

Profil SIP-Connect 2.0 (siplogin.de)

Rufnummernformate:

E.164 ohne führendem „+“ Übermittelte Formate mit führendem "+", "0" oder "00" werden best effort vom outbox System normalisiert

Outgoing Calls SIP TK -> obx NGN

From (Display) Name: User provided Number
 From Number: Network provided Number
 To: Called Party Number
 P-Asserted-Identity: Network provided Number
 P-Preferred-Identity: wird verworfen
 Privacy: Anonymus Anzeige im From Header oder Privacy = id wird unterstützt

Incoming Calls obx NGN -> SIP TK

Request URI: Die Called Party Number wird in der Request URI gesetzt
 Display Name: User provided Number
 User Name: User provided Number
 To: Die Called Party Number wird in den To Header gesetzt
 P-Asserted-Identity: Network provided Number
 P-Preferred-Identity: keine Übermittlung
 Privacy: Anonymisierung über Privacy = id und anonymus Anzeige im From Header

INVITE Beispiel:

```
INVITE sip:4940123456789@siplogin.de SIP/2.0.
Via: SIP/2.0/UDP 1.123.4.56:5060;branch=z9hG4bKdd45.a2a42977.0.
Max-Forwards: 68.
From: "49223630390" <sip:49223630390@siplogin.de;user=phone>;tag=a4ZByv29960pQ.
From: "anonymous" <sip:anonymous@anonymous.invalid;user=phone>;tag=g94ep7325ytpK.
To: <sip:4940123456789@siplogin.de;user=phone>.
Call-ID: c2b818df-13a2-1239-ad87-02bacb569a24.
CSeq: 20347984 INVITE.
Contact: <sip:1.123.4.56:5060;did=f52.55a73544>.
Privacy: none.
P-Asserted-Identity: <sip:4922363030@siplogin.de;user=phone>
Privacy: id.
P-Asserted-Identity: <sip:anonymous@anonymous.invalid;user=phone>
```

Variante Anonym

Variante Anonym