Optimising Turn-Taking Strategies With Reinforcement Learning

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Abstract

In this paper, reinforcement learning (RL) is used to learn an efficient turn-taking management model in a simulated slot-filling task with the objective of minimising the dialogue duration and maximising the completion task ratio. Turn-taking decisions are handled in a separate new module, the Scheduler. Unlike most dialogue systems, a dialogue turn is split into micro-turns and the Scheduler makes a decision for each one of them. A Fitted Value Iteration algorithm, Fitted-Q, with a linear state representation is used for learning the state to action policy. Comparison between a non-incremental and an incremental handcrafted strategies, taken as baselines, and an incremental RL-based strategy, shows the latter to be significantly more efficient, especially in noisy environments.

1 Introduction

Most dialogue systems use a simple turn-taking model: the user speaks and when she finishes her utterance, the system detects a long enough silence and speaks afterwards. Quite often the latter cannot be interrupted neither. On the contrary, incremental dialogue systems are able to understand the user’s utterance on the fly thus enabling a richer set of turn-taking behaviours. They can interrupt the user and quickly report a problem. They can be interrupted as well. In this paper, we explore the extent to which such capacity can improve the overall dialogue efficiency. Reinforcement learning (Sutton and Barto, 1998) is used to find optimal strategies.

Human beings use a rich set of incremental behaviours which help them recover from errors efficiently. As soon as a conversation participant detects a problem, she is able to interrupt the speaker so that he can correct his utterance or repeat a part of it for example. In this work, we implement in an expert handcrafted way 3 turn-taking phenomena amongst those classified in the taxonomy proposed in (Khouzaimi et al., 2015a). The resulting strategy is shown to achieve better performance than a non-incremental handcrafted strategy. Then, it is compared to an automatically learned incremental strategy and the latter is shown to achieve even better results.

Machine learning algorithms often need important sets of data in order to converge. In the field of dialogue systems, gathering data is expensive and as a consequence, researchers use simulated users for learning (Eckert et al., 1997; Chandramohan et al., 2011; Pietquin and Hastie, 2013). To run the experiments in this work, a simulated user interacts with a service that manages a personal agenda (Khouzaimi et al., 2015a).

In our work, the turn-taking task is separated from the common dialogue management one and it is handled by a separated module called the Scheduler (Khouzaimi et al., 2014). A considerable asset of this architecture is that it can just be added to the agenda service in order to make it incremental. Two versions of this module have been developed: the first one embeds the handcrafted strategy and the second one uses reinforcement learning to optimise turn-taking decisions with respect to objective criteria. Our goal is to improve the dialogue efficiency, therefore, as evaluation criteria and in order to design a reward function, dialogue duration and task completion are used. Fitted-Q (a Fitted Value Iteration algorithm) was used and we show that the optimal policy is quickly learned and that it outperforms both the non-incremental and the handcrafted strategies. These three strategies are then compared under different noise conditions and the automatically learned strategy is proven to be the most robust to high levels of noise.

Section 2 presents some related work and Sec-
tion 3 describes the simulated environment used for the experiments. Then Section 4 describes the handcrafted turn-taking model as well as the RL one. Section 5 presents the experimentation and the results and finally, Section 6 gives some concluding remarks.

2 Related work

The idea of interrupting the user in order to improve the dialogue efficiency in terms of dialogue duration and task completion is tackled in (Ghigi et al., 2014). A corpus study shows that the users’ utterances often go off-domain or contain the same piece of information several times. By detecting this kind of sentences and interrupting the user to report the problem promptly, the dialogue is more efficient and users tend to conform to the words and expressions that are known by the system. Only handcrafted settings are explored.

An approach based on Hierarchical Reinforcement Learning is presented in (Dethlefs et al., 2012; Hastie et al., 2013). An efficiency reward is used to optimise the Information Presentation strategy (common dialogue management task) whereas another reward based on Information Density is used for the barge-in and backchannel tasks. In our work, the efficiency reward is directly applied to turn-taking management.

A research branch in incremental dialogue focuses on the following principle laid in (Sacks et al., 1974): Participants in a conversation attempt to minimize gaps and overlaps. (Jonsdottir et al., 2008) uses a reinforcement learning approach based on this principle in order to achieve smooth turn-taking (only prosodic features are considered) whereas (Raux and Eskenazi, 2008; Raux and Eskenazi, 2012) proposes a classification method where the costs for silences and overlaps are handcrafted. Like the majority of contributions in the field of incremental dialogue, the main focus here is smooth turn-taking rather than improving the general dialogue efficiency.

In order to mimic human turn-taking capabilities, in (Kim et al., 2014) Inverse Reinforcement Learning has been applied to a system that can perform three turn taking actions: *speak, silent* and *overlap*. The main focus here is also end of utterance detection and smooth turn-taking.

In (DeVault et al., 2011), the ability of incremental dialogue systems to guess the remaining part of a user’s utterance before its end is explored. (Lu et al., 2011) applies reinforcement learning to explore the tradeoff between the risk of error relative to a barge-in due to an early guess and the lack of reactivity in the case of late system responses.

Finally, reinforcement learning is also applied in (Selfridge and Heeman, 2010) in the case of mixed initiative dialogue systems. However the paper does not tackle the problem of barge-in management but initial turn-taking (who takes the floor first): the dialogue participant that has the most important thing to say to make progress in the dialogue takes the floor first.

3 Simulated environment

To learn the turn-taking policy, a simulated environment has been developed. Figure 1 gives an overview of its architecture. The six modules on
the left constitute the user simulator: the Intention Manager, the Natural Language Generator (NLG), the Verbosity Manager, the ASR Output Simulator, the Patience Manager and the Natural Language Understanding module (NLU). The ASR Output Simulator communicates the N-Best corresponding to the current partial hypothesis to the Scheduler whose responses are conveyed to the NLU module.

3.1 Service task
The service used in our experiments is a personal agenda manager. The user can add events to the agenda, modify their attributes or delete them. To complicate a bit the task and justify the need for interactions a constraint has been introduced: all events must have separate time slots. If the user tries to overload a busy time slot, a warning is generated and the user is required to modify her request.

The simulated dialogue scenarios are defined by two event lists. The first one (InitList) is the list of events that already exist in the agenda before the dialogue and the second one (ToAddList) is the list of events, with priorities and alternative times, to add during the dialogue. The simulated user tries to make the maximum number of events with the highest priority values fit into the agenda. For example, if InitList contains the event 

```
{title: house cleaning, date: January 6th, slot: from 18 to 20, priority: 3, alternative 1: January 7th, from 18 to 20, alternative 2: January 9th, from 10 to 12}
```

and ToAddList contains the event 

```
{title: birthday party, date: January 6th, slot: from 18 to 23, priority: 2}
```

then the user simulator will first try to schedule his birthday party on January 6th from 18 to 19 because this slot is already booked for the house cleaning event. Therefore, the user simulator will reschedule the latter to January 7th from 18 to 20. If the house cleaning event was given with no alternative, it will be deleted as it has a lower priority compared to the birthday party.

3.2 Service implementation
The service is designed to process slot-filling requests in order to manage the user’s personal agenda. The required slots are the action type, the event description, its date and its time slot. A keyword-based NLU module is configured to parse natural language requests.

A mixed initiative strategy is used for the interaction. First, the user takes the lead and utters a request in natural language. The system picks as many information slots as it can and if it still needs more information, it asks the user to complete her request slot by slot. In the following, a dialogue task refers to the portion of dialogue corresponding to a particular add, modify or remove action. In the best case, it consists of two dialogue turns only (request and confirmation) but if there is important noise, it might take several turns to complete (given that the user is patient enough not to hang up before the end of the dialogue task).

The following dialogue example is composed of two dialogue tasks. In the first one, the system manages to catch all the information needed to answer whereas in the second one, it missed the time slot and has to ask for it again. Moreover, the first dialogue task led to a conflict declaration whereas the second one could be accomplished:

- **System:** Hi! Welcome to your personal agenda service. How may I help you?
- **User:** Can you please add the event football game on November 9th from 18 to 20 if possible?
- **System:** Sorry. The time slot from 18 to 19 on November 9th is already booked for the event guitar course.
- **User:** Ok well, then move the event guitar course to November 10th from 18 to 19.
- **System:** Please specify a time slot.
- **User:** From 18 to 19.
- **System:** Great. So you want to move the event guitar course to November 10th from 18 to 19. Is that right?
- **User:** Yes.
- **System:** Good. Do you want to add another modification to your agenda?
- **User:** No.
- **System:** Bye.
3.3 ASR N-Best generation

Given ToAddList and the system responses, the Intent Manager is able to compute the user’s immediate goal. Afterwards, the NLG transforms it into a sentence like add the event birthday party on January 6th from 18 to 23. Moreover, if the simulated user takes the lead and starts asking for a specific slot, it is also able to give the information directly.

In (Ghigi et al., 2014), a corpus study shows that the user is likely to use off-domain expression, to add unnecessary information and to repeat the same piece of information several times in the same sentence (especially after a misunderstanding). To simulate this phenomenon, the user’s request in natural language is given to a Verbosity Manager module that adds prefixes like I would like to and suffixes like if possible, that repeats the same information after a misunderstanding (with a given probability, e.g. 0.3) and that replaces the request with an off-domain sentence (with a given probability, e.g. 0.1).

To our knowledge, apart from the simulator described in (Selfridge et al., 2012), existent user simulators are turn-based and therefore, only the user intent is communicated at each turn (in a concept format) so there is no need to take care of the utterance formulations. This is not the case when incremental dialogue and turn-taking are the object of interest. In this case, the user’s sentence is processed chunk by chunk. The update step is called a micro-turn and in this paper, the unit chosen is the word. Suppose that the current user utterance contains N words \(w_1, w_2, ..., w_N\) then at micro-turn \(t\), the Verbosity Manager sends \(w_t\) to the ASR Output simulator. The latter stores an N-Best list from the previous micro-turn \(\{(s_1^{(t-1)}, hyp_1^{(t-1)}), \ldots, (s_N^{(t-1)}, hyp_N^{(t-1)})\}\) that is updated according to \(w_t\) and WER \(hyp_i\) is the \(i^{th}\) hypothesis in the N-Best and \(s_i\) is the corresponding confidence score. \(w_t\) can be replaced by a new word from a dictionary, deleted or a new word can be added to simulate the ASR noise, before it is added to the N-Best list.

The confidence score associated with the new word is computed as follows: if the word has not been modified, \(X\) is sampled from a Gaussian with mean 1 and variance 1 otherwise the mean is -1. We then compute the \(\text{sigmoid}(X) = (1 + \exp(-X))^{-1}\) as a word score (Figure 2 represents these two symmetric distributions). This is an adaptation of the simple ASR model introduced in (Pietquin and Beaufort, 2005). The score of the current partial utterance is the product of its words.

Another important aspect of incremental ASR is instability. A new ASR input does not necessarily translate into adding elements on top of the current output as it can change an important part if not the totality of it. For instance, in (Schlangen and Skantze, 2011), when the user says forty, it is first understood as four then forty. This is due to the fact that, given the language model, the new received chunk of information is more likely to complete a hypothesis that has a lower score in the N-Best list than the best hypothesis. In this work, as no language model is used, we use the NLU knowledge instead. If a new input leads to a new NLU key concept, then its score is boosted like in the following

\[ s_i \leftarrow s_i + BF \cdot (1 - s_i) \]

where the BF parameter (Boost Factor) is set to 0.2 in this work.

3.4 Time management and patience

In order to evaluate the time spent during the current dialogue, a speech rate of 200 words per minute is used (Yuan et al., 2006). Moreover, when the user hands the floor to the system, a silence of 2 seconds is added to this duration and 1 second the other way around. Finally, a Patience Manager module simulates the user patience: the maximum duration per dialogue task that the user can bear before hanging up. At each dialogue task, this value is computed as
\[ d_{\text{pat}} = 2\mu_{\text{pat}} \cdot \text{sigmoid}(X) \]  
(2)

where \( \mu_{\text{pat}} \) is the mean value (\( \mu_{\text{pat}} = 180s \)).

### 3.5 Scheduler Module

A non-incremental dialogue system can be transformed into an incremental one by adding an extra module: the Scheduler (Khouzaimi et al., 2014). Its objective is to make turn-taking decisions (whether to take the floor or not). When the user speaks, its partial utterance grows over time. At each new change, it is sent to the Scheduler that immediately asks the service for a corresponding response and then rollbacks the system’s context as long as it decides not to take the floor. If, on the other hand, it decides to commit to the last received partial utterance by taking the floor, then no rollback is performed and the dialogue context is effectively updated.

In this work, the Scheduler can perform two types of actions: WAIT and SPEAK, that is to say that it can wait for the next micro-turn without uttering anything or it can start retrieving the last response it got from the service.

Two versions of the Scheduler have been implemented: handcrafted rules were implemented in the first one whereas the second one embeds a reinforcement learning algorithm that learns to make turn-taking decisions by itself.

### 4 Turn-taking model

#### 4.1 Turn-taking phenomena

Several turn-taking phenomena can be observed when analysing human conversations. A taxonomy of these phenomena is introduced in (Khouzaimi et al., 2015b), three of which are replicated here through the SPEAK action:

**FAIL_RAW:** The listener sometimes does not understand the speaker’s message because of noise or unknown vocabulary. Therefore, she can barge-in and report the problem without waiting for the speaker to finish her sentence.

**INCOHERENCE_INTERP:** An incoherence is detected in the user’s utterance in the two following cases:

1. The user tries to modify or delete a non-existing event.
2. The user tries to fill a time slot that is already occupied.

Because of the ASR instability, as a security margin, the SPEAK decision will be taken two words after the incoherence is detected if it is maintained.

**BARGE_IN_RESP:** As soon as the service gives a full response to a partial utterance, the Scheduler considers that all the information needed has been given by the user. Like in the previous case, the decision is taken two words after.

- Hi, I would like to book a room tonight and I...
- Sorry but there are no rooms available at the moment.
4.3 Reinforcement learning

Despite late 20th century initial proposition (Levin and Pieraccini, 1997), reinforcement learning as the machine learning framework in the field of spoken dialogue systems is still largely explored in the current days (Lemon and Pietquin, 2007; Laroche et al., 2010; Pinault and Lefèvre, 2011; Ferreira and Lefevre, 2013; Young et al., 2013).

In non-incremental systems, at each dialogue turn, the system has to make a decision (action) hence moving to a new state. In this paper, as we study dialogue from a turn-taking point of view, the decision unit is the micro-turn.

4.3.1 Background

The turn-taking problem is here cast as a Markovian Decision Process (MDP) (Sutton and Barto, 1998), that is to say a quintuple $(S, A, T, R, \gamma)$ where $S$ is the set of states where the system can be during a dialogue and $A$ is the set of actions that can be performed at each time step. $T$ is the transition model, in other words, the set of probabilities $P(s_{t+1} = s' | s_t = s, a_t = a)$ of getting to state $s'$ at time $t + 1$ if the system was at state $s$ at time $t$ and performed action $a$. Such a decision makes the system get an immediate reward $r_t = R(s_t, a_t, s_{t+1})$ modeled by $R$. The action to choose at each state is given by a policy $\pi$ ($\pi(s_t) = a_t$) and the cumulative (discounted) reward is defined as $R_t = \sum_{t'=t}^{\infty} \gamma^{t'-t} r_{t'}$ ($\gamma$ is called the discount factor). Finally, each couple $(s, a)$ is associated with a value $Q^\pi(s, a) = \mathbb{E}[R_t | s_t = s, a_t = a]$ which is the expected cumulative reward for being at the state $s$, taking action $a$ and following the policy $\pi$ afterwards.

The goal of reinforcement learning is to find an optimal policy $\pi^*$ such that, for every other policy $\pi$, and for every state-action couple $(s, a)$, $Q^\pi(s, a) = Q^{\pi^*}(s, a) \geq Q^\pi(s, a)$.

4.3.2 State representation

The system state is characterised by the following features:

- **SYSTEM_REQ**: The current information that is asked for by the system. It can be a slot value, a confirmation or the system can ask an open question to make the user fill all the slots in one dialogue turn (6 alternatives).

- **LAST_INCR_RESP**: The Scheduler incrementally gets responses from the service.

This feature corresponds to the last response obtained (11 alternatives).

- **NB_USER_WORDS**: The number of words added by the user after the last change in the value of LAST_INCR_RESP (after the last input that made the service catch a new piece of information and change its mind about the response to deliver).

- **NORMALISED_SCORE**: The ASR Output simulator estimates the score of a partial utterance as the product of its components. Therefore, the longer the sentence the worse the confidence score, even if all the components have a decent score. To neutralise this effect we normalise the score by taking its geometric mean given the number of words. Suppose there are $n$ words in the current partial utterance and $s$ its score, then NORMALISED_SCORE $= s^n \frac{1}{n}$.

- **TIME**: The duration in seconds reached so far in the current task. This value is normalised so that it is around zero at the beginning of the task and around 1 for 6 minutes (maximum user patience).

In order to represent the function $Q(s, a)$, we maintain one linear model per action. There are 21 combinations between SYSTEM_REQ and LAST_INCR_RESP values that are the most likely to occur. They have been associated to the features $\phi_1$ to $\phi_{21}$. $\phi_i$ equals 1 when the $i^{th}$ combination happens in the current micro-turn and 0 otherwise. Less frequent combinations have been removed from this initial model: first they make the model more complex with in all likelihood no significant improvement in the performance (making the learning process slower to converge and more data demanding) and second, the Fitted-Q algorithm involves the inversion of a feature covariance matrix which could be ill-conditioned with these rare combinations.

NB_USER_WORDS is represented by three Radial Basis Function (RBF) features (Sutton and Barto, 1998) $\phi_1^{nw}, \phi_2^{nw}$ and $\phi_3^{nw}$. Their means are set to 0, 5 and 10 and the corresponding standard deviations are 2, 3 and 3. The same representation with 2 features is used for NORMALISED_SCORE: $\phi_1^{ns}$ and $\phi_2^{ns}$ centered at 0.25 and 0.75 and with a standard deviation of 0.3 for both.
Finally, TIME is represented as a single feature
\[ T = \text{sigmoid}(\text{TIME} - 180)/60 \] so that it is
almost 0 for \( \text{TIME}=0 \) and almost 1 after 6 min-
utes. As this variable increases, the user is more
and more likely to hangup, therefore the Q func-
tion is supposed to be monotonous with respect
to that feature so it is taken directly in the model
without the use of RBFs.

As a consequence, 28 parameters are involved
for each action (56 in total). Let \( \Theta(a) \) be the pa-
rameter vector \( (\Theta(a) = [\theta_0, \theta_1, \ldots, \theta_{27}]^T) \) cor-
responding to action \( a \) and \( \Phi(s, a) \) the feature vector
corresponding to state \( s \) and action \( a \), therefore:

\[
\Phi(s, a) = [1, \delta_1, \ldots, \delta_{21}, \phi_1^{\text{nw}}, \phi_2^{\text{nw}}, \phi_3^{\text{nw}}, \phi_1^{\text{nw}}, \phi_2^{\text{nw}}, T]^T
\]

\[
Q(s, a) = \Theta(a)^T \Phi(s, a) \tag{3}
\]

\[ Q^*(s, a) = \mathbb{E}_{s'\sim s, a}[R(s, a, s') + \gamma \max_{a'\in A} Q(s', a')] \tag{5}
\]

Therefore, the Fitted-Q algorithm is fed with a set of \( N \) MDP transitions \( (s_j, a_j, r_j, s'_j) \) and
aims to approximate the representation parameter
vector of the Q-function by performing a linear
regression at each iteration step. In our case,
for each action and at the iteration \( i \), the par-
ter vector is updated as follows (for commodity,
\( \phi(s_j, a_j) \) is noted \( \phi_j \)):

\[
\Theta^{(i)}(a) = \arg\min_{\Theta(a)} \sum_{j=1}^{N} (R_j^{(i-1)} - \Theta(a)^T \phi_j)^2 \tag{6}
\]

\[
R_j^{(i-1)} = r_j + \gamma \max_{a'\in A} (\Theta_{i-1}^T \phi(s', a)) \tag{7}
\]

This is a classical linear regression problem and
we use the closed formula for each iteration:

\[
\Theta^{(i)} = (\sum_{j=1}^{N} \phi_j \phi_j^T)^{-1} \sum_{j=1}^{N} \phi_j R_j^{(i-1)} \tag{8}
\]

The iteration stop criterion is

\[
\sum_{a\in A} ||\Theta^{(i)}(a) - \Theta^{(i-1)}(a)||_1 \leq \xi \tag{9}
\]

In our experiments, the convergence threshold
is set to \( \xi = 0.01 \).

5 Experiment

5.1 Experimental setup

Three dialogue scenario types involving diverse
adding, modifying and deleting tasks were used
for the experiments. For the training of the RL
strategy, the simulated speech recognition WER
was fixed at 0.15. We trained the system 50
times and each training session is made of 3000
episodes. The Fitted-Q algorithm was run ev-
every 500 episodes on the total batch from the be-


...
first episodes are enough to learn the optimal policy given our model. The 500 last episodes show that the learned strategy significantly outperforms the handcrafted baseline.

Incremental dialogue systems have the ability to report an error to the user in a more reactive way and to prevent it from speaking for a long time without being understood by the system. In noisy environments, these problems are even more likely to happen. Figures 4 and 5 show the effect of noise over dialogue duration and task completion. They represent the average performance over the 3 dialogue scenarios used in this experiment. Incremental dialogue, and the automatically learned strategy in particular, significantly increase the noise robustness. In the non-incremental case, the mean dialogue duration reaches 3 minutes and the task completion drops below 70%. Our learned strategy makes the dialogues finish 30 seconds earlier on average (17% gain) and the task completion is about 80%.

6 Conclusion and future work

In this paper, a simulated environment for a slot-filling task has been used to compare different expert and learned turn-taking strategies. A first one is non-incremental meaning that the user and the system cannot interrupt each other. As ASR noise increases, the dialogues tend to last longer leading to lower task completion. A second strategy is incremental, starting from three turn-taking phenomena present in the human-human interaction we translated them into a set of handcrafted rules for human-machine dialogue. This rule-based strategy is then shown to have better performance than the non-incremental case in terms of dialogue duration and task completion ratio when the noise is increasing. Eventually the Fitted-Q algorithm has been retained to automatically learn a third turn-taking strategy, still with the same objective (minimising the dialogue duration and maximising the task completion). This third strategy significantly improves noise robustness of the simulated dialogue system. We are now planning to evaluate this approach with real users and by taking subjective scores into account for learning optimal turn-taking strategies with respect to enlarged view of the system performance, such as comfort of use, friendliness etc.
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