

# W3C SIV Workshop. Agenda.

- SIV Architecture
- Use cases
- SIV syntax
- Conclusion



# W3C SIV Workshop. What should SIV in VoiceXML 3.0 support?

#### Combination of SIV with other resources (esp. ASR) :

- SIV only (i.e. without ASR, standalone SIV)
- SIV in parallel to ASR (ASR and SIV are separate resources)
- SIV integrated with ASR as one (combined) resource

### SIV types:

- Text independent
- Text dependent
- Text prompted

#### **Decision control:**

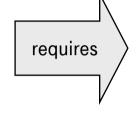
• Either the SIV engine or the application may control decisions (e.g. regarding acceptance/rejection)



# W3C SIV Workshop. SIV Core Functionality in VoiceXML 3.0.

### SIV must support:

- Enrollment
- Verification
- Identification



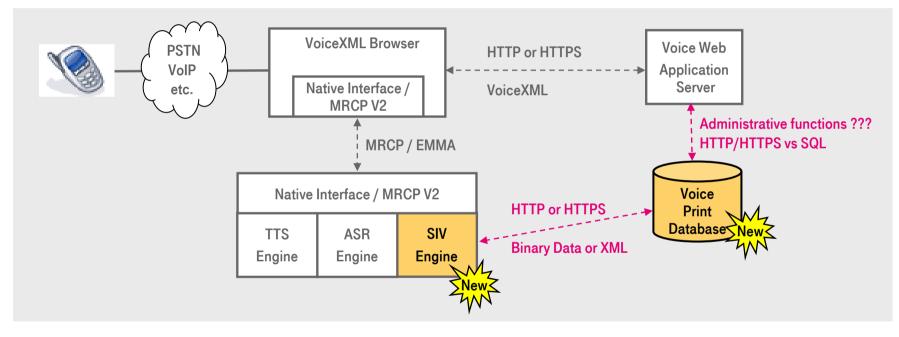
- Save voiceprints (after enrollment)
- Load voiceprints (before verification/identification)

*Note: V3 should load/store voiceprints implicitly (without explicit markup)* 

### Further basic/core functionalities for application development:

- Adaptation of voiceprints (during verification)
- Buffering of user utterances for later use
- Rollback/Undone of last turn
- Query SIV results (e.g. accept/reject information, score etc.)
- Catch SIV events (e.g. "noinput" or "nomatch" events)
- Query, copy, delete voiceprints (administration purposes) ⇒ outside of VoiceXML 3.0

#### **Proposed Architecture**



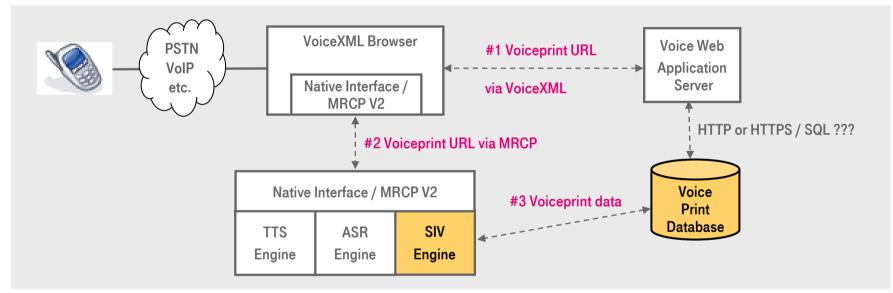
- Standard VoiceXML architecture extended by MRCP-based SIV engine and voiceprint store



#### Architectural key statements

- Support MRCP v2 for integration of SIV engines
  - SIV engine should be integrated using a standardized interface to allow flexible replacement of SIV resources (product replacement).
- Extend MRCP vs. limited SIV functionalities
  - Some SIV vendors require functionalities which are not covered by MRCP v2 (e.g. COPY voiceprint, expected utterance). A decision is necessary for either using a standardized interface or to support the full set of SIV features of various vendors.
- Use EMMA for representation of SIV results
  - SIV results should be represented using EMMA standard.
- Use web protocols for voice print transport
  - Use of HTTP/HTTPS provide flexibility in deployment scenarios

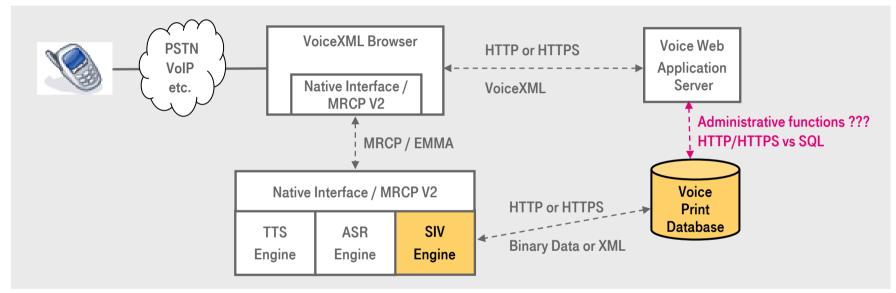
#### Voiceprint management: load and save voiceprints via MRCP



- MRCPv2 supports voiceprint URLs only (i.e. not the voiceprint itself)
- For identification a list of voiceprint URLs or a URL identifying a group will be necessary
- Loading/storing of voiceprints should be implicitly done by V3

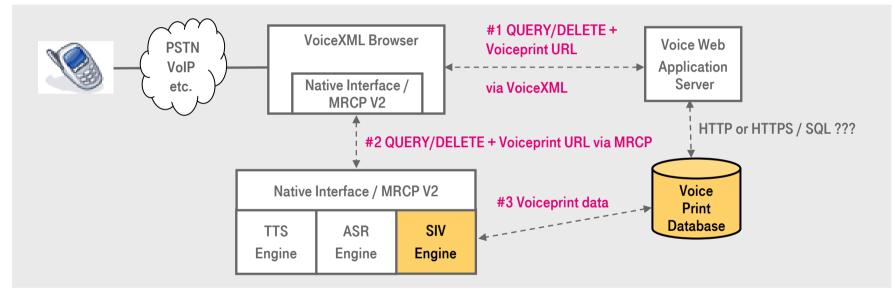
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#### Voiceprint management: query/copy/delete voiceprints (Option 1)



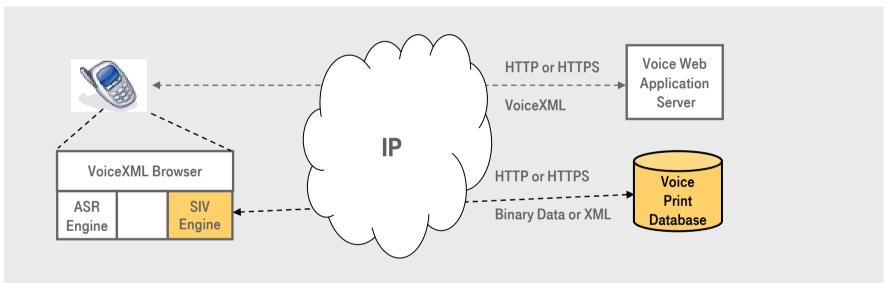
- MRCPv2 does not provide all necessary administrative functions (e.g. COPY).
- Advantages option 1: administrative functions not executed by VoiceXML
- Disadvantage option 1: proprietary interface to voiceprint database.

Voiceprint management: query/copy/delete voiceprints (Option 2)



- MRCPv2 supports QUERY and DELETE commands
- Option 2: Reflect QUERY and DELETE at V3 syntax level
- Disadvantage option 2: admin functions executed via VoiceXML

Embedded deployment supported by proposed architecture



- Usage of web protocols (HTTP/HTTPS) for voiceprint transport supports future deployment scenarios



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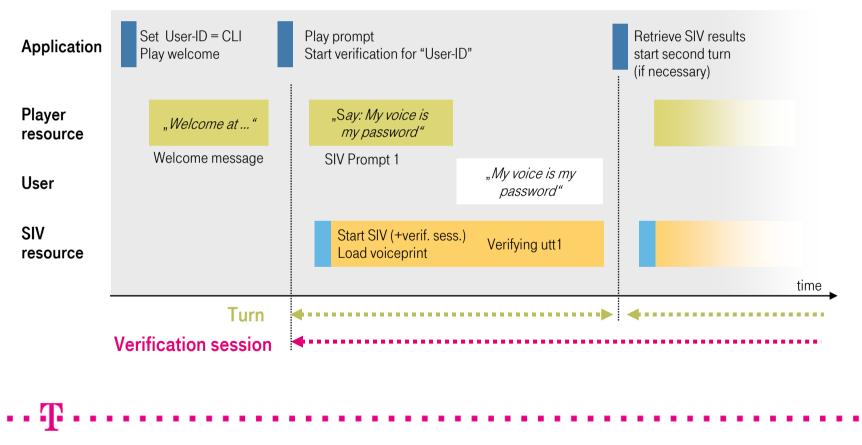
SIV Architecture

### Use cases

- SIV syntax
- Conclusion

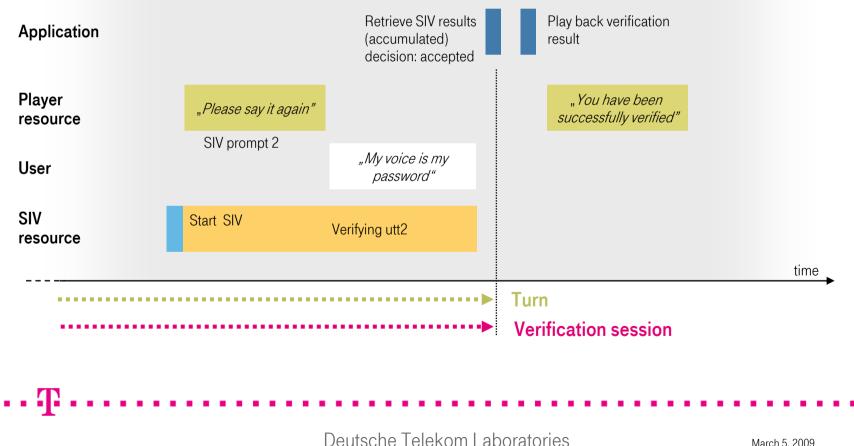


#### Basic uses case #1: standalone SIV without ASR



Deutsche Telekom Laboratories

#### Basic uses case #1: standalone SIV without ASR (cont'd)

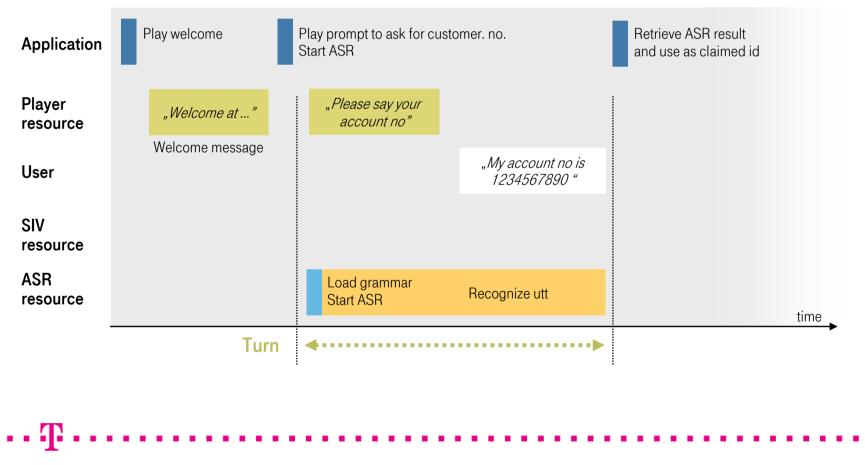


#### Basic uses case #1: standalone SIV without ASR (cont'd)

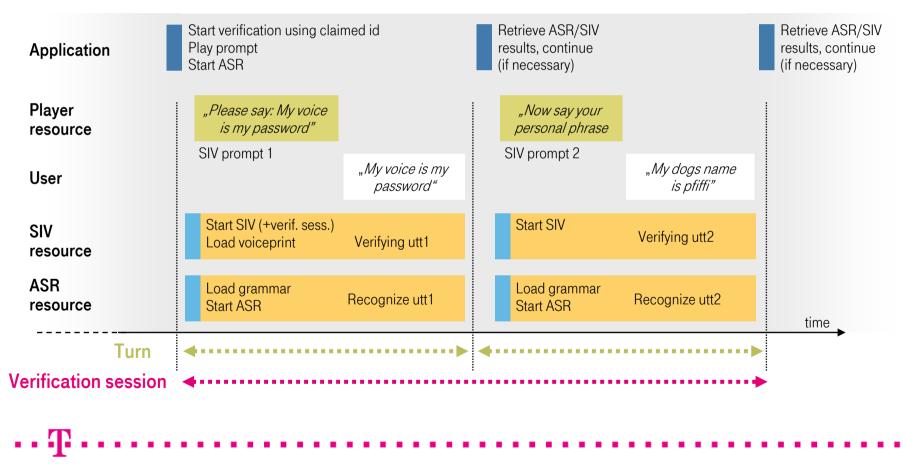
- SIV needs to implement speech detection/endpointing (like ASR)
- SIV needs to implement timeouts (like ASR)
- SIV should in this use case provide bargein functionality
- SIV may need multiple turns (within one SIV session)
- Author needs control of whether another turn is necessary or not (
   ⇒ syntax)



#### Basic uses case #2: SIV + ASR



#### Basic uses case #2: SIV + ASR (cont'd)



### Basic uses case #2: SIV + ASR (cont'd)

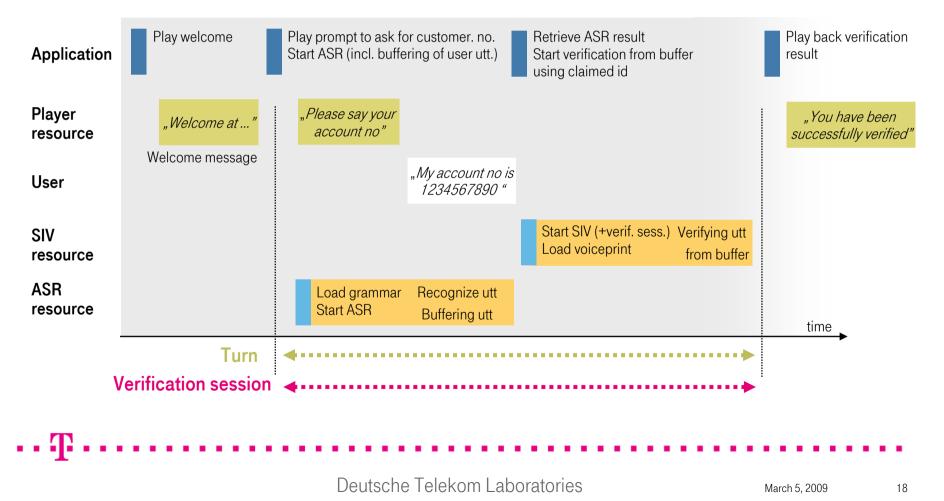
- SIV may run in parallel to ASR (difference to use case #1)
- Idea: use ASR to make sure that the user repeated the correct (prompted) utterance

#### **Issues:**

- What if user repeated wrong utterance and ASR is used to check if SIV is not successful? ⇒
  conclusion: undone/rollback functions necessary to remove latest utterance from cumulated result
- Problem if engine ended session by itself ⇒ conclusion: session has to be ended by app only
- Same problem if adaptation was enabled ⇒ rollback for adaptation necessary (supported by MRCP thru abort header for end-session method)



#### Basic uses case #3: ASR + SIV from buffer



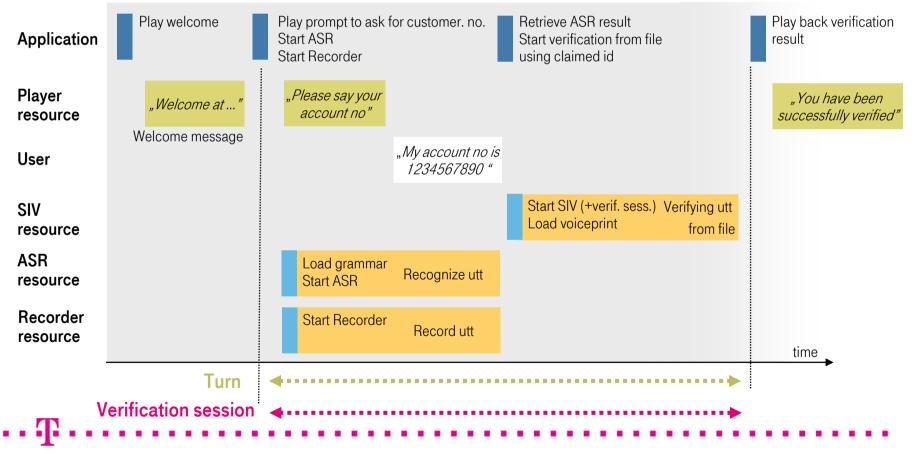
...

### Basic uses case #3: ASR + SIV from buffer (cont'd)

- ASR must be able to buffer one (or more?) utterances for later verification
- Requires new ASR functionality (e.g. new attribute siv\_buffer)



#### Basic uses case #4: ASR + SIV from file



#### Basic uses case #4: ASR + SIV from file

- Recorder resource running in parallel to ASR to record user utterance
- Verification of recorded utterance requires special parameter (WAV file reference for verification from file)
- Which audio-formats are supported?



# W3C SIV Workshop. Agenda.

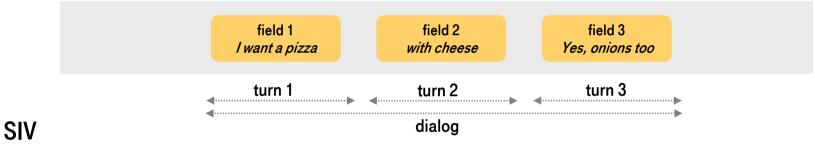
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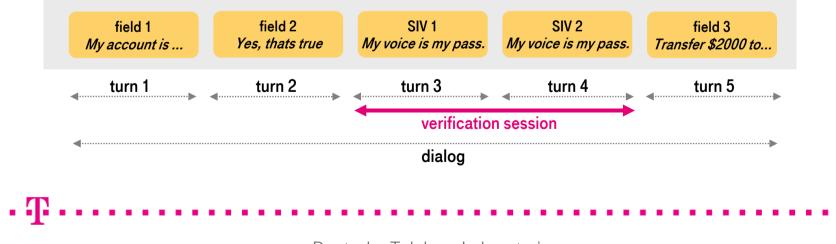
## W3C SIV Workshop. SIV vs. ASR.

### ASR

ASR dialogs consists of one or more independent turns



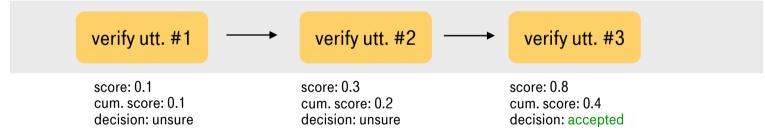
• SIV dialogs consists of one or more turns that are part of an enrollment/verification session



# W3C SIV Workshop. SIV sessions.

#### Sessions:

- Enrollment and verification/identification can be session based
- SIV engines often compute (internally) cumulative results when verifying several utterances (turns)



- MRCP provides Start-Session and End-Session methods
- Voiceprint-ID (given when session is started) defines which voiceprint to be trained or matched during the enrollment/verification session



#### Inputs for VoiceXML 3.0 SIV elements:

- Mode (enroll/verify/identify)
- SIV-ASR (SIV only, SIV+ASR)
- Adaptation (bool)
- Buffering (for <field>) and "useBuffer" for <siv>
- Req. phrase
- Decision threshold
- Timeouts, like ASR
- ID (voiceprint URL), WAV file reference for verification from file (file URL)
- Rollback

#### Administrative functions:

Query/copy/delete function



#### Syntax option 1: Extend existing <field ...> element

• Example:

```
<field name="utt1" siv_type="verify" ...>
   <voiceprint src="voiceprint_url"/>
   <grammar src="speech_grammar"/>
</field>
```

- Advantage:
  - reuse of existing element
- Disadvantages:
  - increased complexity of <field> element
  - control of begin and end of SIV session not sufficient
- Comment
  - multiple fields may belong to a single SIV session and hence use the same voiceprint. Referencing the same voiceprint URL in subsequent <field> is redundant.



#### Syntax option 2: Create one new <siv> element

• Example:

```
<par>
<par>
<siv name="utt1" type="enroll / verify / identify" ...>
<voiceprint src="voiceprint_url"/>
</siv>
<field>
<field>
<grammar src="speech_grammar"/>
</field>
</par>
```

Advantage:

. . .

- no increased complexity of <field> element
- clear separation of SIV and ASR syntax
- Disadvantages:
  - additional element necessary
  - control of begin and end of SIV session not sufficient

#### Syntax option 3: Create a new element for each of the 3 basic functions:

• Example:

enrollment	<enroll></enroll>
verification	<verify></verify>
identification	<identify></identify>

- Advantage:
  - better control of meaningful combinations of attribute values



#### Open issues:

- Control of begin/end of SIV session
- Session needs to be closed by application (to allow control of rollback)
- How to execute a rollback? Separate <rollback> element?



# W3C SIV Workshop. SIV results.

### Training:

- more\_data\_needed [true, false]
- decision [accepted, rejected, undecided]
- score (0 ... 100, 50 = decision threshold)

### Verification:

- more\_data\_needed [true, false]
- decision [accepted, rejected, undecided], cumulative and local
- score (0 ... 100, 50 = decision threshold), cumulative and local
- adapted [true, false]

### Identification:

- more\_data\_needed and adapted like for verification
- array of decision, score and voiceprint-ID

⇒ These are core results, should be mandatory within VoiceXML 3.0

### 

# W3C SIV Workshop. SIV results.

#### Additional results:

• Various vendors provide more results. Most of them are nice-to-have.

⇒Could be optional within VoiceXML 3.0

### Examples:

- valid [true, false] (is the utterance valid?)
- device [cellular phone, electret phone, carbon button phone]
- gender [male, female]
- matched (is gender and device type same as in training?)
- num\_utterances (number of utterances)
- ...
- ⇒ Proposal: Collect list of results of existing technologies and generate list of mandatory results. Decide on whether optional results should be allowed

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### W3C SIV Workshop. Other open issues.

#### The following issues have not been addressed here:

- Events: SIV might generate a "noinput" event, a combination of SIV and ASR leads to doubled or conflicting events
- Timeout parameters: Should SIV and ASR always use the same timeouts? Different resources (e.g. from different vendors) may behave inconsistently on the same timeouts.



## W3C SIV Workshop. Summary / Conclusion.

#### Similarities and differences between ASR and SIV

• SIV and ASR share some similarities, but do also have a lot of differences (e.g. SIV session)

#### Detailed requirements / use case description necessary:

- VoiceXML 3.0 requirements document contains a very generic set of SIV requirements
- For a further discussion, a common understanding regarding use cases is necessary

#### Proposed next steps:

- Collect and describe use cases in detail, to achieve a common understanding
- Decide which use cases to support in VoiceXML 3.0 (and which not)
- Collect list of (mandatory) results and decide whether optional results will be allowed
- Compare with MRCP and decide what functionality from MRCP also to support in VoiceXML 3.0

