada (17 males, 23 females). Thirty of them were university-level students in Japan ($M_{age} = 19.8$ years; $Range = 18–26$ years). The results of their general English proficiency test scores (Test of English for International Communication) suggested that their L2 oral proficiency varied widely ($M = 681.5$ out of 900; $Range = 300–980$), covering basic to proficient users. The other 10 participants were recruited in Vancouver and Ontario ($M_{age} = 41.4$ years; $Range = 35–47$ years). All of them were late L2 speakers ($M_{age of arrival} = 27.3$ years; $Range = 21–35$ years) who used English regularly and had a great deal of immersion experience in Canada ($M_{length of immersion} = 13.1$ years; $Range = 6–27$ years).

---

3There was one predicted score beyond 9 (9.02). This is because the predictive power of the model ($R^2 = .677$) was quite strong but not perfect. In addition, the model assumed a linear relationship between machine and human ratings without any breakpoints (i.e., linear rather than piecewise regression). Taking a close look at the participant in concern, he was a native speaker of English whose speech was characterized as substantially fast speech rate and low pause ratio.
Range = 7–24 years). All of them could be considered highly functional and regular L2 users as they reported English rather than Japanese as their primary language of communication in home and/or work settings. Finally, five L1 speakers of Canadian English were included. The biographical profiles of the participants in Study 2 can be considered comparable to those in Study 1. They were believed to represent different levels of comprehensibility.

**Speaking Task**

The participants’ spontaneous speech in English was elicited using the same task used in Study 1 (timed picture description). The first 10 seconds of three picture descriptions were stored in a single WAV file per participant and used for the comprehensibility judgments.

**Comprehensibility Judgments**

A total of five L1 English speakers were recruited in London, England ($M_{\text{age}} = 20.6$ years). Using the same rating procedure in Study 1, they participated in individual rating sessions with a researcher using a video-conferencing tool. First, they received a brief summary of project, and explanations of the nature of the dataset, the definition of comprehensibility, and the rating procedure. Next, to familiarize themselves with the procedure, they rated the same practice samples used in Study 1, explained their decisions, and received feedback from the researcher. Finally, they proceeded to make comprehensibility judgments of the main dataset (40 minutes without any intermission).

**Automated Speech Analyses**

The speech data was submitted to the same automated fluency (articulation rate, pause frequency), phonological (maximum posterior probabilities, posterior gaps to natives), and melodic analyses (pitch and intensity variability) as Study 1.

**Results**

Similar to Study 1, the averaged interrater agreement among the five listeners was relatively high, $r = .789$ ranging from $r = .718$ (Listeners 1 vs. 2) to .871 (Listeners 2 vs. 3; summarized in Table 6). Their comprehensibility scores were averaged per speaker to represent the human listeners’ overall comprehensibility judgments ($M = 3.67; SD = 2.26; \text{Range} = 1–9$). According to the Kolmogorov–Smirnov test, the resulting scores did not significantly differ from normal distribution, $D = .160, p = .179$. To check the

| Table 6. Interclass correlations among five listeners’ comprehensibility judgments (Study 2) |
|---------------------------------|----------------|----------------|----------------|----------------|
| Listener 2 | Listener 3 | Listener 4 | Listener 5 |
| Listener 1 | .718 | .761 | .769 | .759 |
| Listener 2 | .850 | .845 | .830 |
| Listener 3 | .871 | .745 |
| Listener 4 | .743 |

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degree to which the comprehensibility scores in Study 2 (n = 45) differed from those in Study 1 (n = 100), a nonparametric test (Mann–Whitney) was conducted. The results showed that the participants in Study 2 (M = 3.67) received lower comprehensibility scores than those in Study 1 did (M = 5.40), Z = −4.429, p < .001. Although efforts were made, the two datasets were not comparable. It is noteworthy, however, that the listeners were different between Studies 1 and 2 (n = 10 American listeners vs. 5 British listeners). Thus, we consider this to be a tentative pattern.

Using the regression formula in Table 5, the subsequent analyses set out to examine the extent to which the six automated scores (summarized in Supporting Information C) can predict human comprehensibility scores. To this end, the raw maximum posterior gap scores were inversed. All the temporal, phonological, and melodic scores followed normal distribution (D = .082–.143, p = .281–.896). The predicted comprehensibility scores varied between 1.82 and 7.17 (M = 4.13, SD = 1.23).

As visually summarized in Figure 3, the correlation between human and predicted comprehensibility scores was relatively strong, r = .827 (p < .001) and can be compared to the averaged correlation coefficients among the five listeners (r = .789). An independent t-test did not detect a significant difference between the human listeners’ comprehensibility scores and listeners’ averaged comprehensibility scores, t = −1.186, p = .239, d = 0.252. To sum, the results indicated (a) that the automated measures successfully simulated the human listeners’ comprehensibility scores and (b) that the difference between the predicted and human listener scores was minimal.

**Study 3: Generalization Study**

Study 2 provided some empirical support for the predictive power of the regression model for human listeners’ comprehensibility scores. Notably, the findings have thus
far been based on a relatively structured, semispontaneous speech task (timed picture description) wherein the linguistic content of sample is likely to be highly predictable (e.g., all participants used the same word prompts to describe the same pictures). In Study 3, we aimed to examine the extent to which the regression model can be applied when speech is elicited using a more extemporaneous, free-constructed task.

**Speakers and Listeners**

The same speakers and listeners from Study 2 (40 Japanese learners of English in Japan and Canada; 5 L1 speakers of English; 5 L1 speaking listeners) participated in Study 3.

**Speaking Task**

The participants engaged in an oral interview task which was created by the research team based on the procedure in the IELTS interview task (Crowther et al., 2015). Given that the task format gave participants a certain level of freedom over the content and organization of their speech (rather than describing pictures), participants were encouraged to produce extemporaneous speech with their primary focus on conceptualization (what to say) rather than accuracy (how to say it). In accordance with Skehan’s (1998) task taxonomy, the oral interview task can be considered less structured, and more informal and personal. In such tasks, speakers likely use more contextually rich, varied, and idiosyncratic language. This sharply contrasts with the timed picture description task, wherein participants are likely focused on accurately and fluently using language to describe the provided content in the most effective and efficient manner.

First, the participants were given a familiar and personal topic to discuss (i.e., What was the hardest and toughest change in your life?). To help speakers produce long and meaningful speech, a set of possible discussion points (e.g., Why was it so challenging?) were also provided on a topic card. After the participants spent one minute on planning, they spoke for approximately 2 minutes. Finally, the researcher asked one or two follow-up questions in response to the content of their speech (e.g., What did you learn from the experience?). For the task prompts, see Supporting Information D. The first 30 seconds of each sample were carefully cut and saved in a single WAV file.

**Comprehensibility Judgments**

After the five listeners completed the comprehensibility judgments of the 45 picture description samples in Study 2, they assessed the comprehensibility of 45 interview samples on a 9-point scale using the same procedure as in Studies 1 and 2.

**Automated Speech Analyses**

The speech data was analyzed via automated fluency, phonological, and melodic measures in Studies 1 and 2.

**Results**

Similar to Study 1 ($r = .760$) and Study 2 ($r = .789$), the averaged interrater agreement among the five listeners in Study 3 was relatively high, $r = .756$. As summarized in
The strength of the agreement between certain individual listeners varied from “mid” ($r = .655$ for Listeners 4 vs. 5) to “strong” ($r = .869$ for Listeners 2 vs. 3). To obtain the listeners’ consensus, their comprehensibility scores were averaged per speaker ($M = 4.47$, $SD = 2.05$, $Range = 1–9$). The results of the normality test (Kolmogorov–Smirnov) found the resulting comprehensibility scores to be indistinguishable from normal distribution, $D = .103$, $p = .685$. To calculate the predicted comprehensibility scores, the regression formula in Table 5 was used. Following Studies 1 and 2, the raw maximum posterior gap scores were inversely transformed. The normality test (Kolmogorov–Smirnov) demonstrated that all the automated scores (articulation rate, pause frequency, maximum posterior probabilities and gaps, pitch, and intensity variability) followed normal distribution ($D = .052–.117$, $p = .527–.992$). The predicted comprehensibility scores varied between 2.12 and 8.85 ($M = 4.33$, $SD = 1.73$).

As visually displayed in Figure 4, the predicted comprehensibility scores were strongly associated with the human listeners’ comprehensibility scores, $r = .809$ ($p < .001$). The correlation coefficients here can be considered similar to those among the five listeners ($r = .756$). According to the results of an independent $t$-test, the difference between human and predicted comprehensibility scores did not reach statistical significance, $t = −0.360$, $p = .719$, $d = 0.07$. In a nutshell, the findings here indicated that the automated measures can greatly simulate human judgments of different levels of comprehensibility not only when speech is elicited using a relatively structured task.

### Table 7. Interclass correlations among five listeners’ comprehensibility judgments (Study 2)

<table>
<thead>
<tr>
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<th>Listener 2</th>
<th>Listener 3</th>
<th>Listener 4</th>
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<tbody>
<tr>
<td>Listener 1</td>
<td>.719</td>
<td>.707</td>
<td>.695</td>
<td>.726</td>
</tr>
<tr>
<td>Listener 2</td>
<td>.869</td>
<td>.752</td>
<td>.741</td>
<td>.858</td>
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<tr>
<td>Listener 3</td>
<td></td>
<td>.844</td>
<td>.655</td>
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<td>Listener 4</td>
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**Figure 4.** Relationship between human comprehensibility scores and predicted comprehensibility scores ($r = .809$).
(picture description) but also when the regression model is applied to free, extempo-
aneous speech (oral interview).

**Discussion**

Based on a speech corpus of 90 Japanese speakers of English in Japan and North America, and 10 L1 speakers of English, Study 1 showed the extent to which L1 listeners’ comprehensibility ratings can be simulated by a set of automated fluency (articulation rate, pause ratio), phonological (posterior probabilities and gaps), and melodic (pitch and intensity variation) measures. The composite model’s predicted comprehensibility scores were significantly predictive of human assessments of comprehensibility. The strength of the correlation coefficients ($r = .823$ for machine vs. humans) can be considered comparable to the intrarater agreement among 10 naive listeners ($r = .760$ for humans vs. humans). In Studies 2 and 3, we further confirmed the validity and generalizability of the model as it strongly predicted L2 comprehensibility among a different group of Japanese speakers’ L2 English speech that was elicited not only from the same picture description task ($r = .827$ for machine vs. humans) but also through a more freely constructed oral interview task ($r = .809$ for machine vs. humans).

First and foremost, the findings supported the emerging evidence for the possibility, reliability, and validity of machine assessments of spontaneous speech samples (e.g., $r = .799$ in Fu et al., 2020; $r = .77$ in Chen et al., 2018). Furthermore, our study indicated that such automatic scoring can be used to simulate naïve listeners’ intuitive judgments of L2 speech comprehensibility, an anchor of communicative success among English speakers in global contexts. We would like to stress the importance of using the recent models of fluency-based measures and DNN-based instruments complementarily. Because the fluency, phonological, and melodic measures tap into different dimensions of L2 speech without notable multicollinearity (VIF < 1.5), they make separate contributions for the model to capture the multifaceted nature of L2 comprehensibility.

As shown in the previous studies (e.g., Isaacs & Trofimovich, 2012), it is important to note that listeners’ comprehensibility scores were uniquely related to various areas of language with relatively different weights (the order of importance: phonetic accuracy > fluency > melodic variation). This in turn indicates that comprehensibility is a complex construct as listeners attend to all the available linguistic information in foreign accented speech to decipher as much meaning as possible. This supports the importance of including conceptually and methodologically diverse measures (phonological, fluency, melody). It is possible that the relatively strong predictive power of the current findings ($r = .823$) could be ascribed to the fact that we adopted an integrative approach (including fluency, phonological, and melodic measures) rather than single models typically found in previous studies (e.g., Fu et al., 2020 for phonological measures; Ginther et al., 2010 for fluency measures).

Given that the current study took a first step toward examining the potential of automated assessments of L2 comprehensibility using fluency, phonological, and melodic measures in a complementary fashion, a set of promising directions need to be addressed with a view toward future extension studies. the findings of the current study (and those of the existing research in Table 1) exclusively focused on the correlation analyses with an assumption that the relationship between human and machine scoring could be linear. Future studies should further examine the nature of the relationship (e.g., linear vs. quadratic).
Secondly, although the current study exclusively focused on Japanese speakers of English, it would be intriguing to replicate the findings in the context of different L1–L2 pairings. For example, Crowther et al. (2015) showed that L1 listeners were found to attend to different linguistic dimensions while assessing the comprehensibility of L2 English speech among L1 Chinese speakers (segmentals), L1 Hindi speakers (intonation), and L1 Farsi speakers (segmentals, word stress, fluency). Saito and Akiyama (2017) demonstrated that L1 listeners likely prioritized melodic rather than segmental information during the comprehensibility judgments of L2 Japanese speech. Given that the linguistic weights of perceived comprehensibility vary across different L1 and L2 contexts, further research is necessary to see the extent to which the combination of the automated fluency, phonological, and melodic measures featured in the current study (articulation rate, pause ratio, posterior probabilities/gaps, pitch/intensity variation) can be applied to other large-scale datasets.

Third, although the current study included temporal, phonological, and melodic measures within the same model, there is a growing amount of research showing that naïve listeners take into account not only pronunciation but also lexicogrammatical information while making L2 comprehensibility judgments (Isaacs & Trofimovich, 2012). In fact, previous work has demonstrated that listeners primarily rely on phonological and temporal information (40–50% of the variances) but secondarily on lexicogrammar (10–20% of the variances; Crowther et al., 2015). There is a growing amount of evidence for the relative importance of certain vocabulary features such as lexical appropriateness rather than richness (Isaacs & Trofimovich, 2012) and collocational association rather than frequency (Saito, 2020). One promising direction is to add automated word recognition measures to further increase the predictive power of the model that we have developed for the current study (see Kyle & Crossley, 2015 for their discussion on the automated assessments of L2 spoken vocabulary).

Fourth, although efforts were made to include a range of fluency, melodic, and phonological measures in the model, there are a range of other key automated fluency, melodic and phonological measures that future studies should highlight to further improve the predictive power of the automated L2 speech assessment. For example, Fu et al. (2020) conducted posteriorgram-based analyses of participants’ L1 and L2 speech data. Unlike the current study (using only participants’ L2 speech data), future work should compare the distance between the posteriorgram distribution of participants’ L1 and L2 speech data. On a related note, the results of the multiple regression analyses in the current study found neither of the melodic variation measures (i.e., pitch and intensity variation) to make a significant contribution to the predicted compressibility scores. Given that there is ample research evidence showing that melodic accuracy is strongly associated with L2 comprehensibility (e.g., Isaacs & Trofimovich, 2012), future studies should include both melodic variation and accuracy measures. As we acknowledged in the “Literature Review” section, however, it is important to remember (a) that the existing literature (including the current investigations) has exclusively focused on the variation of melody and (b) that few studies have probed the automatic analyses of the accurate use of melody especially in spontaneous speech (where transcripts and model speech are unavailable). Thus, we may need to wait for more future research to develop, validate, and refine the automated analyses of how L2 speakers use various types of melodic cues while marking word and sentence stress in a contextually appropriate manner.

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Fifth, we acknowledge that the findings in the current investigation were based on a series of cross-sectional analyses. As Knoch and Chapelle (2018) argued, assessment validation entails much more than correlational examinations. The question of whether machine assessments of spontaneous speech will be suitable for making fine-grained decisions may still be at hand. One promising direction for future replication studies concerns a longitudinal investigation of the development of comprehensibility among L2 speakers over time. There is evidence showing that L2 speech learning takes place on a continuum of comprehensibility in relation to increased immersion experience (e.g., Derwing & Munro, 2013). Thus, researchers can track how L2 leaners’ speech behaviors change for a certain period (1–5 years of immersion), how listeners can perceive such changes from the point of comprehensibility, and how automated measures can replicate the developmental trajectories of comprehensibility among the same individuals.

Sixth, a reviewer pointed out the significance of the predicted scores calculated by the automated measures. As seen in Figure 3, for example, although native English speakers received the highest comprehensibility scores (i.e., 9), the predicted comprehensibility scores varied to a great extent (as low as 5). Here, we would like to remind the readers that the correlations between the human and automated assessments of L2 comprehensibility were considered relatively strong but not perfect ($r = .823, .827, \text{and } .809$ in Studies 1, 2, and 3). In addition, it is important to remember that the interrater correlations among the human raters were considered medium-to-strong ($r = .7 – .8$). This in turn suggests that it is difficult to obtain an absolute L2 comprehensibility rating for a particular L2 speech sample because it triggers different types of reactions (thereby different L2 comprehensibility scores) not only between machine and human raters but also within human raters.

Finally, many EFL students suffer from the lack of speech training opportunities, wherein they can receive individualized feedback on their speaking skills especially regarding comprehensibility (rather than nativelikeness). The current studies shed light on the pedagogical potential of automated assessment of the ecologically valid metric of comprehensibility. Such automated comprehensibility judgments can be used to guide individual students to develop their L2 comprehensibility in accordance with their own goals relative to their current proficiency levels. This will in turn allow learners to engage in more tailored, optimal, and autonomous L2 speech learning in the long run. For example, the descriptive statistics showed that Japanese speakers’ comprehensibility levels ranged from 1.3 to 8.5 (relative to native speakers who scored 9 out of 9). Using the values as a rough benchmark for different levels of comprehensibility, inexperienced Japanese speakers can be encouraged to aim at moderately comprehensible speech which corresponds to moderate L2 English proficiency (5–6 out of 9). Similarly, Japanese participants whose speech is already somewhat intelligible/comprehensible can aim to achieve more advanced, adequately comprehensible speech, which corresponds to highly advanced L2 English oral proficiency (7–8 out of 9). Importantly, students should be informed (a) that many L2 learners should aim for mid-to-high L2 comprehensibility to become functional, competent L2 users (6–8 out of 9) and (b) that comprehensibility and accentedness are separate constructs, and even highly accented speech (e.g., 9 out of 9) can be adequately comprehensible. By adopting both quantitative and qualitative analyses, future studies are strongly recommended to further pursue how the use of automated assessment can enrich L2 speech training and then facilitate the different levels of L2 comprehensibility.
Conclusion

Given that eliciting listeners’ L2 speech assessment is a time-consuming task, some previous research has begun to examine the automated assessment of linguistically trained raters’ general speaking proficiency judgments. However, the analyses of these studies have been restricted to controlled speech samples and/or a limited set of speech measures (phonological fluency or accuracy). Using a speech data of L1 and L2 speakers of English, the current study took a first step toward combining three different automated analyses (fluency, phonological, and melodic) to simulate how naïve listeners intuitively perceive the comprehensibility of spontaneous speech samples across different task conditions. Findings showed that the composite model can provide predicted comprehensibility scores ($r = .809–.827$ for machine vs. humans) that are comparable to what different naïve listeners likely agree on ($r = .756–.789$ for humans vs. humans). The current study supports the use of up-to-date automated assessment of L2 oral proficiency on the continuum of comprehensibility which many researchers have suggested as an ecologically valid goal for adult L2 speakers and students (comprehensible rather than nativelike).

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Data Availability Statement The experiment in this article earned an Open Materials badge for transparent practices. The materials are available at https://osf.io/cm7bf/.

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